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Introduction

Welcome to the NAACL-HLT 2019 Student Research Workshop! This year’s submissions were organized in two tracks: research papers and thesis proposals.

- Research papers may describe completed work, or work in progress with preliminary results. For these papers, the first author must be a current graduate or undergraduate student.
- Thesis proposals are geared towards students who have decided on a thesis topic and wish to get feedback on their proposal and broader ideas for their continuing work.

This year, we received a total of 54 submissions: 50 of these were research papers, and 4 were thesis proposals. We accepted 22 research papers and 2 thesis proposals, resulting in an overall acceptance rate of 44%. The main authors of the accepted papers represent a variety of countries: Australia, Canada, India (6 papers), Ireland, Japan (2 papers), Hong Kong, South Korea, Taiwan, USA (10 papers). These papers span topics across a wide range of subdisciplines and topics within NLP and CL.

Accepted research papers will be presented as either talks or posters within the NAACL main conference. We have 9 accepted papers being presented as talks and 14 as posters.

Following previous editions of the Student Research Workshop, we have offered students the opportunity to get mentoring feedback before submitting their work for review. Each student that requested pre-submission mentorship was assigned to an experienced researcher who read the paper and provided some comments on how to improve the quality of writing and presentation of the student’s work. A total of 22 students participated in the mentorship program. During the workshop itself, we will also provide an on-site mentorship program. Each mentor will meet with their assigned students to provide feedback on their poster or oral presentation, and to discuss their research careers.

We would like to express our gratitude for the financial support from the National Science Foundation (NSF), the Computing Research Association Computing Community Consortium (CRA-CCC), the National Research Council (NRC) Canada, and Google. Thanks to their support, this year’s SRW is able to assist students with their registration, travel, and lodging expenses.

We would like to thank the mentors for dedicating their time to help students improve their papers prior to submission, and we thank the members of the program committee for the constructive feedback they have provided for each submitted paper.

This workshop would not have been possible without the help from our faculty advisors, and we thank them for their guidance along this year of workshop preparation. We also thank the organizers of NAACL-HLT 2019 for their continuous support. Finally, we would like to thank all students who have submitted their work to this edition of the Student Research Workshop. We hope our collective effort will be rewarded in the form of an excellent workshop!
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Daria Dzendzik, Carl Vogel and Jennifer Foster

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Shikha Bordia and Samuel R. Bowman

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Deep Learning and Sociophonetics: Automatic Coding of Rhoticity Using Neural Networks
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Data Augmentation by Data Noising for Open-vocabulary Slots in Spoken Language Understanding
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Rohan Mishra, Pradyumn Prakhar Sinha, Ramit Sawhney, Debanjan Mahata, Puneet Mathur and Rajiv Ratn Shah
Is it Dish Washer Safe?
Automatically Answering “Yes/No” Questions using Customer Reviews

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Abstract
It has become commonplace for people to share their opinions about all kinds of products by posting reviews online. It has also become commonplace for potential customers to do research about the quality and limitations of these products by posting questions online. We test the extent to which reviews are useful in question-answering by combining two Amazon datasets, and focusing our attention on yes/no questions. A manual analysis of 400 cases reveals that the reviews directly contain the answer to the question just over a third of the time. Preliminary reading comprehension experiments with this dataset prove inconclusive, with accuracy in the range 50-66%.

1 Introduction
Consumers often carry out online research about a product before purchasing. This can take the form of reading consumer reviews and/or asking specific questions on online fora. In this paper we ask whether a question-answering (QA) system can utilize the information in consumer reviews when answering yes/no questions about a product.

We compile a dataset of questions about Amazon products together with consumer reviews of the same products, and manually analyse a sample of 100 questions from four domains. We find that the reviews contain the answer in only 45% of cases. In 36% of cases, the answer is directly expressed in at least one of the reviews, and 9% of the time, it is indirectly expressed. This suggests that reviews can sometimes be useful and so we go on to experiment with QA systems that use the reviews in addition to the question. We focus on yes/no questions. Being able to answer these is not only an indicator of whether reviews will be useful for other question types but is also a signal of how much comprehension is actually taking place.

In our preliminary experiments with three domains from this new dataset, we compare systems which attempt to answer a yes/no question based on the question alone to those that also use related reviews. We experiment with two methods for selecting relevant sentences from the reviews, and with various representations for encoding the questions and reviews including bag-of-words, word2vec (Le and Mikolov, 2014), ELMo (Peters et al., 2018), and BERT (Devlin et al., 2018). On the development set, our systems tend to outperform the chance baseline but not by a large margin – our development set results range from 50 to 66%. Over the three domains, we also find that the question-only systems tend to perform as well as and sometimes outperform those which also use the reviews, suggesting that separating the answers from the noise in these reviews is not straightforward.

2 Related Work
A number of studies have explored the use of customer reviews in retrieval and question answering. Using Amazon data, Yu et al. (2018) develop a framework which returns a ranked list of sentences from reviews or existing question-answer pairs for a given question. Xu et al. (2019) create a new dataset comprising Amazon laptop reviews and questions and Yelp restaurant reviews and questions, where reviews are used to answer questions in multiple-turn dialogue form. Bogdanova et al. (2017) and Bogdanova and Foster (2016)) do not use review data but also focus on QA over user-generated content, attempting to find similar questions or rank answers in user fora. We use the same Amazon data as Yu et al. (2018) but consider a wider set of domains (they consider only two), and attempt to directly answer yes/no questions. To the best of our knowledge, the novelty in...
our work lies in trying to directly answer customer questions using user-generated reviews.

Unlike popular Reading Comprehension datasets such as MovieQA (Tapaswi et al., 2016) and SQuAD (Rajpurkar et al., 2018, 2016), which are created by crowdsourcing, we work with authentic user-generated data. This means that the data is collected from sources where users spontaneously created content for their own purposes. Since there is no guarantee that reviews contain text related to the question, there is no span data that can be reliably used to provide the answer. This, together with the considerable volume of review text, contributes to the difficulty of the task.

3 Data

We work with two Amazon datasets: the first, He and McAuley (2016),\(^1\) is a collection of product reviews from 24 domains. The second, Wan and McAuley (2016); McAuley and Yang (2016), contains questions and answers about products from 21 domains. These two datasets have 17 domains in common and can be matched using the Amazon Standard Identification Number (ASIN).

In order to obtain data with reviews, questions and answers, we first select all those products which contain reviews and questions, focusing on yes/no questions. We observe that the majority of questions can be answered “Yes” (65-75% depending on the domain), so we balance the data by selecting an equal amount of yes/no questions. This results in 80391 questions about 40806 products – see Table 1 for more details.

All data is fully user-generated except the answer tags which are provided by McAuley and Yang (2016). An example of the combined data is shown below (we keep the original spelling).

**Reviews (R):** 
...I was a little surprised at how much time it took to assemble. There were a lot of the smaller parts that I would have assumed pre-assembled that weren’t...\(^2\)

**Question (Q):** Does it come assembled

**Answer (A):** No, count on, at least an hour to assemble. (Answer Tag (AT): No)

The authentic user-generated nature of this dataset makes it significantly different from other reading comprehension datasets. Table 2 shows a comparison with MovieQA and SQuAD2.0. The length of questions is almost the same (10-11.5 words) in all three datasets, although the number of instances (movie plots for MovieQA, topic for SQuAD and product for Amazon) is significantly bigger. Moreover, the average length of context in the Amazon data is unquestionably larger than in the MovieQA and SQuAD: 188 vs 35 and 5 sentences, and 3265 vs 728 and 117 words.

To better understand the nature of questions we carried out some additional analysis looking into question formulation. 21% of questions are formulated with more than one sentence (16–31% depending on the domain), more than 25% of questions (20686) start with the word Does, and more than 15% (>12000) with Can, Is or Will.

4 Do reviews contain the answers?

According to Kaushik and Lipton (2018) reading comprehension datasets are not studied enough in terms of difficulty. We conduct a manual analysis to better understand the relationship between questions and reviews, to assess the feasibility of using user reviews to answer user questions and to estimate an upper bound on system performance. 100 questions from four domains are analysed. We define seven classes of questions:

**Easy:** Questions are clearly answered in the reviews, e.g.

(1) R: ... I used two of these, one for each side of the bed.

Q: can this product be used if 2 bed rails are needed for one bed? (AT: Yes)

**Error:** Questions where the answer tag contradicts the user-provided answer, e.g.

(2) Q: Can you mount this upside down i.e. The receiver on top of the bumper?

A: I don’t see why not, the is nothing preventing you. (AT: No)

**Indirect:** Questions which can be indirectly answered by the review, e.g.

(3) R: ... it doesn’t give an exact voltage and maxes out at 12.7 volts

Q: Can this be used to charge a 48v battery? (AT: No)

Q: Is this a good charger/jump starter for a 12v deep cell battery? (AT: Yes)
Table 1: Balanced yes/no dataset statistics per domain: Number of products (P) which have yes/no questions, number of questions (# Question), count of sentences in questions (S), total number of words in questions (W), total number of reviews (# Reviews), all number of sentences in reviews, total number of words in reviews.

Table 2: Comparison of balanced yes/no dataset with MovieQA and SQuAD2.0: Number of instances (I): articles, movie plots, or products; Number of text passages (T): context, reviews, or plots; Average number of words in the question (AVG W in Q), sentences in the text (AVG S in T), and words in the text (AVG W in T).

**Real-world**: Questions where the review does not contain the answer but where an educated guess can be made using common sense or real-world knowledge, e.g.

Q: *Can I use the cloth to clean the keys on my clarinet?* (AT: Yes)

Q: *Has anyone traveled with this stroller on an airplane?* (AT: No)

**Opinion**: Questions which can be answered differently based on different reviews (6) or when the answer and review contain contradictory information (7). Often such questions ask for an opinion, so the answer depends on the user providing it, e.g.

R: *...all in all these pans are worthless ...so many folks have had a horrid experience!!* ...At $15.00, it's a good pan for my purposes ...This pan is awesome for the price*

Q: *is this item any good?* (AT: No)

Q: *...but it seems to get a little hot and makes a plastic noise under the sheet...*  

3The sentences are taken from different reviews.

Q: *does it make noise when baby moves around?*

A: *No not with a sheet on it. (AT: No)*

**Unrelated**: Questions which are asked not about the product but about service and delivery, e.g.

Q: *Is there a warranty when you buy it from amazon?*

**No answer**: Questions which cannot be answered without additional information, i.e. reviews do not contain the required information.

The *indirect* and *real-world* classes can be considered to be difficult questions. However, in general, we believe that the *easy*, *indirect* and *real-world* question classes can be answered without resorting to guessing.

Detailed information is provided in Table 3. Around 53.5% of questions can be answered (36.5% are easy and 17% are difficult). Although it is difficult to conclude too much from this sample of 400, we can roughly estimate that the best performance we could expect from an automatic QA system would be around 77%. This means the
system answers all answerable questions correctly (53.5%) and guesses half of those questions which cannot be answered (23.25%).

### 5 Preliminary Experiments

#### 5.1 Approach

In order to establish some baselines on this dataset and task, we carry out preliminary binary classification experiments with a sample of the domains. There are three aspects to the systems we evaluate:

<table>
<thead>
<tr>
<th>Domain</th>
<th>Answerable</th>
<th>Guessing</th>
<th>Total (%)</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>Easy</td>
<td>Indirect</td>
<td>Real-world</td>
</tr>
<tr>
<td>Home &amp; Kitchen</td>
<td>31</td>
<td>9</td>
<td>4</td>
</tr>
<tr>
<td>Beauty</td>
<td>37</td>
<td>8</td>
<td>6</td>
</tr>
<tr>
<td>Baby</td>
<td>44</td>
<td>11</td>
<td>8</td>
</tr>
<tr>
<td>Clothing Shoes &amp; Jewellery</td>
<td>34</td>
<td>8</td>
<td>14</td>
</tr>
<tr>
<td>Total</td>
<td>146</td>
<td>36</td>
<td>32</td>
</tr>
</tbody>
</table>

Table 3: Selection of 100 questions from 4 domains for manual analysis. The last column contains the percentage of the analysed questions from each domain (eg. 100 is 4.6% of the Baby question data, 3.6% of Beauty, etc.).

**Binary Classifier** Following our previous work (Dzendzik et al., 2017) we use logistic regression with the bag-of-word, ELMO and word2vec representations. In the BERT experiments we add a softmax classification layer on top of the final hidden state of the transformer.

**Review Filtering**  The reviews in our dataset are long compared to the answer passages used in other reading comprehension tasks – see Tables 1 and 2. Pascanu et al. (2012) report that long sequences are hard to process from both time and resource perspectives with sequence-to-sequence models. Therefore, rather than using the full text of the reviews, we use string similarity to select only those sentences that are likely to be relevant to the question. We base our selection method on our previous work which achieved state-of-the-art performance on the MovieQA reading comprehension task (Dzendzik et al., 2017; Tapaswi et al., 2016). Review sentences are compared to the question using cosine similarity of tf-idf representations, bag of words overlap, character n-grams, and window slide. Two sets of sentences are extracted. The first one is based on sentence union, in other words, all sentences which have been marked as relevant by any of the metrics are selected. The second one is based on sentence intersection, i.e. only sentences which have been marked as relevant by more than one similarity metric are selected.

**5.2 Experimental Setup**

We compare systems which use the review and question text to systems which just use the question text. We select three of the four domains used for manual analysis. We exclude Clothing Shoes & Jewellery due to the small number of questions.

Table 4 represents the number of questions in the training, development and test set and the ratio of “Yes” and “No” questions in each of them. The evaluation metric is accuracy.

For the bow and word2vec experiments, we normalize the text and remove stop words. We use a pre-trained Google News word2vec model. Every text is encoded as a sum of its word vectors and normalized. For the ELMO representation we average three layers of ELMO output and represent a sentence as concatenation of its word vectors.

In the question-only BERT experiment, we perform single-sentence classification (Devlin et al., 2018, Fig. 3b). In the experiments where the reviews are used, we perform sentence pair classification where the question is the first sentence and the review text the second (Devlin et al., 2018, Fig. 3a). We use the pretrained models B\textsubscript{base} and B\textsubscript{large}. Both of them are uncased.

---

4 We use https://github.com/allenai/allennlp – l.v. 04/2019
5 We use https://github.com/huggingface/pytorch-pretrained-BERT – l.v. 04/2019
6 Although data is balanced, we divide the dataset by products so some fluctuation between the percentage of “Yes” and “No” questions is possible.
7 We use the word “sentence” in the same way as Devlin et al. (2018)
### Table 4: Domain split of the training, development and test sets using the number of questions and the ratio of “Yes” and “No” questions in each of them.

<table>
<thead>
<tr>
<th>Domain</th>
<th>Training</th>
<th>Development</th>
<th>Test</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>All</td>
<td>All</td>
<td>All</td>
</tr>
<tr>
<td></td>
<td>Yes %</td>
<td>No %</td>
<td>Yes %</td>
</tr>
<tr>
<td>Home &amp; Kitchen</td>
<td>8502</td>
<td>50.02</td>
<td>49.98</td>
</tr>
<tr>
<td>Beauty</td>
<td>1965</td>
<td>49.21</td>
<td>50.79</td>
</tr>
<tr>
<td>Baby</td>
<td>1542</td>
<td>50.26</td>
<td>49.74</td>
</tr>
</tbody>
</table>

### Table 5: Results on development set of Logistic Regression (LR) applied to bag of word (bow), word2vec (w2v) and ELMO (elmo) representations, and BERT models (B base and B large) for 3 domains. The best question-only (Q only) and question+review (Q+Review) systems are in bold.

<table>
<thead>
<tr>
<th>Method</th>
<th>Data</th>
<th>Baby</th>
<th>Beauty</th>
<th>Home and Kitchen</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>Q Only</td>
<td>Q + Review</td>
<td>Q Only</td>
<td>Q + Review</td>
</tr>
<tr>
<td></td>
<td>Intersection</td>
<td>Union</td>
<td>Intersection</td>
<td>Union</td>
</tr>
<tr>
<td>LR bow</td>
<td>65.33</td>
<td>58.33</td>
<td>59.66</td>
<td>62.14</td>
</tr>
<tr>
<td>LR w2v</td>
<td>59.33</td>
<td>60.67</td>
<td>59.66</td>
<td>57.44</td>
</tr>
<tr>
<td>LR elmo</td>
<td>53.33</td>
<td>56.99</td>
<td>56.99</td>
<td>60.83</td>
</tr>
<tr>
<td>B base</td>
<td>65.66</td>
<td>55.67</td>
<td>60.00</td>
<td>64.75</td>
</tr>
<tr>
<td>B large</td>
<td>52.00</td>
<td>63.00</td>
<td>48.00</td>
<td>48.82</td>
</tr>
</tbody>
</table>

### Table 6: Results on test set with best-scoring question and review+question systems on development set.

<table>
<thead>
<tr>
<th>Model</th>
<th>Data</th>
<th>Accuracy</th>
</tr>
</thead>
<tbody>
<tr>
<td>B base</td>
<td>Baby</td>
<td>64.17</td>
</tr>
<tr>
<td>B large_inter</td>
<td>Question Only</td>
<td>49.22</td>
</tr>
<tr>
<td>B base</td>
<td>Question Only</td>
<td>64.41</td>
</tr>
<tr>
<td>B large_union</td>
<td>Question+Review</td>
<td>47.22</td>
</tr>
<tr>
<td>B base</td>
<td>Question Only</td>
<td>58.57</td>
</tr>
<tr>
<td>B large_inter</td>
<td>Question+Review</td>
<td>60.23</td>
</tr>
</tbody>
</table>

### 5.3 Results

The development set results are shown in Table 5. Apparently, questions themselves provide some information and help find the correct answer: two out of three domains show the best performance using the question only. Only the Home and Kitchen domain shows better performance with the question+review systems. When selecting sentences from the reviews, there is no clear winner between the intersection and union methods. It varies according to method and domain.

BERT, the base and large models, perform better than logistic regression on the development set so we apply the best question-only and question+review models to the test set (Table 6). Performance drops below chance for two domains. It remains to be seen why we are seeing this unstable performance.

### 6 Conclusions

We introduce a fully user-generated reading comprehension dataset by composing two existing datasets into a new one designed to address yes/no questions about products using reviews. All data in this work is substantial and comes from real users. We provide a preliminary analysis of data and share that reviews can, to some extent, be used to answer yes/no questions.

We build several baseline systems. Although performance does not reach our estimated upper bound of 77%, our results show that they are doing more than mere majority classification. The relatively good performance of the question-only systems leads us to believe that the systems are applying closed-world assumptions by associating terms in the training set questions with terms in the test set questions.

Each of the components of our systems can be replaced or improved. Our immediate next step is to investigate more closely the part of the system which selects relevant sentences from the reviews.

### Acknowledgments

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References


Identifying and Reducing Gender Bias in Word-Level Language Models

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\textbf{Abstract}

Many text corpora exhibit socially problematic biases, which can be propagated or amplified in the models trained on such data. For example, \textit{doctor} cooccurs more frequently with male pronouns than female pronouns. In this study we (i) propose a metric to measure gender bias; (ii) measure bias in a text corpus and the text generated from a recurrent neural network language model trained on the text corpus; (iii) propose a regularization loss term for the language model that minimizes the projection of encoder-trained embeddings onto an embedding subspace that encodes gender; (iv) finally, evaluate efficacy of our proposed method on reducing gender bias. We find this regularization method to be effective in reducing gender bias up to an optimal weight assigned to the loss term, beyond which the model becomes unstable as the perplexity increases. We replicate this study on three training corpora—Penn Treebank, WikiText-2, and CNN/Daily Mail—resulting in similar conclusions.

\section{Introduction}

Dealing with discriminatory bias in training data is a major issue concerning the mainstream implementation of machine learning. Existing biases in data can be amplified by models and the resulting output consumed by the public can influence them, encourage and reinforce harmful stereotypes, or distort the truth. Automated systems that depend on these models can take problematic actions based on biased profiling of individuals. The National Institute for Standards and Technology (NIST) evaluated several facial recognition algorithms and found that they are systematically biased based on gender (Ngan and Grother, 2015). Algorithms performed worse on faces labeled as female than those labeled as male.

Models automating resume screening have also proved to have a heavy gender bias favoring male candidates (Lambrecht and Tucker, 2018). Such data and algorithmic biases have become a growing concern. Evaluation and mitigation of biases in data and models that use the data has been a growing field of research in recent years.

One natural language understanding task vulnerable to gender bias is language modeling. The task of language modeling has a number of practical applications, such as word prediction used in onscreen keyboards. If possible, we would like to identify the bias in the data used to train these models and reduce its effect on model behavior.

Towards this pursuit, we aim to evaluate the effect of gender bias on word-level language models that are trained on a text corpus. Our contributions in this work include: (i) an analysis of the gender bias exhibited by publicly available datasets used in building state-of-the-art language models; (ii) an analysis of the effect of this bias on recurrent neural networks (RNNs) based word-level language models; (iii) a method for reducing bias learned in these models; and (iv) an analysis of the results of our method.

\section{Related Work}

A number of methods have been proposed for evaluating and addressing biases that exist in datasets and the models that use them. Recasens et al. (2013) studies the neutral point of view (NPOV) edit tags in the Wikipedia edit histories to understand linguistic realization of bias. According to their study, bias can be broadly categorized into two classes: framing and epistemological. While the framing bias is more explicit, the epistemological bias is implicit and subtle. Framing bias occurs when subjective or one-sided words are used. For example, in the...
sentence—“Usually, smaller cottage-style houses have been demolished to make way for these McMansions.”, the word McMansions has a negative connotation towards large and pretentious houses. Epistemological biases are entailed, asserted or hedged in the text. For example, in the sentence—“Kuypers claimed that the mainstream press in America tends to favor liberal viewpoints;” the word claimed has a doubtful effect on Kuypers statement as opposed to stated in the sentence—“Kuypers stated that the mainstream press in America tends to favor liberal viewpoints.” It may be possible to capture both of these kinds of biases through the distributions of co-occurrences. In this paper, we deal with identifying and reducing gender bias based on words co-occurring in a context window.

**Bolukbasi et al. (2016)** propose an approach to investigate gender bias present in popular word embeddings, such as word2vec (Mikolov et al., 2013). They construct a gender subspace using a set of binary gender pairs. For words that are not explicitly gendered, the component of the word embeddings that project onto this subspace can be removed to debias the embeddings in the gender direction. They also propose a softer variation that balances reconstruction of the original embeddings while minimizing the part of the embeddings that project onto the gender subspace. We use the softer variation to debias the embeddings while training our language model.

**Zhao et al. (2017)** look at gender bias in the context of using structured prediction for visual object classification and semantic role labeling. They observe gender bias in the training examples and that their model amplifies the bias in its predictions. They impose constraints on the optimization to reduce bias amplification while incurring minimal degradation in their model’s performance.

Word embeddings can capture the stereotypical bias in human generated text leading to biases in NLP Applications. Caliskan et al. (2017) conduct Word Embedding Association Test (WEAT). It is based on the hypothesis that word embeddings closer together in high dimensional space are semantically closer. They find strong evidence of social biases in pretrained word embeddings.

**Rudinger et al. (2018)** introduce Winogender schemas\(^1\) and evaluate three coreference resolution systems—rule-based, statistical and neural systems. They find that these systems’ predictions strongly prefer one gender over the other for occupations.

Font and Costa-Jussà (2019) study the impact of gender debiasing techniques by Bolukbasi et al. (2016) and Zhao et al. (2018) in machine translation. They find these methods to be effective, and even a noted BLEU score improvement for the debiased model. Our work is closely related but while they use debiased pretrained embeddings, we train the word embeddings from scratch and debias them while the language model is trained.

**May et al. (2019)** extend WEAT to state-of-the-art sentence encoders: the Sentence Encoder Association Test (SEAT). They show that these tests can provide an evidence for presence of bias.

---

\(^1\)It is Winograd Schema-style coreference dataset consisting of pair of sentences that differ only by a gender pronoun.
However, the cosine similarity between sentences can be an inadequate measure of text similarity in sentences. In this paper, we attempt to minimize the cosine similarity between word embeddings and gender direction.

Gonen and Goldberg (2019) conduct experiments using the debiasing techniques proposed by Bolukbasi et al. (2016) and Zhao et al. (2018). They show that bias removal techniques based on gender direction are inefficient in removing all aspects of bias. In a high dimensional space, spatial distribution of the gender neutral word embeddings remain almost same after debiasing. This enables a gender-neutral classifier to still pick up the cues that encode other semantic aspects of bias. We use softer variation of the debiasing method proposed by Bolukbasi et al. (2016) and attempt to measure the debiasing effect from the minimal changes in the embedding space.

3 Methods

We first examine the bias existing in the datasets through qualitative and quantitative analysis of trained embeddings and cooccurrence patterns. We then train an LSTM word-level language model on a dataset and measure the bias of the generated outputs. As shown in Figure 1, we then apply a regularization procedure that encourages the embeddings learned by the model to depend minimally on gender. We debias the input and the output embeddings individually as well as simultaneously. Finally, we assess the efficacy of the proposed method in reducing bias.

We observe that when both input and output embeddings are debiased together, the perplexity of the model shoots up by a much larger number than the input or the output embeddings debiased individually. We report our results when only input embeddings are debiased. This method, however, does not limit the model to capture other forms of bias being learned in other model parameters or output embeddings.

The code implementing our methods can be found in our GitHub repository.2

3.1 Datasets and Text Preprocessing

We compare the model on three datasets—Penn Treebank (PTB), WikiText-2 and CNN/Daily Mail. The first two have been used in language modeling for a long time. We include CNN/Daily Mail dataset in our experiments as it contains a more diverse range of topics.

PTB Penn Treebank comprises of articles ranging from scientific abstracts, computer manuals, etc. to news articles. In our experiments, we observe that PTB has a higher count of male words than female words. Following prior language modeling work, we use the Penn Treebank dataset (PTB; Marcus et al., 1993) preprocessed by Mikolov et al. (2010).

WikiText-2 WikiText-2 is twice the size of the PTB and is sourced from curated Wikipedia articles. It is more diverse and therefore has a more balanced ratio of female to male gender words than PTB. We use preprocessed WikiText-2 (Wikitext-2; Merity et al., 2016).

CNN/Daily Mail This dataset is curated from a diverse range of news articles on topics like sports, health, business, lifestyle, travel etc. This dataset has an even more balanced ratio of female to male gender words and thus, relatively less biased than the above two. However, this does not mean that the use of pronouns is not biased. This dataset was released as part of a summarization dataset by Hermann et al. (2015), and contains 219,506 articles from the newspaper the Daily Mail. We subsample the sentences by a factor of 100 in order to make the dataset more manageable for experiments.

3.2 Word-Level Language Model

We use a three-layer LSTM word-level language model (AWD-LSTM; Merity et al., 2018) with 1150 hidden units implemented in PyTorch.3 These models have an embedding size of 400 and a learning rate of 30.

We use a batch size of 80 for Wikitext-2 and 40 for PTB. Both are trained for 750 epochs. The PTB baseline model achieves a perplexity of 62.56. For WikiText-2, the baseline model achieves a perplexity of 67.67.

For CNN/Daily Mail, we use a batch size of 80 and train it for 500 epochs. We do early stopping for this model. The hyperparameters are chosen through a systematic trial and error approach. The baseline model achieves a perplexity of 67.67.

For CNN/Daily Mail, we use a batch size of 80 and train it for 500 epochs. We do early stopping for this model. The hyperparameters are chosen through a systematic trial and error approach. The baseline model achieves a perplexity of 118.01.

All three baseline models achieve reasonable perplexities indicating them to be good proxies for standard language models.

2https://github.com/BordiaS/language-model-bias

3https://github.com/salesforce/awd-lstm-lm
### 3.3 Quantifying Biases

For numeric data, bias can be caused simply by class imbalance, which is relatively easy to quantify and fix. For text and image data, the complexity in the nature of the data increases and it becomes difficult to quantify. Nonetheless, defining relevant metrics is crucial in assessing the bias exhibited in a dataset or in a model’s behavior.

#### 3.3.1 Bias Score Definition

In a text corpus, we can express the probability of a word occurring in context with gendered words as follows:

\[
P(w|g) = \frac{c(w, g) / \sum_i c(w_i, g)}{c(g) / \sum_i c(w_i)}
\]

where \(c(w, g)\) is a context window and \(g\) is a set of gendered words that belongs to either of the two categories: male or female. For example, when \(g = f\), such words would include \(she, her, woman\), etc. \(w\) is any word in the corpus, excluding stop words and gendered words. The bias score of a specific word \(w\) is then defined as:

\[
\text{bias}_{\text{train}}(w) = \log \left( \frac{P(w|f)}{P(w|m)} \right)
\]

This bias score is measured for each word in the text sampled from the training corpus and the text corpus generated by the language model. A positive bias score implies that a word cooccurs more often with female words than male words. For an infinite context, the words \(doctor\) and \(nurse\) would cooccur as many times with a female gender as with male gender words and the bias scores for these words will be equal to zero.

We conduct two sets of experiments where we define context window \(c(w, g)\) as follows:

**Fixed Context** In this scenario, we take a fixed context window size and measure the bias scores.
We generated bias scores for several context window sizes in the range \((5, 15)\). For a context size \(k\), there are \(k\) words before and \(k\) words after the target word \(w\) for which the bias score is being measured. Qualitatively, a smaller context window size has more focused information about the target word. On the other hand, a larger window size captures topicality (Levy and Goldberg, 2014). By choosing an optimal window of \(k = 10\), we give equal weight of 5\% to the ten words before and the ten words after the target word.

**Infinite Context**  In this scenario, we take an infinite window of context with weights diminishing exponentially based on the distance between the target word \(w\) and the gendered word \(g\). This method emphasizes on the fact that the nearest word has more information about the target word. The farther the context gets away from a word, the less information it has about the word. We give 5\% weight to the words adjacent to the target word as in Fixed Context but reduce the weights of the words following by 5\% and 95\% to the rest; this applied recursively gives a base of 0.95. This method of exponential weighting instead of equal weighting adds to the stability of the measure.

### 3.3.2 Bias Reduction Measures

To evaluate debiasing of each model, we measure the bias for the generated corpus.

\[
bias_{\lambda}(w) = \log \left( \frac{P(w|f)}{P(w|m)} \right)
\]

To estimate the amplification or reduction of the bias, we fit a univariate regression model over bias scores of context words \(w\) as follows:

\[
bias_{\lambda}(w) = \beta \ast bias_{\text{train}}(w) + c
\]

where \(\beta\) is the scaled amplification measure relative to the training data. Reducing \(\beta\) implies debiasing the model.

We also look at the distribution of the bias by evaluating mean absolute bias and deviation in bias scores for each context word in each of the generated corpora.

\[
\mu_{\lambda} = \text{mean}(abs(bias_{\lambda})); \sigma_{\lambda} = \text{stdev}(bias_{\lambda})
\]

We take the mean of absolute bias score as the word can be biased in either of the two directions.

### 3.4 Model Debiasing

Machine learning techniques that capture patterns in data to make coherent predictions can unintentionally capture or even amplify the bias in data (Zhao et al., 2017). We consider a gender sub-

space present in the learned embedding matrix in our model as introduced in the Bolukbasi et al. (2016) paper. We train these embeddings on the word level language model instead of using the debiased pretrained embeddings (Font and Costa-

Jussà, 2019). We conduct experiments for the three cases where we debias—input embeddings, output embeddings, and both the embeddings simultaneously.

Let \(w \in S_W\) be a word embedding corresponding to a word in the word embedding matrix \(W\). Let

\[
D_1, \ldots, D_n \subset S_W
\]

be the defining sets\(^4\) that contain gender-opposing words, e.g. man and woman. The defining sets are designed separately for each corpus since certain words may not appear in another corpus. We consider it a defining set if both gender-opposing words occur in the training corpus.

If \(u_i, v_i\) are the embeddings corresponding to the words man and woman, then \(\{u_i, v_i\} = D_1\). We consider the matrix \(C\) which is defined as a stack of difference vectors between the pairs in the defining sets. We have:

\[
C = \begin{pmatrix}
\left( \frac{u_1 - v_1}{2} \right) \\
\vdots \\
\left( \frac{u_n - v_n}{2} \right)
\end{pmatrix} = U \Sigma V
\]

The difference between the pairs encodes the gender information corresponding to the gender pair. We then perform singular value decomposition on \(C\), obtaining \(U\Sigma V\). The gender subspace \(B\) is then defined as the first \(k\) columns (where \(k\) is chosen to capture 50\% of the variation) of the right singular matrix \(V\):

\[
B = V_{1:k}
\]

Let \(N\) be the matrix consisting of the embeddings for which we would like the corresponding words to exhibit unbiased behavior. If we want the embeddings in \(N\) to have minimal bias, then its projection onto the gender subspace \(B\) should be small in terms its the squared Frobenius norm.

\(^4\)See the supplement for corpus-wise defining sets
<table>
<thead>
<tr>
<th>Target Word</th>
<th>( \lambda )</th>
<th>Sample From Generated Text</th>
</tr>
</thead>
<tbody>
<tr>
<td>crying</td>
<td>0.0</td>
<td>“she was put on her own machine to raise money for her own wedding &lt;unk&gt; route which saw her crying and &lt;unk&gt; down a programme today . effects began by bottom of her marrow the &lt;unk&gt;”</td>
</tr>
<tr>
<td></td>
<td>0.5</td>
<td>“he &lt;unk&gt; in the americas with the &lt;unk&gt; which can spread a &lt;unk&gt; circumcision ceremony made last month . as he &lt;unk&gt; his mother s &lt;unk&gt; crying to those that”</td>
</tr>
<tr>
<td></td>
<td>1.0</td>
<td>“he discovered peaceful facebook remains when he was caught crying officers but was arrested after they found the crash hire a man &lt;unk&gt; brown shocked his brother &lt;unk&gt; over”</td>
</tr>
<tr>
<td>fragile</td>
<td>0.0</td>
<td>“camilla said she talked to anyone and had previously left her love of two young children . it all comes with her family in conviction of her son s death . it s been fragile . the &lt;unk&gt; and retail boy that was rik s same maker identified genuinely &lt;unk&gt; attacked all”</td>
</tr>
<tr>
<td></td>
<td>0.5</td>
<td>“his children at nearby children s hospital in &lt;unk&gt; and went &lt;unk&gt; years after he was arrested on &lt;unk&gt; bail , she spent relaxed weeks in prison after being sharply in fragile &lt;unk&gt; while she was jailed and strangled when she was born in &lt;unk&gt; virginia”</td>
</tr>
<tr>
<td></td>
<td>1.0</td>
<td>“could they possibly have a big barrier to jeff &lt;unk&gt; and &lt;unk&gt; my son all intelligence period that will contain the east country s world from all in the world the truth is when we moved clear before the split twenty days earlier that day . none of the distributed packs on the website can never &lt;unk&gt; re able to &lt;unk&gt; it the second time so that fitting fragile &lt;unk&gt; are and less the country is &lt;unk&gt; . it came as it was once &lt;unk&gt; million lead jobs mail yorkshire . adoption of these first product is olio but it is currently almost impossible for the moon to address and fully offshore hotly”</td>
</tr>
<tr>
<td>leadership</td>
<td>0.0</td>
<td>“mr &lt;unk&gt; worked traditions at the squadron base in &lt;unk&gt; rbs to marry the us government .he referring to the mainland them in february &lt;unk&gt; he kept communist leadership from undergoing”</td>
</tr>
<tr>
<td></td>
<td>0.5</td>
<td>“obama s first wife janet had a chance to run the opposition for a superbowl event for charity the majority of the south african people s travel stage &lt;unk&gt; leadership while it was married off christmas”</td>
</tr>
<tr>
<td></td>
<td>1.0</td>
<td>“the woman s lungs and drinking the ryder of his daughters s leadership morris said businesses . however being of his mouth around wiltshire and burn talks from the hickey s &lt;unk&gt; employees”</td>
</tr>
<tr>
<td>prisoner</td>
<td>0.0</td>
<td>“his legs and allegedly killed himself by suspicious points . in the latest case after an online page he left prisoner in his home in &lt;unk&gt; near &lt;unk&gt; manhattan on saturday when he was struck in his car operating in &lt;unk&gt; bay smoking &lt;unk&gt; and &lt;unk&gt; &lt;unk&gt; when he had”</td>
</tr>
<tr>
<td></td>
<td>0.5</td>
<td>“it is something that the medicines can target prisoner and destroy &lt;unk&gt; firms in the uk but i hope that there are something into the on top getting older people who have more branded them as poor .”</td>
</tr>
<tr>
<td></td>
<td>1.0</td>
<td>“the ankle follows a worker &lt;unk&gt; her &lt;unk&gt; prisoner she died this year before now an profile which clear her eye borrowed for her organ own role . it was a huge accident after the drugs she had”</td>
</tr>
</tbody>
</table>

Table 4: Generated text comparison for CNN/Daily Mail for different \( \lambda \) values

Therefore, to reduce the bias learned by the embedding layer in the model, we can add the following bias regularization term to the training loss:

\[
\mathcal{L}_B = \lambda \|NB\|^2_F 
\]

where \( \lambda \) controls the importance of minimizing bias in the embedding matrix \( W \) (from which \( N \) and \( B \) are derived) relative to the other components of the model loss. The matrices \( N \) and \( C \) are updated each iteration during the model training.

We input 2000 random seeds in the language model as starting points to start word generation. We use the previous words as an input to the language model and perform multinomial selection to generate up the next word. We repeat this up to 500 times. In total, we generate \( 10^6 \) tokens for all three datasets for each \( \lambda \) and measure the bias.

4 Experiments

4.1 Model

After achieving the baseline results, we run experiments to tune \( \lambda \) as hyperparameter. We report an in-depth analysis of bias measure on the models with debiased input embeddings.

4.2 Results and Text Examples

We calculate the measures stated in Section 3.3 for the three datasets and the generated corpora using the corresponding RNN models. The results are shown in Tables 1, 2 and 3. We see that the \( \mu \) consistently decline as we increase \( \lambda \) until a point, beyond which the model becomes unstable. So there is a scope of optimizing the \( \lambda \) values. The detailed analysis is presented in Section 4.3.

Table 4 shows excerpts around selected target words from the generated corpora to demonstrate the effect of debiasing for different values of \( \lambda \). We highlight the words crying and fragile that are typically associated with feminine qualities, along
with the words leadership and prisoners that are stereotyped with male identity. These biases are reflected in the generated text for $\lambda = 0$. We notice increased mention of the less probable gender in the subsequent generated text with debiasing ($\lambda = 0.5, 1.0$). For fragile, the generated text at $\lambda = 1.0$ has reduced the mention of stereotyped female words but had no mentions of male words; resulting in a large chunk of neutral text. Similarly, in prisoners, the generated text for $\lambda = 0.5$ has no gender words.

However, these are small snippets and the bias scores presented in the supplementary table quantifies the distribution of gender words around the target word in the entire corpus. These target words are chosen as they are commonly perceived gender biases and in our study, they show prominent debiasing effect.\(^5\)

### 4.3 Analysis and Discussion

We consider a text corpus to be biased when it has a skewed distribution of words cooccurring with one gender vs another. Any dataset that has such demographic bias can lead to (potentially unintended) social exclusion (Hovy, 2015). PTB and WikiText-2 consist of news articles related to business, science, politics, and sports. These are all male dominated fields. However, CNN/Daily Mail consists of articles across diverse set of categories like entertainment, health, travel etc. Among the three corpora, Penn Treebank has more frequent mentions of male words with respect to female words and CNN/Daily Mail has the least.

As defined, bias score of zero implies perfectly neutral word, any value higher/lower implies female/male bias. Therefore, the absolute value of bias score signifies presence of bias. Overall bias in a dataset can be estimated as the average of absolute bias score ($\mu$). The aggregated absolute bias scores $\mu$ of the three datasets—Penn Treebank, WikiText-2, and CNN/Daily Mail—are 0.83, 0.80, and 0.72 respectively. Higher $\mu$ value in this measure means on-an-average the words in the entire corpus are more gender biased. As per the Tables 1, 2, and 3, we see that the $\mu$ consistently decline as we increase $\lambda$ until a point, beyond which the model becomes unstable. So there is a scope of optimizing the $\lambda$ values.

The second measure we evaluated is the standard deviation ($\sigma$) of the bias score distribution. Less biased dataset should have the bias score concentrating closer to zero and hence lower $\sigma$ value. We consistently see that, with the initial increase of $\lambda$, there is a decrease in $\sigma$ of the bias score distribution.

The final measure to evaluate debiasing is comparison of bias scores at individual word level. We regress the bias scores of the words in generated text against their bias scores in the training corpus after removing the outliers. The slope of regression $\beta$ signifies the amplification or dampening effect of the model relative to the training corpus. Unlike the previous measures, this measure gives clarity at word level bias changes. A drop in $\beta$ signifies reduction in bias and vice versa. A negative $\beta$ signifies inversion in bias assuming there are no other effects of the loss term. In our experiments, we observe $\beta$ to increase with higher values of $\lambda$ possibly due to instability in model and none of those values go beyond 1.

We observe that corpus level bias scores like $\mu$, $\sigma$ are less effective measures to study efficacy of debiasing techniques because they fail to track the improvements at word level. Instead, we recommend a word level score comparison like $\beta$ to evaluate robustness of debiasing at corpus level.

To choose the context window in a more robust manner, we take exponential weightings to the cooccurrences. The results for aggregated average of absolute bias and standard deviation show the same pattern as in fixed context window.

As shown in the results above, we see that the standard deviation ($\sigma$), absolute mean ($\mu$) and slope of regression ($\beta$) reduce for smaller $\lambda$ relative to those in training data and then increase with $\lambda$ to match the variance in the original corpus. This holds for the experiments conducted with fixed context window as well as with exponential weightings.

### 5 Conclusion

In this paper, we quantify and reduce gender bias in word level language models by defining a gender subspace and penalizing the projection of the word embeddings onto that gender subspace. We devise a metric to measure gender bias in the training and the generated corpus.

In this study, we quantify corpus level bias in two different metrics—absolute mean ($\mu$) and standard deviation ($\sigma$). However, for evaluating debiasing effects, we propose a relative metric ($\beta$)
to study the change in bias scores at word level in generated text vs. training corpus. To calculate $\beta$, we conduct an in-depth regression analysis of the word level bias measures in the generated text corpus over the same for the training corpus.

Although we found mixed results on amplification of bias as stated by Zhao et al. (2017), the debiasing method shown by Bolukbasi et al. (2016) was validated with the use of novel and robust bias measure designed in this paper. Our proposed methodology can deal with distribution of words in a vocabulary in word level language model and it targets one way to measure bias, but it’s highly likely that there is significant bias in the debiased models and data, just not bias that we can detect on this measure. It can be concluded different bias metrics show different kinds of bias (Gonen and Goldberg, 2019).

We additionally observe a perplexity bias trade-off as a result of the additional bias regularization term. In order to reduce bias, there is a compromise on perplexity. Intuitively, as we reduce bias the perplexity is bound to increase due to the fact that, in an unbiased model, male and female words will be predicted with an equal probability.

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References


Emotion Impacts Speech Recognition Performance

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Abstract

It has been established that the performance of speech recognition systems depends on multiple factors including the lexical content, speaker identity and dialect. Here we use three English datasets of acted emotion to demonstrate that emotional content also impacts the performance of commercial systems. On two of the corpora, emotion is a bigger contributor to recognition errors than speaker identity and on two, neutral speech is recognized considerably better than emotional speech. We further evaluate the commercial systems on spontaneous interactions that contain portions of emotional speech. We propose and validate on the acted datasets, a method that allows us to evaluate the overall impact of emotion on recognition even when manual transcripts are not available. Using this method, we show that emotion in natural spontaneous dialogue is a less prominent but still significant factor in recognition accuracy.

1 Introduction

Alexa and Google Home are becoming increasingly popular, their use spanning a range of applications from reducing loneliness in the elderly (Reis et al., 2017; Ferland et al., 2018) to child entertainment and education (Druga et al., 2017). As these conversational agents become commonplace, people are likely to express emotion during their interactions, either because of their perception of the agent or because of the emotion-eliciting situations in which the agent is deployed.

In this paper, we set out to study the extent to which emotional content in speech impacts speech recognition performance of commercial systems. Similar studies have been conducted in the past to study how recognition varies with gender and dialect (Adda-Decker and Lamel, 2005; Tatman, 2017), lexical content (Goldwater et al., 2010), topical domain (Traum et al., 2015) and delivery style (Siegler and Stern, 1995; Nakamura et al., 2008). A number of studies have studied the impact of stress (Hansen and Patil, 2007; Bou-Ghazale and Hansen, 2000; Steeneken and Hansen, 1999; Hansen, 1996) and emotional factors (Polzin and Waibel, 1998; Kostoulas et al., 2008; Benzeghiba et al., 2007) on speech recognition. Multiple studies (Byrne et al., 2004; Athanaselis et al., 2005; Pan et al., 2006; Meng et al., 2007; Ijima et al., 2009; Sun et al., 2009; Sheikhan et al., 2012) tried to improve upon speech recognition accuracies for emotional speech. However, these studies were carried out with older recognition systems. Recently automatic speech recognition has seen unprecedented gains in accuracy. Yet our work shows that emotional content still poses problems to speech recognition systems.

In our work we seek to quantify the influence of emotion on recognition accuracy for three commercial systems, on several datasets. We start out with two datasets of acted emotion, which are in some respects ideal for the task because the spoken content is constrained to pre-selected utterances and thus manual transcription is not required. In addition, the lexical content for each emotion is identical, so no special adjustment for that confounding factor is needed in the acted corpora.

At the same time, it is important to validate these results on spontaneous, more natural exchanges, so we also present results on such a corpus of emotion in spontaneous speech. As the speech becomes more natural, it becomes harder to obtain large manual transcripts for a large portion of the data to carry out the studies that we present, so we also validate an alternative method for finding factors that influence the performance of commercial systems, relying on agreement between systems rather than manual transcripts. We present convincing evidence that the approach is a...
reasonable approximation and it can be used for broader studies on factors influencing automatic speech recognition. Here, we apply the method to analyze data from a spontaneous emotional speech corpus.

2 Related Work

There has been much research in the field of emotion recognition from speech. However, relatively less research has been conducted on how emotion affects Automatic Speech Recognition (ASR).

(Polzin and Waibel, 1998) study the variation in word error rate with different emotions - NEUTRAL, ANGER, HAPINESS, AFRAID and SADNESS. They observed that SADNESS and AFRAID/FEAR perform worst while NEUTRAL and ANGER perform best. They integrate prosodic features into the model using Hidden Markov Models to first disambiguate the emotional state of the speaker, and then use emotion specific ASR models for transcription. They report a significant increase in ASR performance.

(Kostoulas et al., 2008) conduct a similar study over a much wider range of emotions (About 15 emotions) on the Wall Street Journal database with Sphinx III as the ASR system and report a large variance in the WER across emotions, ranging from about 6% for NEUTRAL to about 44% for HOT ANGER.

(Athanaselis et al., 2005) extract an emotionally colored subset of the British National Corpus (BNC) and append it multiple times to the BNC before training an emotionally-enhanced ASR system.

(Sheikhan et al., 2012) propose that the emotion in speech leads to changes in Mel-frequency cepstral coefficients (MFCC) and thus propose neutralizing MFCCs by warping the first three formant frequencies and conduct their experiments to analyze improvement in ASR performance for the emotions ANGER and HAPINESS.

In this work, we analyze the performance of multiple modern commercial ASR systems on emotional speech. We further quantify the correlation between the Word Error Rate and emotion. We also compare the dependence of ASR performance on other factors - speaker identity and spoken content with the dependence on emotion.

3 Datasets and APIs

We use three acted Emotion datasets: CREMA-D (Cao et al., 2014), RAVDESS (Steven R. Livingston1, 2018) and MSP-IMPROV (Busso et al., 2017). CREMA-D has 12 sentences recorded by 91 actors in 6 different emotions (Anger, Disgust, Fear, Happy, Neutral and Sad), for a total of 7,442 utterances in the dataset.

RAVDESS has just two sentences, which are very similar to each other (Kids are talking by the door and Dogs are sitting by the door), recorded by 24 actors in 8 emotions with the addition of Surprised and Calm. RAVDESS has a total of 1,440 utterances, and each sentence is recorded in two intensities with two repetitions of each.

In addition to actors recording sentences in a pre-specified emotion, the MSP-IMPROV dataset contains ‘improvised recordings’, where actors converse to induce the desired emotion. In these interactions, there is at least one emotional rendition of the target utterance but other utterances may be emotionally neutral. MSP-IMPROV is comprised of 1,272 utterances distributed over 20 target sentences and four emotions (Neutral, Anger, Happy, Fear). We refer to this part of the corpus as MSP-IMPROV Target, where we only concern ourselves with recognizing the target sentence. We refer to the set of complete conversations as the MSP-IMPROV Dialogue Corpus. Manual transcripts are not available for this part. MSP-IMPROV has 1,085 complete conversations.

Commercial systems used for speech recognition are IBM Watson Speech-to-Text, Google Cloud Speech-to-Text and Amazon Transcribe. 2. We will denote the APIs simply by IBM, GCP and AWS respectively.

4 Evaluation Metrics

We report two measures of automatic speech recognition performance: the Word Error Rate (WER) and the percentage of completely recognized sentences (CR). Minor semantic and grammatical errors are ignored by manually listing semantically equivalent sentences for computing CR. We report $1 - CR$ instead of $CR$ to maintain consistency with WER interpretation, lower

1One of the sentences is recorded in three different intensities.

We then perform an ANOVA analysis to determine the statistical significance of each factor in determining the WER.

The dialogues in the MSP-IMPROV corpus do not contain manual transcripts. We propose a metric to analyze the relative performance of different systems with varying emotions when manual transcripts are not available.

We calculate the performance of every system relative to other systems and then report the average of these cross-comparisons. This method accurately predicts the relative performance on emotional and neutral speech consistent with WER/CR results on the other corpora for which transcripts are available.

<table>
<thead>
<tr>
<th>Dataset</th>
<th>Metric</th>
<th>IBM</th>
<th>AWS</th>
<th>GCP</th>
</tr>
</thead>
<tbody>
<tr>
<td>CREMA-D</td>
<td>WER</td>
<td>10.00</td>
<td>13.09</td>
<td>18.80</td>
</tr>
<tr>
<td></td>
<td>1-CR</td>
<td>24.72</td>
<td>39.20</td>
<td>41.78</td>
</tr>
<tr>
<td>RAVDESS</td>
<td>WER</td>
<td>5.08</td>
<td>13.31</td>
<td>6.19</td>
</tr>
<tr>
<td></td>
<td>1-CR</td>
<td>9.17</td>
<td>56.38</td>
<td>15.49</td>
</tr>
<tr>
<td>MSP-IMPROV</td>
<td>WER</td>
<td>13.90</td>
<td>9.21</td>
<td>12.56</td>
</tr>
<tr>
<td></td>
<td>1-CR</td>
<td>38.76</td>
<td>35.72</td>
<td>39.54</td>
</tr>
</tbody>
</table>

Table 1: Overall Performance of IBM, AWS and GCP

5 Observations

5.1 Variations across various factors

The overall performance of the APIs on the three datasets is given in Table 1. IBM performs best on 2 out of 3 datasets (CREMA-D and RAVDESS). AWS is, however, more consistent across datasets.

Figure 1 shows how WER varies with emotion. On CREMA-D, NEUTRAL speech is recognized more accurately than emotional speech, with ANGER most accurately recognized among the emotions, while SADNESS and FEAR are poorly recognized. Similarly on RAVDESS, FEAR has the worst WER. NEUTRAL and ANGER are recognized better than other emotions. On MSP-IMPROV Target, ANGER is recognized most accurately, followed by NEUTRAL speech. Overall NEUTRAL utterances for all datasets are more accurately recognized than the combined class of emotional speech. Further, there is a high variation in performance between different emotions. Improving performance while focusing on poorly performing emotions like sadness and fear, which have an extremely bad performance, will help improve speech recognition.

<table>
<thead>
<tr>
<th>Corpus</th>
<th>Pearson</th>
<th>Spearman</th>
</tr>
</thead>
<tbody>
<tr>
<td>CREMA-D</td>
<td>0.73</td>
<td>0.86</td>
</tr>
<tr>
<td>RAVDESS</td>
<td>0.82</td>
<td>0.93</td>
</tr>
<tr>
<td>MSP-IMPROV</td>
<td>0.74</td>
<td>0.84</td>
</tr>
<tr>
<td>Target</td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

Table 2: Spearman and Pearson Correlation between the cross-comparison WER and the observed WER

Performance varies largely with sentences—some show excellent performance while others do not. Performance also varies across APIs—sentences with good performance with one system may perform well with others.

RAVDESS has two similar sentences, and hence it does not make sense to look at performance variation with spoken content on RAVDESS. On MSP-IMPROV Target, the WER varies from lower than 5% for some sentences to above 30% for others. Similar variations are also observed with Speaker Identity.

<table>
<thead>
<tr>
<th>Corpus</th>
<th>IBM</th>
<th>IBM</th>
<th>AWS</th>
<th>AWS</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>WER</td>
<td>GCP</td>
<td>WER</td>
<td>GCP</td>
</tr>
<tr>
<td>CREMA-D</td>
<td>0.84</td>
<td>0.74</td>
<td>0.85</td>
<td></td>
</tr>
<tr>
<td>RAVDESS</td>
<td>0.84</td>
<td>0.63</td>
<td>0.84</td>
<td></td>
</tr>
<tr>
<td>MSP-IMPROV</td>
<td>0.67</td>
<td>0.69</td>
<td>0.65</td>
<td></td>
</tr>
<tr>
<td>Target</td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

Table 3: Spearman Correlation between two-system cross comparison WER and the observed WER

5.2 Evaluating Performance without Manual Transcripts

When manual transcripts are not available, we treat the output of one API as the reference and get WER for other APIs with respect to it. We refer to this as the cross-comparison WER. We then change the reference API and repeat the process. The performance is reported as the average of all cross-reference WERs. In our case, we use three API’s. Thus the average cross-comparison WER is the average of 9 cross-comparison WERs.

We test our metric on CREMA-D, RAVDESS, and MSP-IMPROV Target by computing Spearman and Pearson’s correlations between the true WER and the average of the nine cross-comparison WERs. The correlations, mentioned in Table 2, are high, all above 0.7, indicating that the approximation is not perfect but overall accurate.
We also conduct experiments to check whether the metric can be based on two systems instead of three. The Spearman Correlations are tabulated in Table 3. The two system cross-comparison metric is representative of the WER but the correlation is not as strong.

In Table 5, we report cross-comparison WER on the MSP-IMPROV Dialogue subcorpus which includes conversations used to evoke the desired emotion for the target sentence so that the required emotion sounds natural rather than acted. Each conversation is between a male and a female speaker. Emotions in natural speech are not as intense as they are in the acted versions. Also, only parts of the conversations contain emotional speech (neutral speech with parts of emotional speech) and it is natural to expect that the influence on recognition rates will be attenuated. It is however still present: ANGER and HAPPINESS have much worse recognition than NEUTRAL speech. Here however the best recognition is for SAD speech.

<table>
<thead>
<tr>
<th>Emotion</th>
<th>Cross-Comparison WER</th>
</tr>
</thead>
<tbody>
<tr>
<td>SAD</td>
<td>18.84</td>
</tr>
<tr>
<td>NEU</td>
<td>20.73</td>
</tr>
<tr>
<td>ANG</td>
<td>23.29</td>
</tr>
<tr>
<td>HAP</td>
<td>23.47</td>
</tr>
<tr>
<td>Overall</td>
<td>21.45</td>
</tr>
</tbody>
</table>

Table 5: Cross-comparison WER for MSP-IMPROV Dialogue

6 Statistical Significance of Emotion, Speaker and Spoken Content

We now report the results of ANOVA analysis on each of the datasets, to compare the statistical significance of emotion, speaker and spoken content (sentence identity in our case) on performance. We compute the WER for each sentence separately. The F-values and P-values of the above-mentioned factors are listed in Table 4. As expected, spoken content has the highest impact on performance, other than RAVDESS which is expected to have low F-value for spoken-content. On CREMA-D and RAVDESS, Emotion impacts performance more than Speaker Identity.
On CREMA-D, the F-value for spoken content is about 40 times that of Emotion. Nevertheless, the Emotion and Actor Identity factors are statistically significant. Emotion has a much larger impact than actor identity. On RAVDESS, Emotion is slightly more impactful than Speaker Identity.

On MSP-IMPROV Target, the impact of Speaker Identity is more pronounced. However, on splitting the corpus based on whether the samples were improvised or not, on improvised speech, Emotion has a higher impact than actor identity. For non-improvised speech, Speaker Identity becomes important. Recognizing improvised speech, which is closer to natural speech, is more difficult. The impact of Actor Identity is thus lower (in improvised speech) than non-improvised speech. Note that the WER is higher for improvised speech (13.5%) compared to non-improvised speech (10.4%). Surprisingly, the impact of Emotion is higher in improvised speech.

For dialogues, Spoken Content has low significance, likely because factors are averaged out. Actor identity and gender of the interlocutor (together) impact recognition most.

7 Conclusions

We quantified the impact of Emotion on speech recognition performance. We developed a metric to analyze performance for audio samples where manual transcripts are unavailable and showed empirically that this metric works. In future work, we plan to analyze whether acoustic features are predictive of what sentences are likely to be misrecognized and the characteristic features per emotion.

References


The Strength of the Weakest Supervision: Topic Classification Using Class Labels

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Abstract

When developing topic classifiers for real-world applications, we begin by defining a set of meaningful topic labels. Ideally, an intelligent classifier can understand these labels right away and start classifying documents. Indeed, a human can confidently tell if a news article is about science, politics, sports, or none of the above, after knowing just the class labels.

We study the problem of training an initial topic classifier using only class labels. We investigate existing techniques for solving this problem and propose a simple but effective approach. Experiments on a variety of topic classification data sets show that learning from class labels can save significant initial labeling effort, essentially providing a “free” warm start to the topic classifier.

1 Introduction

When developing topic classifiers for real-world tasks, such as news categorization, query intent detection, and user-generated content analysis, practitioners often begin by crafting a succinct definition, or a class label, to define each class. Unfortunately, these carefully written class labels are completely ignored by supervised topic classification models. Given a new task, these models typically require a significant amount of labeled documents to reach even a modest initial performance. In contrast, a human can readily understand new topic categories by reading the class definitions and making connections to prior knowledge. Labeling initial examples for every new task can be time-consuming and labor-intensive, especially in resource-constrained domains like medicine and law. Therefore it is desirable if a topic classifier can proactively interpret class labels before the training starts, giving itself a “warm start”. An imperfect initial model can always be fine-tuned with more labeled documents.

As conceptually shown in Figure 1, a warm start can reduce the total number of training labels for a classifier to reach certain performance level.

In this work, we study algorithms that can initialize a topic classifier using class labels only. Since class labels are the starting point of any topic classification task, they can be viewed as the earliest hence weakest supervision signal. We propose a simple and effective approach that combines word embedding and naive Bayes classification. On six topic classification data sets, we evaluate a suite of existing approaches and the proposed approach. Experimental results show that class labels can train a topic classifier that generalizes as well as a classifier trained on hundreds to thousands of labeled documents.

2 Related Work

Text retrieval. Classifying documents by short labels can be viewed as evaluating textual similarity between a document and a label. Baeza-Yates et al. (2011) called this approach “naive text classification”. Treating labels as search queries, we can classify a document into a class if it best matches the label of that class. Well-studied text retrieval methods, such as vector space models and probabilistic models (Croft et al., 2010), can produce matching scores. To mitigate vocabulary mismatch, such a classifier can be further enhanced...
by self-training: the classifier assigns pseudo labels to top-ranked documents as done in pseudo relevance feedback (Rocchio, 1965), and updates itself using those labels.

**Semi-supervised learning.** Our problem setting can be seen as an extreme case of weak supervision: we only use class labels as the (noisy) supervision signal, and nothing else. If we view class labels as “labeled documents”, one from each class, and to-be-classified documents as unlabeled documents, then we cast the problem as semi-supervised learning (Zhu, 2006). Self-training is one such technique: a generative classifier is trained using only class labels, and then teaches itself using its own predictions on unlabeled data. If we view class labels as “labeled features”, then we expect the classifier to predict a class when a document contains the class label words. For instance, Druck et al. (2008) proposed generalized expectation criteria that uses feature words (class labels) to train a discriminative classifier. Jagarlamudi et al. (2012) and Hingmire and Chakraborti (2014) proposed Seeded LDA to incorporate labeled words/topics into statistical topic modeling. The inferred document-topic mixture probabilities can be used to classify documents.

**Zero-shot learning** aims to classify visual objects from a new class using only word descriptions of that class (Socher et al., 2013). It first learns visual features and their correspondence with word descriptions, and then constructs a new classifier by composing learned features. Most research on zero-shot learning focuses on image classification, but the same principle applies to text classification as well (Pushp and Srivastava, 2017). Our proposed method constructs a new classifier by composing learned word embeddings in a probabilistic manner. Since the new classifier transfers semantic knowledge in word embedding to topic classification tasks, it is broadly related to transfer learning (Pan and Yang, 2010). The main difference is that in transfer learning the information about the new task is in the form of labeled data, not class definition words.

### 3 Proposed Method

Let a test document \( x \) be a sequence of words \((w_1, \ldots, w_j, \ldots)\), and a class topic description \( y \) be a sequence of words \( d_y = (w_1, \ldots, w_y, \ldots)\). All words are in vocabulary \( V \). We propose a generative approach, where the predictive probability \( p(y|x) \propto p(x|y)p(y) \). Generative approaches tend to perform well when training data is scarce, which is the case in our setting.

We assume there exists weak prior knowledge on which classes are popular and which are rare. We can then construct rough estimates \( \hat{p}(y) \) using simple heuristics as described in (Schapire et al., 2002). It distributes probability mass \( q \) evenly among majority classes, and \( 1 - q \) evenly among minority classes. We treat the most frequent class as the majority class, the rest as minority classes, and \( q = 0.7 \) in our experiments.

By interpreting class topic description as words, we obtain \( \hat{p}(x|y) = p(x|d_y) \). We assume that the \( d_y \) expresses a noisy-OR relation of the words it contains (Oni´sko et al., 2001). Up to first-order approximation:

\[
p(x|d_y) = 1 - \prod_{w_j \in d_y} (1 - p(x|w_j)) \approx \sum_{w_j \in d_y} p(x|w_j),
\]

where each \( w_j \) is a word in the class topic description \( d_y \). Further, we assume that words in document \( x \) are conditionally independent given a label word \( w_y \) (naive Bayes assumption):

\[
p(x|w_y) = \prod_{w_j \in x} p(w_j|w_y).
\]

Combining (1) and (2), the document likelihood is

\[
\hat{p}(x|y) = \sum_{w_y \in d_y} \prod_{w_j \in x} p(w_j|w_y).
\]

To this end, we need a word association model \( p(w_1|w_2), \forall w_1, w_2 \in V \). It can be efficiently learned by word embedding algorithms. The skip-gram algorithm (Mikolov et al., 2013) learns vector representations of words, such that for words \( w_1, w_2 \), their vectors \( u_{w_1}, v_{w_2} \) approximate the conditional probability\(^1\)

\[
p(w_1|w_2) = \frac{\exp(w_1^\top v_{w_2})}{\sum_{w \in V} \exp(w_1^\top v_w)}.
\]

\(^1\) The two sets of word vectors \( \{u_w : w \in V\} \) and \( \{v_w : w \in V\} \) produced by skip-gram correspond to the input and output parameters of a two-layer neural network. Typically, only the output parameters are used as the “learned word vectors”. Here we need both input and output parameters to compute \( p(w_1|w_2) \).
Combining (3) with (4), the document likelihood becomes
\[
\hat{p}(x|y) = \sum_{w_y \in d_y} \exp \left( \sum_{w_j \in x} (u^T w_j v_{wy} - C_{wy}) \right),
\]
where \(C_{wy} = \log \sum_{w \in \mathcal{V}} \exp (u^T v_{wy})\) is independent of document \(x\) and only related to label word \(w_y\), therefore can be precomputed and stored to save computation.

Finally, we construct a generative classifier as \(\hat{p}(y|x) \propto \hat{p}(x|y)\hat{p}(y)\). We call this method word embedding naïve Bayes (WENB).

### 3.1 Continued Training

The proposed method produces pseudo labels \(\hat{p}(y|x_j)\) for unlabeled documents \(\{x_j\}_{j=1}^m\). When true labels \(\{(x_i, y_i)\}_{i=1}^n\) are available, we can train a new discriminative logistic regression classifier \(p_0(y|x)\) using both true and pseudo labels (\(\theta\) is the model parameter):

\[
J(\theta) = \sum_{i=1}^n \sum_{y \in \mathcal{Y}} -1_{\{y_i = y\}} \log p_0(y|x_i) + \lambda \|\theta\|^2 \\
+ \mu \sum_{j=1}^m \sum_{y \in \mathcal{Y}} -\hat{p}(y|x_j) \log p_0(y|x_j). \tag{5}
\]

To find the balance of pseudo vs. true labels in (5), we search the hyperparameter \(\mu\) on a 5-point grid \(\{10^{-2}, 10^{-1}, 0.4, 0.7, 1\}\). We expect pseudo labels to have comparable importance as true labels when \(n\) is small (fine granularity for \(\mu \in [10^{-1}, 1]\)), and their importance will diminish as \(n\) gets large (\(\mu = 10^{-2}\)). \(\mu\) is automatically selected such that it gives the best 5-fold cross-validation accuracy on \(n\) true labels.

### 4 Experiments

We compare a variety of methods on six topic classification data sets. The goals are (1) to study the best classification performance achievable using class labels only, and (2) to estimate the equivalent amount of true labels needed to achieve the same warm-start performance.

#### 4.1 Compared Methods

**Retrieval-based methods.** We use language modeling retrieval function with Dirichlet smoothing (Zhai and Lafferty, 2001) (\(\mu = 2500\)) to match a document to class labels (IR). The top 10 results are then used as pseudo-labeled documents to retrain three classifiers: IR+Roc: a Rocchio classifier (\(\alpha = 1, \beta = 0.5, \gamma = 0\)); IR+NB: a multinomial naïve Bayes classifier (Laplace smoothing, \(\alpha = 0.01\)); IR+LR a logistic regression classifier (linear kernel, \(C = 1\)).

**Semi-supervised methods.** ST-0: the initial self-training classifier using class labels as “training documents” (multinomial naïve Bayes, Laplace smoothing \(\alpha = 0.01\)). ST-1: ST-0 retrained on 10 most confident documents predicted by itself. GE: a logistic regression classifier trained using generalized expectation criteria (Druck et al., 2008). Class labels are used as labeled features. sLDA: a supervised topic model trained using seeded LDA (Jaggaralami et al., 2012). Besides \(k\) seeded topics (\(k\) is the number of classes), we use an extra topic to account for other content in the corpus.

**Word embedding-based methods.** Cosine: a centroid-based classifier, where class definitions and documents are represented as average of word vectors. WENB: The proposed method (Section 3). WENB+LR: a logistic regression classifier trained only on pseudo labels produced by WENB (Section 3.1, \(n = 0\)).

For general domain tasks, we take raw text from English Wikipedia, English news crawl (WMT, 2014), and 1 billion word news corpus (Chelba et al., 2013) to train word vectors. For medical domain tasks, we take raw text from MEDLINE abstracts (NLIM, 2018) to train word vectors. We find 50-dimensional skip-gram word vectors perform reasonably well in the experiments.

### 4.2 Data Sets

We consider six topic classification data sets with different document lengths and application domains. Table 1 summarizes basic statistics of these data sets. Table 4 and 5 in the appendix show actual class labels used in each data set.

<table>
<thead>
<tr>
<th>Data set</th>
<th>Avg word/doc</th>
<th># classes</th>
<th># docs</th>
</tr>
</thead>
<tbody>
<tr>
<td>Wiki Titles</td>
<td>3.1 (1.1)</td>
<td>15</td>
<td>30,000</td>
</tr>
<tr>
<td>News Titles</td>
<td>6.7 (9.5)</td>
<td>4</td>
<td>422,937</td>
</tr>
<tr>
<td>Y Questions</td>
<td>5.0 (2.6)</td>
<td>10</td>
<td>1,460,000</td>
</tr>
<tr>
<td>20 News</td>
<td>101.6 (438.5)</td>
<td>20</td>
<td>18,846</td>
</tr>
<tr>
<td>Reuters</td>
<td>76.5 (117.3)</td>
<td>10</td>
<td>8,246</td>
</tr>
<tr>
<td>Med WSD</td>
<td>202.8 (46.6)</td>
<td>2/task</td>
<td>190/task</td>
</tr>
</tbody>
</table>

Table 1: Statistics of topic classification data sets. Numbers in column “Avg word/doc” are “mean (standard deviation)".

Three long text data sets are (1) 20 News: The well-known 20 newsgroup data set. (2) Reuters. The Reuters-21578 data set (Lewis). We take the articles from the 10 largest topics. (3) Med WSD: The MeSH word sense disambiguation (WSD) data set (Jimeno-Yepes et al., 2011).

Each WSD task aims to tell the sense (meaning) of an ambiguous term in a MEDLINE abstract. For instance, the term “cold” may refer to Low Temperature, Common Cold, or Chronic Obstructive Lung Disease, depending on its context. These senses are used as the class labels. We use 198 ambiguous words with at least 100 labeled abstracts in the data set, and report the average statistics over 198 independent classification tasks.

Although no true labels are used for training, some methods require unlabeled data for retrieval, pseudo-labeling, and re-training. We split unlabeled data into 5 folds, using 4 folds to “train” a classifier and 1 fold for test. We use macro-averaged $F_1$ as the performance metric because not all data sets have a balanced class distribution.

4.3 Results and Discussion

Label savings. Table 2 shows that overall, class labels can train text classifiers remarkably better than majority guess. This is no small feat considering that the classifier has not seen any labeled documents yet. Such performance gain essentially comes “for free”, as any text classification task has to start by defining classes. In Table 3, we report the number of true labels needed for a logistic regression model to achieve the same performance as WENB+LR. The most significant savings happen on short documents: class labels are equivalent to hundreds to thousands of labeled documents at the beginning of the training process.

Effect of document length. On short documents (Wiki Titles, News Titles, Y Questions), leveraging unlabeled data does not help with most semi-supervised methods due to severe vocabulary mismatch. The proposed methods (WENB and WENB+LR) show robust performance, because pretrained word vectors can capture semantic similarity even without any word overlap between a class label and a document. This prior knowledge is essential when documents are short. On long documents (20 News, Reuters, Med WSD), leveraging unlabeled data helps, since long documents have richer content and are more likely to contain not only label words themselves, but also other topic-specific words. Retrieval-based and semi-supervised methods are able to learn these words by exploiting intra-document word co-occurrences.

Performance of other methods. Learning from class labels themselves provides very limited help (IR and ST-0). Using class labels as search queries and labeled documents are closely related: IR and ST-0 perform similarly; so do IR+NB and ST-1. When using class labels as search queries,
Figure 2: Continued training behavior: Atheism vs. Autos. Colored band: ±1 standard deviation.

Figure 3: Continued training behavior: Medical vs. Mideast. Colored band: ±1 standard deviation.

re-ranking (IR+Roc) is less useful than training classifiers (IR+NB and IR+LR). After initial retrieval, training a naïve Bayes classifier is almost always better than a logistic regression classifier (IR+NB vs. IR+LR), demonstrating the power of generative models when supervision signal is sparse. Using class labels as labeled features (GE and sLDA) performs well occasionally (GE on 20 News; sLDA on Y Questions), but not consistently. The Cosine method performs well only on Wiki Titles, the shortest documents, because without supervision, representing a long document as an average of word vectors causes significant information loss. Finally, it is encouraging to see WENB+LR sometimes outperform WENB, as WENB+LR is much smaller than WENB+LR in terms of model size.

4.4 Continued Training and Error Analysis

Figure 2 and 3 compare logistic regression classifiers trained with and without pseudo labels generated by WENB. Note that the classifier trained with pseudo labels (cont. train) has a much lower performance variance than the logistic regression classifier trained only on true labels (LR).

The warm-started classifier can serve as a good starting point for further training. Figure 2 shows a salient warm-start effect on a balanced binary classification task in 20 News. The weight \( \mu \) of pseudo labels increases when true labels are few (initial classifier as an informative prior). As expected, \( \mu \) decreases when true labels become abundant.

Figure 3 shows another binary classification task in 20 News where the warm-start effect is limited. Correspondingly, \( \mu \) quickly diminishes as more true labels are available. With 100 or more true labels, pseudo labels have a negligible weight (\( \mu = 10^{-2} \)). In machine learning terms, these pseudo labels specify an incorrect prior that the model should quickly forget, so that it will not hinder the overall learning process.

A closer investigation reveals that the word vector for \textit{mideast} (the class label of one topic in Figure 3) is not well-trained. This is because in general text corpus, the word \textit{mideast} is rather infrequent compared to commonly used alternatives, such as \textit{middle east}. The word vector of \textit{mideast} is surrounded by other infrequent words or misspellings (such as \textit{hizballah}, \textit{jubeir}, \textit{saudis}, \textit{isreal}) as opposed to more frequent and relevant ones (such as \textit{israel}, \textit{israeli}, \textit{saudi}, \textit{arab}). Since WENB uses the semantic knowledge in word vectors to infer pseudo labels, the quality of class label word vectors will affect the pseudo label accuracy.

5 Conclusion and Future Directions

We studied the problem of training topic classifiers using only class labels. Experiments on six data sets show that class labels can save a significant amount of labeled examples in the beginning. Retrieval-based and semi-supervised methods tend to perform better on long documents, while the proposed method performs better on short documents.

This study opens up many interesting avenues for future work. First, we introduce a new perspective on text classification: can we build a text classifier by just providing a short description of each class? This is a more challenging (but more user-friendly) setup than standard supervised classification. Second, future work can investigate tasks such as sentiment and emotion classification, which are more challenging than topic classification tasks. Third, the two approaches – leveraging unlabeled data (retrieval-based and semi-supervised methods) and leveraging pretrained models (the proposed method) – could be combined to give robust performance on both short and long documents. Finally, we can invite users into the training loop: in addition to labeling documents, users can also revise the class definitions to improve the classifier.
Acknowledgments

We thank the anonymous reviewers for their helpful comments. This work was in part supported by the National Library of Medicine under grant number 2R01LM010681-05. Qiaozhu Mei’s work was supported in part by the National Science Foundation under grant numbers 1633370 and 1620319. Yue Wang would like to thank the support of the Eleanor M. and Frederick G. Kilgour Research Grant Award by the UNC-CH School of Information and Library Science.

References


## A Class Labels Used in Each Data Set

<table>
<thead>
<tr>
<th>Data set</th>
<th>Class labels</th>
</tr>
</thead>
<tbody>
<tr>
<td>News Titles</td>
<td>Business, Technology, Entertainment, Health</td>
</tr>
<tr>
<td>20 News</td>
<td>Atheism, Graphics, Microsoft, IBM, Mac, Windows, Sale, Autos, Baseball, Motorcycles, Hockey, Encrypt, Electronics, Medical, Space, Christian Guns, Mideast, Politics, Religion</td>
</tr>
<tr>
<td>Reuters</td>
<td>Earnings/Forecasts, Mergers/Acquisitions, Crude Oil, Trade, Foreign Exchange, Interest Rates, Money Supply, Shipping, Sugar, Coffee</td>
</tr>
</tbody>
</table>

Table 4: Class labels in 5 topic classification data sets.

<table>
<thead>
<tr>
<th>Task (ambiguous term)</th>
<th>Class labels (senses)</th>
</tr>
</thead>
<tbody>
<tr>
<td>AA</td>
<td>Amino Acids, Alcoholics Anonymous</td>
</tr>
<tr>
<td>ADA</td>
<td>Adenosine Deaminase, American Dental Association</td>
</tr>
<tr>
<td>ADH</td>
<td>Alcohol dehydrogenase, Argpressin</td>
</tr>
<tr>
<td>ADP</td>
<td>Adenosine Diphosphate, Automatic Data Processing</td>
</tr>
<tr>
<td>Adrenal</td>
<td>Adrenal Glands, Epinephrine</td>
</tr>
<tr>
<td>Ala</td>
<td>Alanine, Alpha-Linolenic Acid, Aminolevulinic Acid</td>
</tr>
<tr>
<td>ALS</td>
<td>Antilymphocyte Serum, Amyotrophic Lateral Sclerosis</td>
</tr>
<tr>
<td>ANA</td>
<td>American Nurses’ Association, Antibodies, Antinuclear</td>
</tr>
<tr>
<td>Arteriovenous</td>
<td>Arteriovenous anastomosis procedure, Structure of anatomic-arteriovenous anastomosis</td>
</tr>
<tr>
<td>Arteriovenous Anastomoses</td>
<td></td>
</tr>
<tr>
<td>Astragalus</td>
<td>Talus, Astragalus Plant</td>
</tr>
<tr>
<td>B-Cell Leukemia</td>
<td>B-Cell Leukemia, Chronic Lymphocytic Leukemia</td>
</tr>
<tr>
<td>BAT</td>
<td>Chiroptera, Brown Fat</td>
</tr>
<tr>
<td>BLM</td>
<td>Bloom Syndrome, Bleomycin</td>
</tr>
<tr>
<td>Borrelia</td>
<td>Lyme Disease, Borrelia bacteria</td>
</tr>
<tr>
<td>BPD</td>
<td>Bronchopulmonary Dysplasia, Borderline Personality-Disorder</td>
</tr>
<tr>
<td>BR</td>
<td>Brazil, Bromides</td>
</tr>
<tr>
<td>Brucella abortus</td>
<td>Brucella abortus infection, Brucella abortus bacterium</td>
</tr>
<tr>
<td>BSA</td>
<td>Body Surface Area, Bovine Serum Albumin</td>
</tr>
<tr>
<td>BSE</td>
<td>Bovine Spongiform-Encephalopathy, Breast Self-Examination</td>
</tr>
<tr>
<td>Ca</td>
<td>Hippocampus (Brain), Calcium, California, Canada</td>
</tr>
</tbody>
</table>

Table 5: The first 20 ambiguous terms/tasks in Med WSD data set.
Handling Noisy Labels for Robustly Learning from Self-Training Data for Low-Resource Sequence Labeling

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Abstract

In this paper, we address the problem of effectively self-training neural networks in a low-resource setting. Self-training is frequently used to automatically increase the amount of training data. However, in a low-resource scenario, it is less effective due to unreliable annotations created using self-labeling of unlabeled data. We propose to combine self-training with noise handling on the self-labeled data. Directly estimating noise on the combined clean training set and self-labeled data can lead to corruption of the clean data and hence, performs worse. Thus, we propose the Clean and Noisy Label Neural Network which trains on clean and noisy self-labeled data simultaneously by explicitly modelling clean and noisy labels separately. In our experiments on Chunking and NER, this approach performs more robustly than the baselines. Complementary to this explicit approach, noise can also be handled implicitly with the help of an auxiliary learning task. To such a complementary approach, our method is more beneficial than other baseline methods and together provides the best performance overall.

1 Introduction

For many low-resource languages or domains, only small amounts of labeled data exist. Raw or unlabeled data, on the other hand, is usually available even in these scenarios. Automatic annotation or distant supervision techniques are an option to obtain labels for this raw data, but they often require additional external resources like human-generated lexica which might not be available in a low-resource context. Self-training is a popular technique to automatically label additional text. There, a classifier is trained on a small amount of labeled data and then used to obtain labels for unlabeled instances. However, this can lead to unreliable or noisy labels on the additional data which impede the learning process (Pechenizkiy et al., 2006; Nettleton et al., 2010). In this paper, we focus on overcoming this slowdown of self-training. Hence, we propose to apply noise-reduction techniques during self-training to clean the self-labeled data and learn effectively in a low-resource scenario.

Inspired by the improvements shown by the Noisy Label Neural Network (NLNN, Bekker and Goldberger (2016)), we can directly apply NLNN to the combined set of the existing clean data and the noisy self-labeled data. However, such an application can be detrimental to the learning process (Section 6). Thus, we introduce the Clean and Noisy Label Neural Network (CNLNN) that treats the clean and noisy data separately while training on them simultaneously (Section 3).

This approach leads to two advantages over NLNN (Section 6 and 7) when evaluating on two sequence-labeling tasks, Chunking and Named Entity Recognition. Firstly, when adding noisy data, CNLNN is robust showing consistent improvements over the regular neural network, whereas NLNN can lead to degradation in performance. Secondly, when combining with an indirect-noise handling technique, i.e. with an auxiliary target in a multi-task fashion, CNLNN complements better than NLNN in the multi-task setup and overall leads to the best performance.

2 Related Work

Self-training has been applied to various NLP tasks, e.g. Steedman et al. (2003) and Sagae and Tsujii (2007). While McClosky et al. (2006) are able to leverage self-training for parsing, Charniak (1997) and Clark et al. (2003) obtain only minimal improvements at best on parsing and POS-tagging.
respectively. In some cases, the results even deteriorate. Other successful approaches of automatically labeling data include using a different classifier trained on out-of-domain data (Petrov et al., 2010) or leveraging external knowledge (Dembowski et al., 2017).

A detailed review of learning in the presence of noisy labels is given in (Fréñay and Verleysen, 2014). Recently, several approaches have been proposed for modeling the noise using a confusion matrix in a neural network context. Many works assume that all the data is noisy-labeled (Bekker and Goldberger, 2016; Goldberger and Ben-Reuven, 2017; Sukhbaatar et al., 2015). Hedderich and Klakow (2018) and Hendrycks et al. (2018) propose a setting where a mix of clean and unlabeled data is used. However, they require external knowledge sources for labeling the data or evaluate on synthetic noise. Alternatively, instances with incorrect labels might be filtered out, e.g. in the work by Guan et al. (2011) or Han et al. (2018), but this involves the risk of also filtering out difficult but correct instances. Another orthogonal approach is the use of noise-robust loss functions (Zhang and Sabuncu, 2018).

3 Clean and Noisy Label Neural Network

The Noisy Label Neural Network (NLNN, Bekker and Goldberger (2016)) assumes that all observed labels in the training set pass through a noise channel flipping some of them from a correct to an incorrect label (see left part of Figure 1). In our scenario, this means that both the human-annotated and the additional automatically-labeled (self-training) corpora are assumed to be noisy. In our experiments (Section 6 and 7), treating both corpora in this fashion degrades the overall performance. To remedy this effect, we propose to treat the human-annotated data as clean data and the self-training data as noisy.

We assume a similar setup as Bekker and Goldberger (2016), training a multi-class neural network soft-max classifier

\[ p(y = i|x; w) = \frac{\exp(w_i^T h)}{\sum_{j=1}^{k} \exp(w_j^T h)} \]

where \( x \) is the feature vector, \( y \) is the label, \( w \) denotes the network weights, \( k \) is the number of possible labels, \( u \) are soft-max weights and \( h = h(x) \) denotes the multi-layer neural network applied to \( x \). In contrast to Bekker and Goldberger (2016), we assume that not all of the training data passes through a noisy channel changing the correct labels \( y \) to noisy ones \( z \). A part of the training set remains clean \( z \in C \) such that \( |C| + |N| = n \) where \( n \) is the total number of training examples. The clean labels are a copy of the corresponding correct labels. A schematic representation of this model is shown on the right side of Figure 1. The correct labels \( y \) and the noise distribution \( \theta \) are hidden for the noisy labels.

We define the probability of observing a label \( z \), which can either be noisy or clean and is, thus, dependent on the label’s membership to \( C \) or \( N \):

\[ p(z = j|x, w, \theta) = \begin{cases} \sum_{i=1}^{k} p(z = j|y = i; \theta)p(y = i|x; w) & \text{if } z \in N \\ p(y = j|x; w) & \text{if } z \in C \end{cases} \]

Using this probability function and \( t \) to index training instances, the log-likelihood of the model parameters is defined as

\[ L(w, \theta) = \sum_{t \in C} \log p(z_t|x_t, w) + \sum_{t \in N} \log (\sum_{i=1}^{k} p(z_t|y_t = i; \theta) \cdot p(y_t = i|x_t; w)) \]

As in Bekker and Goldberger (2016) the model parameters are computed using Expectation Maximization. In the E-step, \( \theta \) and \( w \) are fixed and an estimate \( c \) of the true labels \( y \) is obtained for the noisy labels \( z \):

\[ c_t = p(y_t = i|x_t, z_t; w, \theta) = \frac{p(z_t|y_t = i; \theta)p(y_t = i|x_t; w)}{\sum_{j} p(z_t|y_t = j; \theta)p(y_t = j|x_t, w)} \quad \text{for } z_t \in N \]

Note that the estimate \( c \) is calculated only for the noisy labels whereas the clean labels remain unchanged. Similarly, the noise distribution \( \theta \) is calculated only for the noisy labels. The initialization of \( \theta \) and the \( \theta \)'s update step in M-step remain the same as in Bekker and Goldberger (2016), also
shown below.

\[
\theta(i, j) = \frac{\sum_{l} c_{li} 1_{l(i, j)}}{\sum_{l} c_{li}}, \quad i, j \in \{1, ..., k\}, z_{i} \in N
\]

During the M-step, the neural network weights \( w \) are estimated as well. The loss function, however, changes compared to the original approach (Bekker and Goldberger, 2016) to (1) and thus, changing the calculation of the gradient to (2):

\[
S(w) = \sum_{z_{i} \in C} \log p(z_{i}|x_{i}, w) + \sum_{z_{i} \in N} \sum_{k=1}^{k} c_{ki} \log p(y_{i}|x_{i}; w)
\]

\[
\frac{\partial S}{\partial u_{t}} = \sum_{z_{i} \in C} (1_{z_{i} \cdot t} - p(z_{i}|x_{i}, w))h(x_{i}) + \sum_{z_{i} \in N} (c_{ti} - p(y_{i}|x_{i}, w))h(x_{i})
\]

Interestingly, the gradient calculation (2) is a summation of two parts: one to learn from the clean labels and another to learn from the noisy labels. We refer to this model as the Clean and Noisy Label Neural Network (CNLNN).

4 Training with Noisy Labels in a Multi-Task Setup

NLNN and CNLNN form explicit ways of handling noise as the noise distribution is calculated during training. In contrast, we apply a Deep Multi-Task Learning (MTL) approach (Søgaard and Goldberg, 2016), which, unlike NLNN and CNLNN, does not estimate the noise directly and thus, is an implicit noise-cleaning approach. The MTL method leverages an auxiliary task that augments the data providing other reliable labels and hence, ignoring noisy labels (Ruder, 2017). In our experiments, we combine the implicit noise handling of Deep MTL with the explicit noise handling of NLNN and CNLNN to complement each other and obtain a more powerful noise handling model than the individual models. Schematic depiction of combining MTL and CNLNN is shown in Figure 1. MTL and NLNN can also be combined in a similar way.

5 Experimental Setup

We evaluate CNLNN and other methods on a Chunking and a Named Entity Recognition (NER) task with \( F_1 \)-score as the metric in each case. For Chunking, we use the same data splits as (Søgaard and Goldberg, 2016) based on the English Penn Treebank dataset (Marcus et al., 1993). For NER, the data splits of the English CoNLL 2003 task are used (Sang and Buchholz, 2000). Note that in our NER setup, we evaluate using BIO-2 labels, so \( F_1 \)-scores reported below might not be comparable to prior work.

To mimic a low resource setting, we limit each training set to the first 10k tokens. The development sets are randomly chosen sentences from the original training set restricted to 1k tokens. The test sets remain unchanged. For the rest of the training data, the original labels are removed and the words are automatically labeled using the baseline model (NN described below). We add variable amounts of this automatically-annotated data for self-training in our experiments.

5.1 Models

We apply the following models to the above two tasks: NN (the simple baseline) is an architecture with bidirectional LSTMs (Hochreiter and Schmidhuber, 1997). For Chunking, we use three LSTM layers, for NER five. The NN model, only
trained on the clean data, is used for automatically labeling the raw data (obtaining the noisy data). NLNN combines the NN with the original noise channel (Bekker and Goldberger, 2016), training it both on clean and noisy instances. CNLNN is our new approach of modeling noise, treating clean and noisy labels separately (section 3).

In contrast to the explicit noise handling of NLNN and CNLNN, we also apply MTL for implicit noise handling. Here, we use NN as the base architecture and POS-tagging as an auxiliary task. We hypothesise that this low-level task helps the model to generalise its representation and that the POS-tags are helpful because e.g. many named entities are proper nouns. The auxiliary task is trained jointly with the first LSTM layer of NN for Chunking and with the second LSTM layer for NER. In our low-resource setting, we use the first 10k tokens of section 0 of Penn Treebank for the auxiliary POS-tagging task for the MTL (Søgaard and Goldberg, 2016). This data is disjunct from the other datasets.

Additionally, we combine both the explicit and implicit noise handling. In the low-resource setting, in general, such a combination addresses the data scarcity better than the individual models. NLNN and CNLNN combinations with MTL are labeled as MTL+NLNN and MTL+CNLNN respectively.

5.2 Implementation Details

During training, we minimize the cross entropy loss which sums over the entire sentence. The networks are trained with Stochastic Gradient Descent (SGD). To determine the number of iterations for both the NN model and the EM algorithm we use the development data. All models are trained with word embeddings of dimensionality 64 that are initialized with pre-trained Polygot embeddings (Al-Rfou et al., 2013). We add Dropout (Srivastava et al., 2014) with p=0.1 in between the word embedding layer and the LSTM.

6 Results

In Figure 2, we present the $F_1$ scores of the models introduced in the previous section. We perform experiments on Chunking and NER with various amounts of added, automatically-labeled data. In general, adding additional, noisy data tends to improve the performance for all models. This includes the plain NN, showing that this model is somewhat robust to noise. Especially for the Chunking task, the possibility for improvement seems limited for NN as the performance converges after adding 40k noisy instances. In the Chunking 10k case, the negative effect of the noisy instances results in a score lower than if no data is added.

The original NLNN model performs similarly to the NN model without a noise-handling component. In some cases, the score is even lower. In contrast, CNLNN is able to consistently improve over these scores. This demonstrates the importance of our proposed CNLNN which treats clean and noisy data separately.

MTL is able to improve somewhat over NN even without adding automatically-annotated data thanks to the auxiliary task. Additionally, MTL performs even better when noisy data is added showing its implicit noise handling capabilities. On their own, both CNLNN and MTL are able to eliminate some of the negative effects of the noisy data and to leverage the additional data effectively.

Combining MTL with NLNN results in small improvements at best and can decrease perfor-
performance, especially on Chunking. The best results are achieved with our combined MTL+CNLNN model as it outperforms all other models. Even when adding 19 times the amount of self-labeled data, the model is still able to cope with the noise and improve the performance.

7 Analysis

**NLNN vs. CNLNN:** In NLNN, we observed that clean training tokens were subverted to become noisy in subsequent EM iterations mostly due to the influence of noisy labels from self-labeled data and this effect leads to NLNN’s worse performance. Figure 3 presents one such case where the corruption of the confusion matrix from 1. iteration to 3. iteration is displayed. CNLNN treats clean and noise data separately and therefore avoids the corruption of clean labels.

**MTL+CNLNN vs. MTL+NLNN:** We noted that MTL+CNLNN consistently outperforms MTL and MTL+NLNN, whereas the MTL+NLNN combination can degrade MTL’s performance. For nearly all predicted labels the improvements in precision over MTL are higher for MTL+CNLNN when compared to MTL+NLNN (Figure 4). This shows that CNLNN complements MTL better than NLNN.

8 Concluding Remarks

In this paper, we apply self-training to neural networks for Chunking and NER in a low-resource setup. Adding automatically-labeled data, the performance of the classifier can wane or can even decline. We propose to mitigate this effect by applying noisy label handling techniques.

However, we found that directly applying an off-the-shelf noise-handling technique as NLNN leads to corruption of the clean training set and worse performance. Thus, we propose the Clean and Noisy Label Neural Network to work separately on the automatically-labeled data. Our model improves the performance faster for a lesser amount of additional data. Moreover, combing the training with auxiliary information can further help handle noise in a complementary fashion.

Meanwhile, more complex neural network architectures (Goldberger and Ben-Reuven, 2017; Luo et al., 2017; Veit et al., 2017) are available for handling noise and we look forward to working with these to upgrade our approach in the future.

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References


Opinion Mining with Deep Contextualized Embeddings

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Abstract
Detecting opinion expression is a potential and essential task in opinion mining that can be extended to advanced tasks. In this paper, we considered opinion expression detection as a sequence labeling task and exploited different deep contextualized embedders into the state-of-the-art architecture, composed of bidirectional long short-term memory (BiLSTM) and conditional random field (CRF). Our experimental results show that using different word embeddings can cause contrasting results, and the model can achieve remarkable scores with deep contextualized embeddings. Especially, using BERT embedder can significantly exceed using ELMo embedder.

1 Introduction
One of the crucial tasks in sentiment analysis field is opinion mining, which right now becomes more and more popular for survey. The purpose of opinion mining is to detect the emotional expression from a sentence or an entire document. To be more specific, those expressions usually contain human beings’ emotions, interests, even attitudes via natural language.

Fine-grained opinion mining is not only fundamental but also important because bountiful Natural Language Processing (NLP) applications can benefit from it. For example, detecting opinion expression can be extended to identify opinion entity, such as Katiyar and Cardie (2016); Miwa and Bansal (2016); Katiyar and Cardie (2017), recognize stance (Somasundaran and Wiebe, 2010), and extract aspect (Xu et al., 2018; Wang et al., 2016).

Opinion expression detection can be viewed as a linguistic sequence labeling problem. Therefore, recognizing opinionated span from a sentence can be designed as a token-level sequence tagging problem. In this case, standard BIO encoding is usually applied in the same way in Irsoy and Cardie (2014); Choi et al. (2005). Thus, we used the dataset tagged with B, I, and O characters, which represent the beginning, inside, and outside respectively. The first token in each opinionated span is attached to B character, and then the rest of the span are assigned to I character. Table 1 is an example of BIO scheme.

Out of exactness, in this study, we chose the dataset MPQA 1.2 used in Xie (2017); Irsoy and Cardie (2014). To estimate the generalization of our model, we also took another opinion-oriented dataset MOAT from the organization NTCIR7 (Seki et al., 2008).

Opinion expression usually contains a speaker’s emotion; hence, we assume that semantics contributes more than syntax. Even though pre-trained word embedding can improve the performance, it still doesn’t fully utilize word meaning and its context. Therefore, generating word representations based on the contexts is critical. Owing to deep neural network, model can produce dynamic word representation, such as McCann et al. (2017); Peters et al. (2018); Akbik et al. (2018), on the contrary to fixed representation (Pennington et al., 2014; Mikolov et al., 2013). In this paper, we applied two state-of-the-art and innovative models, ELMo (Peters et al., 2018) and BERT (Devlin et al., 2018), to produce word representations and compared the performances. After obtaining the contextualized word representations,
we fed them into a robust neural network.

2 Related Work

Word Representation. In the past few years, distributed word representations, also known as word embeddings, have been broadly applied to NLP tasks because adding pre-trained one can improve the performance by 1 to 2 point.

Many researches are done with GloVe (Pennington et al., 2014) or Word2Vec (Mikolov et al., 2013). However, one vector can not represent all the meanings of a word. For the sake of deep neural networks, we can add contextual characteristic into word representation during converting. For instance, CoVe (McCann et al., 2017) learns contextualized word vectors via encoders in translation. ELMo (Peters et al., 2018) produces contextualized embeddings from language models computed on 2-layer BiLSTMs with character convolutions. Akbik et al. (2018) trained a character language model to produce contextual string embeddings.

Different from the models mentioned above, BERT (Devlin et al., 2018) extends Radford (2018), extracting features by Transformer (Vaswani et al., 2017), from uni-direction into bi-direction and outperforms impressively in many tasks. In this paper, we took BERT as our embedder because it can generate contextual features. In contrast, ELMo is selected as our baseline embedder.

According to Ruder and Howard (2018), pre-training on a large amount of unlabeled data do improve the model; thus, we used pre-trained BERT and ELMo model and then did fine-tuning with our data.

Neural Network. Recurrent Neural Network (RNN) (Jain and Medsker, 1999) handles variable length input and performs well on NLP tasks, in particular Long Short-Term Memory (LSTM) (Hochreiter and Schmidhuber, 1997). At present, state-of-the-art approaches for sequence labeling typically use bidirectional LSTMs (BiLSTMs) (Schuster and Paliwal, 1997), and a conditional random field (CRF) decoding layer. BiLSTMs can capture not only the previous information but also the following information.

Recently, RNNs have been generally utilized in sequential modeling, such as Irsoy and Cardie (2014), which stacked 3-layer bidirectional vanilla RNNs and the model outperformed the variants of CRF-based methods. Moreover, LSTM has multiple gates allowing the model to learn long-distance dependencies. For sequential labeling tasks, such as Name Entity Recognition (NER) and Part-of-speech (POS) tagging, BiLSTM can take both former and latter contexts into consideration without length limitation and perform better (e.g., Liu et al. (2015) resorted to LSTM with some linguistic features and Katiyar and Cardie (2016) applied deep BiLSTMs on detecting opinion entities and relations).

So far, there have been many variants of LSTM-based neural network models proposed to improve the model and achieve competitive performance against traditional models. Miwa and Bansal (2016) made use of dependency tree and fed it into tree-structured BiLSTM. Simpler and more ingenious, Katiyar and Cardie (2017) only employed attention-based BiLSTM for entity and relation without any manual features or dependency structure information.

Conditional random fields (CRFs) (Lafferty et al., 2001) have also been quite successful for sequence tasks. Thanks to CRF layer, the model can take advantage of sentence-level and neighbor tag information to predict current tag (Huang et al., 2015; Ma and Hovy, 2016).

Some researches combined CNN with RNN to gain the benefits from each other. They extracted character-level features via CNN and handled sentences via RNN, such as Chiu and Nichols (2016). BiLSTM-CNNs-CRF (Ma and Hovy, 2016) obtained features via CNN and stacks CRF on BiLSTM. Xie (2017) applied CNN with bi-direction Gated Recurrent Units (GRUs). Xu et al. (2018) applied CNN to extract features with dual embeddings.

Our study compared BiLSTM-CNNs-CRF (2016) with the model using ELMo as embedder (ELMo-BiLSTM-CRF) and then competed ELMo-BiLSTM-CRF with the one using BERT as embedder (BERT-BiLSTM-CRF). Afterwards, we compared the performance and the difference between two embedders.

Eventually, based on Katiyar and Cardie (2017); Vaswani et al. (2017), we employed attention mechanism in BERT-BiLSTM-CRF layer expected to improve the accuracy.
3 Model

The main focus of our research is to improve the performance by using the deep contextualized word representation and compare two kinds of word embedders. Therefore, rather than static word representations, we employed dynamic word embedders that create the word representation based on its context.

After transforming all tokens into continuous vector representations, we fed them into BiLSTM instead of feed-forward network because BiLSTM neural network is prevailing and dominant on NLP tasks and many opinion-related tasks are completed with it.

After finishing each epoch, we updated the parameters simultaneously via back-propagation through time (BPTT) (Werbos, 1990). Last, after our comparisons, we chose the better model and added attention mechanism on BiLSTM layer to strengthen the performance.

In the following subsections, we decompose our neural network architectures and describe the components (layers) in detail. Hence, we introduce the neural layers in our models one-by-one from bottom to top. Before describing the models, we first illustrate the annotation format of data, which is followed by the most important part, word embedders. Afterwards, the BiLSTM layers as well as CRF layer will be mentioned.

3.1 Data Scheme

Opinion expression detection can be viewed as a linguistic sequence labeling problem. In this case, BIO encoding is usually applied to identify the opinionated span in a sentence. Thus, in the dataset, each token is tagged with BIO.

3.2 Word Embedders

According to previous research, using pre-trained word embeddings in downstream tasks usually outperforms using randomly initialized vectors. Therefore, choosing a robust embedding way to transform tokens is influential in experiments. Nevertheless, although there have been abundant ways to convert words into dense distributions so far, we first did some experiments to discover how effective each approach is.

After observing the datasets, we found that opinion detection task concentrates more on semantics than syntax. Consequently, we used a deep neural network, BERT, to figure out this problem because it is flexible enough to produce the representation of each token based on its context, even more the whole sentence.

Moreover, the BERT model is so overwhelming that it works impressively on plenty of tasks. In order to examine the performance on different embedding in opinion mining, we used BERT model as the word embedder in our experiments. Besides, it is necessary to compare the main model with a baseline. Therefore, we took another contextualized embedding, ELMo, as word embedding, and regarded the results as our baseline.

3.3 Architecture

We stacked BiLSTM on top of embedders because of the following reasons. First of all, in our research, BERT and ELMo are only used as word embedders instead of the whole architecture. Second, many RNN-based neural network models are proposed to figure out the sequence labeling tasks, and also achieved competitive performance against traditional models (Ma and Hovy, 2016; Huang et al., 2015). Last but not least, we would like to compare the performances between different contextualized embeddings so we must fix the other parts of architecture.

In addition, we also added CRF (Lafferty et al., 2001) layer because it can consider the correlations between labels in neighborhoods to predict current tag. Thanks to CRF layer, the model took full advantage of sentence-level tag information and enhanced itself to decode the best chain of labels for a given input sentence. In summary, we combined BERT or ELMo embedders with BiLSTM and CRF layers (Bert-BiLSTM-CRF and ELMo-BiLSTM-CRF). Each model can efficiently benefit from the forward and backward input features through BiLSTM layer and sentence-level tag information via CRF layer.

3.4 Attention

Katiyar and Cardie (2017) displayed that attention mechanism can reinforce the model. Therefore, we extended the Bert-BiLSTM-CRF network with multi-head attention approach because it allows the model to jointly attend to information from different representation sub-spaces at different positions (Vaswani et al., 2017). More concretely, the hidden states from BiLSTM layers went through attention layer and then linear layer as well as CRF layer.
Table 2: The number of sentences for each dataset.

<table>
<thead>
<tr>
<th>Dataset</th>
<th>Count</th>
</tr>
</thead>
<tbody>
<tr>
<td>DSE</td>
<td>14492</td>
</tr>
<tr>
<td>ESE</td>
<td>14492</td>
</tr>
<tr>
<td>NTCIR7-MOAT</td>
<td>3376</td>
</tr>
</tbody>
</table>

4 Experiments

4.1 Data Sets

Due to the development of opinion mining, there are numerous existing datasets which are useful for us. For example, Multi-Perspective Question Answering (MPQA) offers the annotated dataset.

In this paper, we conducted the experiments on the processed dataset, MPQA 1.2, provided by İrsoy and Cardie (2014). It includes two types of opinion expression proposed by Wiebe et al. (2005) - direct subjective expressions (DSEs) and expressive subjective expressions (ESEs). DSEs represent both subjective speech events and explicitly mentioned private states, while ESEs consist of tokens that express emotion or sentiment in an indirect or implicit way. Table 1 is also an example of DSE and ESE.

In addition to MPQA 1.2, we have another dataset from MOAT task, which is also related to opinion expression and organized by NTCIR (Seki et al., 2008). Different datasets can confirm that our model is generalized enough. Table 2 depicts the number of sentences in each dataset.

In the research, we first used one tenth of the dataset as the development set, and the rest of the data are applied 10-fold in order to get a more balanced result.

4.2 Word Embeddings

This experiment contains the details of the comparison between ELMo and other embedding ways. Table 3 is the results from different notable embedding ways on DSEs, and the evaluation is token-based calculation.

We selected some pre-trained word embedding - Word2Vec (Mikolov et al., 2013), GloVe (Pennington et al., 2014), and dependency-based word embedding (Levy and Goldberg, 2014). Besides, we also tried BiLSTM-CNNs-CRF (Ma and Hovy, 2016), which extracted character features by CNN and concatenated GloVe.

All the architectures are identical except for the word representations and decoding ways. Each combination is composed of one type of embeddings and three BiLSTM layers with or without CRF layer.

According to the results, the model with CRF layer does defeat the model without CRF layer. It proves that adding CRF layer becomes more powerful. Moreover, all the scores are close except for ELMo one. BiLSTM-CNNs-CRF doesn’t exceed the other models only with pre-trained word embedding. We assumed that opinion mining emphasizes more semantics than syntax. Therefore, character features do not boost the model much.

Subsequently, we compared BiLSTM-CNNs-CRF with the one using ELMo embedder (ELMo-BiLSTM-CRF), and it turned out that ELMo-BiLSTM-CRF surpasses BiLSTM-CNNs-CRF substantially. Therefore, we took it as our baseline in this paper.

4.3 Parameter Settings

We made use of pre-trained BERT and ELMo models and did the fine-tuning during training procedure guided by Devlin et al. (2018) in order to make the model learn the distribution of the dataset.

For ELMo model, we used ELMoForMany-Langs\(^1\) provided by HIT-SCIR (Che et al., 2018; Fares et al., 2017) and took the average of the hidden states from all the layers as token representation.

For BERT model, we used “BERT-Base, Multilingual Cased” provided by Google\(^2\) (Devlin et al., 2018). Most of the hyper-parameters are instructed by the paper. Only the hidden state from the last layer is picked and regarded as token representation.

For the rest of shared properties, we stacked 3 layers of BiLSTM\(^3\) based on İrsoy and Cardie (2014). Learning rate is 0.00005 advised by (Devlin et al., 2018). Dropout is set to 0.5 before fully connected layer. Optimizer uses Adam (Kingma and Ba, 2014).

4.4 Evaluation

In the evaluation, we calculated precision, recall, and F1-score. However, in order to evaluate our model comprehensively, we measured our model not only by token basis but also span basis. This

\(^1\)https://github.com/HIT-SCIR/ELMoForManyLangs

\(^2\)https://github.com/google-research/bert

\(^3\)According to our experiments, the number of layers do not affect the result much.
is because it is difficult to define the boundaries of expressions, even for human annotators. To put it another way, for token-based evaluation, F1-score pays attention to whether the individual tag is predicted correctly or not. In contrast, span-based evaluation cares about the count of overlap; hence, we used binary overlap along with proportional overlap. Binary overlap computes the number of matching overlaps between predicted sequence and ground-truth sequence. As long as predicted span overlaps ground-truth span, binary overlap views it as a correct prediction. To refine the evaluation, proportional overlap considers the length of overlap and imparts a partial correctness to each match, which is able to assess the model more accurate. We used these three evaluations to measure our models.

5 Results and Discussion

5.1 Results

Table 4, 5, and 6 display the results of the data sets individually. The scores show that using BERT embedder does achieve better performance than using ELMo embedder by dramatic difference, which implies that BERT is promising in opinion mining tasks. More interestingly, adding attention mechanism on BiLSTM can increase the performance slightly.

5.2 Discussion

According to the scores, it is easier to detect DSEs than ESEs because DSEs contain clear opinion-related expressions, whereas ESEs may or may not contain explicit opinion words. To put it another way, for token-based evaluation, F1-score pays attention to whether the individual tag is predicted correctly or not. In contrast, span-based evaluation cares about the count of overlap; hence, we used binary overlap along with proportional overlap. Binary overlap computes the number of matching overlaps between predicted sequence and ground-truth sequence. As long as predicted span overlaps ground-truth span, binary overlap views it as a correct prediction. To refine the evaluation, proportional overlap considers the length of overlap and imparts a partial correctness to each match, which is able to assess the model more accurate. We used these three evaluations to measure our models.
ated expression, which may be some adjectives. However, detecting ESEs requires understanding more implicit semantics, but BERT-BiLSTM-CRF works well on ESEs.

Adding attention mechanism does not improve much probably because BERT layer has already incorporated multi-head attention mechanism and caught well-represented information.

Moreover, the training epochs for ELMo-BiLSTM-CRF is 4 times more than that of BERT-BiLSTM-CRF to converge. In other words, applying BERT embedder can save much more time. To conclude, in our experiments, BERT embedder is much more efficient than other embeddings.

### 5.3 Error Analysis

In this section, we observed the predictions and analyzed the defect in our models. Table 7 is our several predictions from our model on each dataset. Many sentences are predicted approximately or even same; however, in some sentences ELMo-BiLSTM-CRF has lower recall.

BERT-BiLSTM-CRF can predict well, but some failures are caused by the inconsistency in dataset. For example, whether definite articles (e.g. ‘the’) or punctuation should be included or not is one of the problems. Besides, the same verb, such as ‘say’, in similar contexts is not always annotated, either.

For ESEs, it is much more difficult to clearly identify implicit semantics because there are many fragmented predictions. Besides, although we have CRF layer to consider the entire sequence predictions, it still exists some wrong tagging, such as starting with I tag.

The opinionated spans in NTCIR7-MOAT data are usually too long, which is a little different from the other datasets. Besides, the number of sentences is not much; thus, the result does not meet the expectation.

Once the flaws in dataset are figured out, we can gain a better performance. Furthermore, adding another feature, such as GLoVe (Pennington et al., 2014) or linguistic characteristics, is also a way to enhance the model.

### 6 Conclusion

In this paper, we introduced contextualized embeddings into opinion mining task. Experimentally, our models have significant promotion for changing embedder and prove that deep contextualized embeddings perform well in opinion mining task. Specifically, our comparison shows that using BERT embedder dramatically surpasses using ELMo embedder.

In the future work, it would be better to supplement other word embedding (Pennington et al., 2014; Mikolov et al., 2013) as auxiliary, just like

<table>
<thead>
<tr>
<th>DSE S1</th>
<th>My public affairs keepers [could]$_B$ [n’t care less]$_I$.</th>
</tr>
</thead>
<tbody>
<tr>
<td>ELMo-BiLSTM-CRF</td>
<td>My public affairs keepers could n’t care less.</td>
</tr>
<tr>
<td>Bert-BiLSTM-CRF</td>
<td>My public affairs keepers [could]$_B$ [n’t care]$_I$ less.</td>
</tr>
</tbody>
</table>

<table>
<thead>
<tr>
<th>ESE S1</th>
<th>By comparison , the al Qaedans [look]$_B$ [pretty fat , if not happy]$_I$.</th>
</tr>
</thead>
</table>

<table>
<thead>
<tr>
<th>ESE S2</th>
<th>Their restroom arrangements [are]$_B$ [pretty spartan]$_I$.</th>
</tr>
</thead>
<tbody>
<tr>
<td>ELMo-BiLSTM-CRF</td>
<td>Their restroom arrangements are [pretty]$_B$ [spartan]$_I$.</td>
</tr>
<tr>
<td>Bert-BiLSTM-CRF</td>
<td>Their restroom arrangements are [pretty]$_B$ [spartan]$_I$.</td>
</tr>
<tr>
<td>Bert-BiLSTM-Attn-CRF</td>
<td>Their restroom arrangements are [pretty]$_B$ [spartan]$_I$.</td>
</tr>
</tbody>
</table>

<table>
<thead>
<tr>
<th>ESE S3</th>
<th>We can see for ourselves , [sort]$_B$ [of]$_I$.</th>
</tr>
</thead>
<tbody>
<tr>
<td>ELMo-BiLSTM-CRF</td>
<td>We can see for ourselves , [sort]$_I$ of .</td>
</tr>
<tr>
<td>Bert-BiLSTM-Attn-CRF</td>
<td>We can see for ourselves [.]$_B$ [sort]$_I$ [of]$_I$.</td>
</tr>
</tbody>
</table>

<table>
<thead>
<tr>
<th>MOAT S1</th>
<th>[He]$_B$ [considers them as Indonesian as he is .]$_I$.</th>
</tr>
</thead>
<tbody>
<tr>
<td>ELMo-BiLSTM-CRF</td>
<td>He considers [them]$_I$ as [Indonesian]$_I$ as [he]$_I$ is .</td>
</tr>
<tr>
<td>Bert-BiLSTM-CRF</td>
<td>He considers them as [Indonesian]$_I$ as he is .</td>
</tr>
<tr>
<td>Bert-BiLSTM-Attn-CRF</td>
<td>[He]$_B$ [considers them as Indonesian as he is .]$_I$.</td>
</tr>
</tbody>
</table>

Table 7: Output from our models for each dataset.
Chiu and Nichols (2016); Xu et al. (2018). Even more, we can add contextual string embedding (Akbik et al., 2018) to support character-level features and apply it to advanced opinion mining tasks.

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A bag-of-concepts model improves relation extraction in a narrow knowledge domain with limited data

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Abstract

This paper focuses on a traditional relation extraction task in the context of limited annotated data and a narrow knowledge domain. We explore this task with a clinical corpus consisting of 200 breast cancer follow-up treatment letters in which 16 distinct types of relations are annotated. We experiment with an approach to extracting typed relations called window-bounded co-occurrence (WBC), which uses an adjustable context window around entity mentions of a relevant type, and compare its performance with a more typical intra-sentential co-occurrence baseline. We further introduce a new bag-of-concepts (BoC) approach to feature engineering based on the state-of-the-art word embeddings and word synonyms. We demonstrate the competitiveness of BoC by comparing with methods of higher complexity, and explore its effectiveness on this small dataset.

1 Introduction

Applying automatic relation extraction on small data sets in a narrow knowledge domain is challenging. Here, we consider the specific context of a small clinical corpus, in which we have a variety of relation types of interest but limited examples of each. Transformation of clinical texts into structured sets of relations can facilitate the exploration of clinical research questions such as the potential risks of treatments for patients with certain characteristics, but large-scale annotation of these data sets is notoriously difficult due to the sensitivity of the data and the need for specialized clinical knowledge.

Rule-based methods (Abacha and Zweigenbaum, 2011; Verspoor et al., 2016) typically determine whether a particular type of relation exists in a given text by leveraging the context in which key clinical entities are mentioned. For instance, if two entities with type TestName and TestResult, respectively, are observed in a given sentence, it is likely that a relation of type TestFinding exists between them. However, construction of high-precision rules defining relevant contexts is time-consuming and expensive, requiring extensive effort from domain experts.

The state-of-the-art machine learning algorithms such as neural network models (Nguyen and Grishman, 2016; Ammar et al., 2017; Huang and Wang, 2017) may over-fit in performing relation extraction in this context, due to a limited quantity of training instances.

In this work, we experiment with two automatic approaches to semantic relation extraction applied to a small corpus consisting of breast cancer follow-up treatment letters (Pitson et al., 2017), comparing a simple rule-based co-occurrence approach to machine learning classifiers.

The first approach, simple co-occurrence (Verspoor et al., 2016), is based on the assumption that most relevant relations are intra-sentential, that is, the relation between a pair of named entities is expressed within the scope of a single sentence. However, some relations may be expressed across sentence boundaries, and thus a single sentence may not be the ideal choice of scope, as shown in prior work that considers inter-sentential relations (also known as non-sentence or cross-sentence relations) (Panyam et al., 2016; Peng et al., 2017). We extend the co-occurrence approach to allow explicit adjustment of context window size, from one to two sentences, a method called Window-Bounded Co-occurrence (WBC). The best window size for a given relation is identified by choosing the one which produces the highest score under $F_1$-measure on a development set.

The second approach is based on supervised binary classification. We transform the multi-relation extraction task into several independent
binary tasks. We build on a bag-of-concepts (BoC) (Sahlgren and Cöster, 2004) approach which models the text in terms of phrases or pre-identified concepts, extending it with word embeddings and word synonyms. We compare two different pre-trained word embedding models, and a number of other model variations. We also explore grouping of synonyms into abstracted concepts.

We find that the intra-sentential rule-based approach outperforms the approach which allows for a larger context window. The supervised learning models outperform rule-based approaches under F1 measure, and their results show that models using BoC features outperform models with BoW, dependency parse, or sentence embedding features. We also show that SVM outperforms complex models such as a feed-forward ANN in our low resource scenario, with less tendency to over-fitting.

2 Background

At present, the two primary approaches to automatic relation extraction over biomedical corpora are rule-based approaches (Verspoor et al., 2016; Abacha and Zweigenbaum, 2011) and machine learning approaches based on learners such as logistic regression, support vector machines (SVM) (Panyam et al., 2016) and convolutional neural networks (CNN) (Nguyen and Verspoor, 2018) together with sophisticated feature engineering methods.

Verspoor et al. (2016) established a typical intra-sentential co-occurrence baseline with competitive performance comparing to a comprehensive machine learning-based system, PKDE4J (Song et al., 2015), on the extraction of relations between human genetic variants and disease on the Variome corpus (Verspoor et al., 2013). The sentential baseline is based on the assumption that the scope of relations is within one sentence, and further assumes that any pair of two entities mentioned in the same sentence and satisfying the type constraints of a given relation, expresses that relation. For example, if two entities with type TimeDescriptor and EndocrineTherapy respectively, the relation TherapyTiming will be extracted. The sentential co-occurrence baseline set a benchmark for relation extraction on Variome corpus.

Abacha and Zweigenbaum (2011) explored semantic rules for the extraction of relations between medical entities on PubMed Central (PMC) articles using linguistic patterns. They provide an example of implementing an end-to-end relation extraction system, applying named entity recognition in the first stage, then followed by the stage of relation extraction. They define several relation patterns based on medical knowledge, and leveraging the dependency parse tree of sentences in which entities occur. However, the linguistic patterns and rules developed for their corpus likely are not directly applicable to our semantically distinct context of clinical letters.

Machine learning methods vary based on the choice of models and the features considered. In model selection, the multi-type relation extraction task can be assigned to several independent binary classifiers, each making the decision of whether a certain type of relation exist or not. A basic binary classifier such as logistic regression with ridge regularization is capable of performing relation extraction in this scenario. Panyam et al. (2016) used support vector machines (SVM) with a dependency graph kernel to perform relation extraction on two biomedical relation extraction tasks, showing competitive results. Brown Clustering (Brown et al., 1992) is a hierarchical approach to clustering words into classes through maximizing mutual information of bi-grams; it showed competitive performances in many NLP tasks (Turian et al., 2010). Nguyen and Verspoor (2018) implemented a method using character-based word embeddings which can capture unknown words within the context, coupled with CNN and LSTM neural network models. This approach obtained state-of-the-art performance in extracting chemical-disease relations on the BioCreative-V CDR corpus (Li et al., 2016).

For feature engineering, text features can generally be divided into the two categories of lexical features and syntactic features. Typical features used in other relation extraction tasks are summarized here.

- **Bag-of-words (BoW) features** based on white-space delimited tokens, are used in many tasks as a starting point.

- **Bag-of-concepts (BoC) features** (Sahlgren and Cöster, 2004) represent the text in terms of concepts, that is, phrases in the text that correspond to meaningful units. The current methods for generating concepts are
based on techniques such as mutual information (Sahlgren and Cöster, 2004), or through dictionary-based strategies (Funk et al., 2014). For clinical texts, the MetaMap tool (Aronson, 2001) is often used to recognize clinical concepts.

- **Syntactic features** take sentence structure into account. For example, ReLEx (Fundel et al., 2006) uses dependency parse trees associated with small numbers of rules in extracting relations from MEDLINE abstract and reaches an overall 80% precision and recall. Approximate Sub-graph Matching (ASM) (Liu et al., 2013) enables sentences to be matched by considering the similarity of the structure of dependency parse subgraphs that connect relevant entities to subgraphs in the training data.

- **Word embeddings** aim to capture word semantics through lower-dimension projections of word contexts and can be used to find word synonyms by measuring cosine similarity between word vectors. There are two widely used approaches to train word embeddings, co-occurrence matrix based methods such as GloVe (Pennington et al., 2014), and learning-based methods using skip-grams (Mikolov et al., 2013b) and CBOW (Mikolov et al., 2013a).

### 3 Methods

We improve the approaches described above to achieve better efficiency in relation extraction in our context of a narrow knowledge domain with limited data, specific to cancer follow-up treatment.

#### 3.1 Corpus

We consider a previously introduced corpus related to breast cancer follow-up treatment, randomly sampled and manually annotated by two physicians (Pitson et al., 2017). The corpus contains around 1000 sentences and 47,186 tokens. Despite its small size, the corpus is richly annotated with 16 medical named entity types and 16 types of semantic relations linking those entities with over 1,500 relation occurrences. Entities within the corpus are related to clinical therapies, temporal events, diseases, and so on. The annotation of clinical relations includes the associations between identified entities. For example, in the context "She remains on Arimidex tablets.", "remains on" is a TimeDescriptor, and "Arimidex" is an EndocrineTherapy. The relation TherapyTiming holds between these two entities in this context (e.g., TherapyTiming(TimeDescriptor, EndocrineTherapy)). For conciseness, we use abbreviations to refer to the entity types; hence we will use TD to represent the entity type of TimeDescriptor. While the dataset hasn’t been published yet, the full terminology list of entity types can be found in Pitson et al. (2017). To focus exclusively on the relation extraction task, we decouple the named entity recognition task from the relation extraction task by utilizing the gold standard entity annotations from the corpus.

#### 3.2 Method 1: Typed Sentential Co-occurrence

The simplest rule-based approach, given typed named entities, is to extract every pair of entities in a document that satisfies the type constraints of a relation. Such an approach yields high recall but poor precision, due to lack of use of context needed to ensure that a specific relation is expressed as holding between the entities. For example, in our data, only the semantic relation of Toxicity is defined as connecting a Therapy to a ClinicalFinding (expressing that a therapy was found to cause a specific toxic effect) and so it might seem reasonable to assume that a Toxicity event is being expressed in a document where both a Therapy and a ClinicalFinding are mentioned. However, the occurrence of these two entity types together in a document do not strictly indicate an occurrence of Toxicity relation between them; the entities may be connected to other mentioned entities via different relations. Hence assuming a Toxicity relation between them would result in a false positive.

We used intra-sentential constraints as described by (Verspoor et al., 2016) to improve precision by only considering that named entities that co-occur in the same sentence can have valid semantic relations. With this intra-sentential co-occurrence constraint, the relation extraction performance of the sentential baseline achieved strong recall and competitive precision, as well as reasonable overall F-score, on the Variome corpus (Verspoor et al., 2013).

We introduce the approach of Window-
Bounded Co-occurrence (WBC) to explore the impact of relaxing the constraint that the sentence boundary defines the scope of a relation. WBC defines a context window for a relation as the expansion to a base window of the sentence where an entity of the appropriate type occurs, and the adjustment of tokens beyond that sentence within which the related entity appears. We assume the occurrence of an inter-sentential relation relies on the distance of two entities, where distance is defined based on the number of tokens considered beyond the base sentence in which an entity occurs. We introduce a hyperparameter, $\rho$, to represent the distance of an entity pair in a context, namely the number of tokens allowed in exceeding the single base sentence. $\rho = 0$ denotes the entity pair is intra-sentential; $\rho = x$ denotes the second entity in a pair is $x$ number of tokens away from the base window. If $\rho$ is large than the length of the second context window, then the context window will set to the scope of two sentences by default.

Figure 1 shows an example when $\rho = 5$, WBC allows for the extraction of the semantic relation of TestToAssess across the base window containing the entity of ClinicalFinding, and into the expanded window encompassing the subsequent sentence and containing the TestName entity.

3.3 Method 2: Supervised Binary Classification Approach

We adopt a traditional pipeline as the architecture of the relation extraction system. Each stage is introduced below.

3.3.1 Data Preparation

Considering the semantic variation in the texts, and the small number of examples, training several independent binary classifiers is more robust for mining individual type of semantic relation patterns. Therefore, we transform the original dataset into 16 independent subsets, by grouping instances by their relation type. An instance consists of a typed entity pair, one or two sentence(s) with the relevant named entities inside as context, a label as an indication of relation occurrence, and a treatment letter id for mapping its position in the original dataset. We remove four relation types with fewer than 10 annotated instances, specifically the relations Intervention, EffectOf, RecurLink and GetOpinion.

We apply context selection for generating instances during data transformation. One instance represents the occurrence of a single semantic relation, containing one relevant entity pair. The context for each instance is the entire raw text of the sentence where the entity pair appears. In cases where more than one entity pair occurs in the same text context, we generate an independent instance for each entity pair. Where cross-sentence relations occur, we concatenate the two sentences containing the relevant entities into a single sentence, structurally indicating that the two sentences are related. In each instance, we replace the two named entity phrases with their type. In the case of overlapping named entities, such as where one named entity partially or completely collides with another entity, both types are retained, adjacent to each other in the text. Other entities mentioned in the context not relevant for the specific relation are left in their original textual form.

We use NLTK (Bird et al., 2009) to perform tokenization and lemmatization to normalize the representation of the text, and strip punctuation. We use the Snowball English stopword list to remove stopwords.

Further details are presented in the feature engineering section below.

3.3.2 Feature Engineering

We implement a set of traditional semantic features and three main feature sets based on ASM, BoC, and sentence embeddings in the sections below.

- Traditional Semantic Features

  The traditional semantic features includes bag-of-words (count-based), lemmas (base, uninflected form of a noun or verb), algebraic expressions, named entity type (derived from the

1http://anoncvs.postgresql.org/cvsweb.cgi/pgsql/src/backend/snowball/stopwords/
gold-standard), POS tags and dependency parse
based on Stanford CoreNLP (Manning et al.,
2014), and a transformation from dependency
parse tree to graph using NetworkX (Hagberg
et al., 2008) where edges are dependencies and
nodes are tokens/labels.

- **ASM features**
  The classical ASM measurement was developed by Liu et al. (2013), and was later extended to
  kernel method by Panyam et al. (2016). The
  ASM kernel was applied to the chemical in-
duced disease (CID) task (Wei et al., 2015) and
  SeedeV shared tasks (Chaix et al., 2016). The
  performance of ASM significantly depends on
  the result of POS-tagging and dependency tree
  parsing. All nodes are normalized to their lem-
as. Here, Stanford CoreNLP (Manning et al.,
2014) is used for POS tagging and dependency
tree parsing of the text. The context is split into
sentences before dependency parsing is applied
on individual sentences.

We produce the ASM features following Pan-
yam et al. (2016). Where the context includes
two sentences, a dummy root node is introduced
to connect the root nodes of two dependency
parse trees. Figure 2 shows an example. After
pre-processing, the dependency tree structure is
transformed into a graph where nodes are lem-
as with their POS tags, edges are dependen-
cies across lemmas within sentence. Then, a
shortest path algorithm is applied on the depend-
dency graph to generate flat features. In cases
where sentences are very long, processing time
is unacceptable, and no ASM features are gen-
erated.

- **Bag-of-Concepts features**
  Word embeddings are used to capture word
similarity based on shared surrounding context.
The size of the surrounding context, known
as window size control, varies the representa-
tion of word embeddings from more semantic
(shorter window size) to more syntactic (longer
window size). Synonyms can be identified by
identifying two words with similar embeddings,
based on cosine similarity measurement. Over-
fitting can occur for word embeddings, where
a training corpus is not large enough or a cor-
pus is limited to a narrow domain of knowledge.
Therefore, instead of training word embeddings
on our corpus, we use two publicly available
pre-trained word embeddings, GloVe (Penning-
ton et al., 2014) and a Wikipedia-PubMed-
PMC embedding (Moen and Ananiadou, 2013)
to capture more clinically relevant vocabulary.
The vocabulary of word embeddings denotes
the total number of words that are represented.
In our experiment, only the top 20,000 most fre-
quent lemmas are selected. Gensim (Rehurek
and Sojka, 2010) is used to find the synonyms
of a lemma from the vocabulary by measuring
similarity between GloVe word vectors.

We then implement an algorithm for build-
ing BoC. Using a word2concept algorithm (see
Equation 1), we map a lemma (key) to a con-
cept (value) based on the embedding of the
lemma expressed as $E(\text{lemma})$ and the simi-
arity threshold expressed as $\mu$ as a tunable hyper-
parameter.

$$CONCEPT = f(E(\text{lemma}), \mu)$$ (1)

The algorithm starts by extracting BoW features
for each generated instance after data prepara-
tion process into a list $L$. Then, starting from
the first lemma $w_1$ from $L$, we retrieve its em-
bedding $x_{w_1} = E(w_1)$. We then retrieve a
new lemma $w_i$ and its embedding $x_{w_i}$ from
the vocabulary $V$, and calculate the similarity
score $S = \cos(x_{w_1}, x_{w_i})$. If $S \geq \mu$, cre-
ate mappings between $w_1 \rightarrow \text{concept}_1$ and
$w_i \rightarrow \text{concept}_i$. If no $w_i$ satisfies the condition
of $S \geq \mu$, then $w_1$ will be kept in its original
form. Next, we move to the second word $w_2$ in
$L$, check whether $w_2$ has already been mapped
to a concept $\text{concept}_s$, and if so, directly create
the mapping $w_2 \rightarrow \text{concept}_s$. Otherwise, we
iterate.

Note that in this model, named entities will ef-
fectively be treated as out-of-vocabulary terms,
since they have been mapped to and replaced with the names of the relevant entity types (e.g., “TestName” which is not a token that would be expected to be represented in any pre-trained word embedding model).

- **Sentence embedding features**
  Apart from being a tool for finding word synonyms and generating BoC data representation, word embeddings can also be used to obtain sentence embeddings through weighted average pooling. If $S$ denotes a sentence, $E(S)$ denotes the embedding of sentence $S$. $E(w_1)$ denotes the embedding of the first word $w_1$ of sentence $S$, $Score(w_1)$ denotes the TF-IDF score of word $w_1$, the sentence embeddings based on weighted average pooling can be expressed as Equation 2. We calculate TF-IDF scores of each word using the original documents, as each instance has an index to its original document id. Out-of-vocabulary tokens and gold standard named entities will be ignored. However, entity information has been considered during that data preparation stage, because the generation of instances takes gold standard entities into account.

$$E(S) = \frac{1}{n} \sum_{i=1}^{n} E(w_i) \times Score(w_i) \quad (2)$$

### 3.4 Supervised Learning Models
We build individual binary classifiers for each relation type. We introduce the SVM classifiers and Feed-forward ANN models briefly here.

- **Support SVM Machine** We select the general SVM model (Hearst, 1998) and kernels provided by scikit-learn (Pedregosa et al., 2011). For SVM kernels, we integrate ASM kernel as part of feature engineering, to avoid colliding with the use of the linear kernel and the RBF (Radial Basis Function) kernel.

- **Feed-forward ANN** We use Keras (Chollet et al., 2015) to construct a simple feed-forward neural network model with two fully connected layers as shown in Figure 3.

The dimension of each input and the number of hidden units in each layer is the same as the dimension of feature vectors under each type of relation. The activation function for the first dense layer is ReLU, and for the second dense layer is softmax.

### 3.5 Experiment Design
Considering the dataset is small, we split the transformed dataset into three independent combinations of training, development, and test sets with the ratio of 6:1:3.

In training stage, we train independent models for each specific relation type. We use cross-validation and grid search to tune the hyperparameters of the classifiers.

In prediction stage, the decision of applying sentence-bounded or window-bounded approach is made by setting the size of sliding window. Setting the window size to 0 will apply the sentence-bounded co-occurrence constraint. We choose window sizes of 0, 5, 10 to explore the value of additional context. In supervised binary classification approach, utilizing a similarity threshold of 1 leads to strict use of BoW (word) features, while relaxing the similarity threshold $\mu$ of 0.9, 0.8 will generate BoC (concepts). We compare the influence of different word embeddings in generating BoC based on their relation extraction performance on the test set.

Both rule-based approach and supervised binary classification approach will make predictions on the same test set, which allows empirical comparison between rule-based and machine learning approaches.

We compare the impact of increasing data size for BoW and BoC by sub-sampling the training set into nine instance-incremental and non-overlapping sub-sets (combining them into sets representing 10% to 90% of the original training set) and visualize the performance variation. We explore whether word embeddings as a medium for generating BoC are more effective than the direct use of sentence embeddings in cases where the dataset is small and knowledge domain is restricted to the specific domain of breast cancer.
treatment. We also explore the combination of BoC and sentence embeddings feature, in order to investigate how best to make integration of them.

Finally, we explore the possibility of applying more complex models for analysis of our small and specific knowledge domain dataset. We start from simple linear models including logistic regression and lin-SVM, then apply rbf-SVM, feed-forward ANN on Keras with 32 batch per time, and epochs of 2, 10, 100, and 500 for each relation type.

3.6 Evaluation

We evaluate both overall and per relation type performance. Evaluation of the two approaches is performed on the same test set derived from the data preparation stage. In addition, considering the dataset is small, we calculate the mean score from three independent combinations of training, development and test sets. We evaluate results using micro F\_1-measure since the number of positives and negatives were highly imbalanced across all relation types. We finally evaluate the performance growth over nine sub-sampled training sets of increasing size.

4 Result and Discussion

We present experiment results with micro F\_1-measurement. After experimenting with different similarity threshold values to generate BoC features, the best performance is achieved when the threshold is set to 0.9 (results not shown). Word embeddings derived from Wiki-PubMed-PMC outperform GloVe-based embeddings (Table 1). The models using BoC outperform models using BoW as well as ASM features. As shown in Table 2, the intra-sentential co-occurrence baseline outperforms other approaches which allow boundary expansion. This is because a majority of relations in the corpus are intra-sentential.

A visualization of the growth in performance for both BoW and BoC-based models as training set size increases, over 12 relation types, based on micro F\_1, is shown in Figure 4. The results of BoC in this figure is collected from lin-SVM\_Wiki-PubMed-PMC, \(\mu = 0.9\), and BoW is collected from lin-SVM. We find BoC tends to outperform BoW with only a small number of training instances, and also performs better than BoW with incremental training instance.

Figure 4: Performance variation of BoW and BoC(Wiki-PubMed-PMC, \(\mu = 0.9\)) with increasing data fractions, under linSVM, F\_1 measure.

The reason that Wikipedia-PubMed-PMC embeddings (Moen and Ananiadou, 2013) outperforms GloVe (Mikolov et al., 2013a) in the extraction of most relation types (Table 1) is because its training corpus has a more similar domain and vocabulary as our dataset. Therefore, it leads to more relevant models of the distributional semantics of words. On the other hand, the GloVe embeddings are derived from a more general corpus; thus the semantics of domain specific terms in our dataset are not captured.

By observation, most lemmas that map into concepts are digits, time stamps, common verbs and medical terminologies. For example, in TimeStamp\((TD, TP)\), drugs with similar effects such as “letrozole” and “anastrozole” map into “CONCEPT\_3”; all single digits, ranges from 0-9 map into “CONCEPT\_26”; year tags such as “2011, 2009” map into “CONCEPT\_56”. We consider the cause of differences between the BoW and BoC representations. In the normalization process for BoW, stop-words collected from general knowledge domain are not well-suited to the knowledge domain for our specific task, while limited data does not allow the construction of an appropriate stop-words list from the data. In contrast, BoC models normalize the differences between individual words with shared meaning. While intuitively this should support an improvement over BoW models, we find BoW outperforms BoC when extracting certain relations such as TestTiming\((TN, TD)\), TestToAssess\((TN, TR)\) with gold standard named entities as arguments. The reason is that synonyms may express slightly different meanings; concept mapping discards such differences and leads to information loss, potentially causing more mis-classifications. The deci-
### Table 1: Performance of supervised learning models with different features.

<table>
<thead>
<tr>
<th>Feature</th>
<th>LR</th>
<th>SVM</th>
<th>ANN</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>P</td>
<td>R</td>
<td>F1</td>
</tr>
<tr>
<td>+BoW</td>
<td>0.93</td>
<td>0.91</td>
<td>0.92</td>
</tr>
<tr>
<td>+BoC (Wiki-PubMed-PMC)</td>
<td>0.94</td>
<td>0.92</td>
<td><strong>0.93</strong></td>
</tr>
<tr>
<td>+BoC (GloVe)</td>
<td>0.93</td>
<td>0.92</td>
<td>0.92</td>
</tr>
<tr>
<td>+ASM</td>
<td>0.90</td>
<td>0.85</td>
<td>0.88</td>
</tr>
<tr>
<td>+Sentence Embeddings (SEs)</td>
<td>0.89</td>
<td>0.89</td>
<td>0.89</td>
</tr>
<tr>
<td>+BoC (Wiki-PubMed-PMC) + SEs</td>
<td>0.92</td>
<td>0.92</td>
<td>0.92</td>
</tr>
</tbody>
</table>

### Table 2: F1 score results per relation type of the best performing models.

<table>
<thead>
<tr>
<th>Relation type</th>
<th>Count</th>
<th>Intra-sentential co-occ.</th>
<th>BoC (Wiki-PubMed-PMC)</th>
<th>LR</th>
<th>SVM</th>
<th>ANN</th>
</tr>
</thead>
<tbody>
<tr>
<td>TherapyTiming (TP, TD)</td>
<td>428</td>
<td>0.84</td>
<td>0.59</td>
<td>0.47</td>
<td>0.78</td>
<td>0.81</td>
</tr>
<tr>
<td>NextReview (Followup, TP)</td>
<td>164</td>
<td>0.90</td>
<td>0.83</td>
<td>0.63</td>
<td>0.86</td>
<td>0.88</td>
</tr>
<tr>
<td>Toxicity (TP, CF, TR)</td>
<td>163</td>
<td>0.91</td>
<td>0.77</td>
<td>0.55</td>
<td>0.85</td>
<td>0.86</td>
</tr>
<tr>
<td>TestTiming (TN, TD, TP)</td>
<td>184</td>
<td>0.90</td>
<td>0.81</td>
<td>0.42</td>
<td>0.96</td>
<td><strong>0.97</strong></td>
</tr>
<tr>
<td>TestFinding (TN, TP)</td>
<td>136</td>
<td>0.76</td>
<td>0.60</td>
<td>0.44</td>
<td>0.82</td>
<td>0.79</td>
</tr>
<tr>
<td>Threat (O, CF, TR)</td>
<td>32</td>
<td>0.85</td>
<td>0.69</td>
<td>0.54</td>
<td><strong>0.95</strong></td>
<td><strong>0.95</strong></td>
</tr>
<tr>
<td>Intervention (TP, YR)</td>
<td>5</td>
<td>0.88</td>
<td>0.65</td>
<td>0.47</td>
<td>-</td>
<td>-</td>
</tr>
<tr>
<td>EffectOf (Com, CF)</td>
<td>3</td>
<td>0.92</td>
<td>0.62</td>
<td>0.23</td>
<td>-</td>
<td>-</td>
</tr>
<tr>
<td>Severity (CF, CS)</td>
<td>75</td>
<td>0.61</td>
<td>0.53</td>
<td>0.47</td>
<td>0.52</td>
<td>0.55</td>
</tr>
<tr>
<td>RecurLink (YR, YR, CF)</td>
<td>7</td>
<td>1.0</td>
<td>1.0</td>
<td>0.64</td>
<td>-</td>
<td>-</td>
</tr>
<tr>
<td>RecurInfer (NR, YR, TR)</td>
<td>51</td>
<td>0.97</td>
<td>0.69</td>
<td>0.43</td>
<td><strong>0.99</strong></td>
<td><strong>0.99</strong></td>
</tr>
<tr>
<td>GetOpinion (Referral, CF, other)</td>
<td>4</td>
<td>0.75</td>
<td>0.75</td>
<td>0.5</td>
<td>-</td>
<td>-</td>
</tr>
<tr>
<td>Context (Dis, DisCont)</td>
<td>40</td>
<td>0.70</td>
<td>0.63</td>
<td>0.53</td>
<td>0.60</td>
<td>0.41</td>
</tr>
<tr>
<td>TestToAssess (TN, CF, TR)</td>
<td>36</td>
<td>0.76</td>
<td>0.66</td>
<td>0.36</td>
<td><strong>0.92</strong></td>
<td><strong>0.92</strong></td>
</tr>
<tr>
<td>TimeStamp (TD, TP)</td>
<td>221</td>
<td><strong>0.92</strong></td>
<td>0.83</td>
<td>0.50</td>
<td>0.86</td>
<td>0.85</td>
</tr>
<tr>
<td>TimeLink (TP, TP)</td>
<td>20</td>
<td>0.92</td>
<td>0.85</td>
<td>0.45</td>
<td>0.91</td>
<td><strong>0.92</strong></td>
</tr>
<tr>
<td>Overall</td>
<td>1569</td>
<td>0.90</td>
<td>0.73</td>
<td>0.45</td>
<td>0.92</td>
<td><strong>0.93</strong></td>
</tr>
</tbody>
</table>

The choice of whether to use the BoC or BoW will depend on the characteristics of particular relation types.

Table 1 also shows the combination feature of BoC and sentence embeddings outperforms sentence embeddings alone, but do not exceed the upper boundary of BoC feature, in which again demonstrating the competitiveness of BoC feature.

Since this corpus is much smaller than other narrow knowledge domain corpus such as CID (Wei et al., 2015) and Seedev shared task (Chaix et al., 2016), the training instances are not enough for the learners to generalize well using syntactic representation. Therefore, the models using ASM kernel (Panyam et al., 2016) do not outperform the simple linear classifiers.

As the results of applying the co-occurrence baseline (\(\rho = 0\)) shows (Table 2), the semantic relations in this data are strongly concentrated within a sentence boundary, especially for the relation of RecurLink, with an F1 of 1.0. The machine learning approaches based on BoC lexical features effectively complement the deficiency of cross-sentence relation extraction.

Lin-SVM outperforms other classifiers in extracting most relations. The feed-forward ANN displays significant over-fitting across all relation types, as the performance decreases when increasing the training epochs. Specifically, with only two training epochs, the performance of ANN is still slightly worse than lin-SVM. The result of lin-SVM present its robustness of avoiding over-fitting compares to feed-forward ANN with BoW, BoC, ASM flat features and sentence embeddings.
5 Conclusion and Future Work

We proposed two ways to perform relation extraction for a narrow knowledge domain, with only small available data set. We implemented a window-based context approach and experiment with determining the best context size for the relation extraction in the rule-based settings. The typical sentential co-occurrence baseline is competitive when most relations are intra-sentential. We implemented a BoC feature engineering method, by leveraging word embeddings as a tool for finding word synonyms and mapping them to concepts. BoC feature outperforms BoW, ASM syntactic feature and sentence embeddings derived by weighted average pooling across word embeddings in small dataset with respect to its significant improvements in micro F1 score. In addition, it would be expected to show competitive results on other relation extraction tasks where it is useful to generalize specific tokens such as digits or time stamps.

We also explored the performance of models with different level of complexity, such as logistic regression, lin-SVM, rbf-SVM, and a simple feed-forward ANN. The results highlight that strategies to avoid over-fitting must be considered since the number of training instances is limited.

In future work, we will explore a number of directions, including some unsupervised learning approaches. We will test the performance of BoC on other corpora, to explore BoC vs. BoW as a baseline data representation. We will address comparisons between BoC and other word clustering methods such as Brown Clustering (Brown et al., 1992). Finally, the integration of current named entity recognition tools and end-to-end relation extraction, to remove the reliance on gold standard named entity annotations, will also be explored.

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Generating Text through Adversarial Training
using Skip-Thought Vectors

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Abstract

GANs have been shown to perform exceedingly well on tasks pertaining to image generation and style transfer. In the field of language modelling, word embeddings such as GLoVe and word2vec are state-of-the-art methods for applying neural network models on textual data. Attempts have been made to utilize GANs with word embeddings for text generation. This study presents an approach to text generation using Skip-Thought sentence embeddings with GANs based on gradient penalty functions and f-measures. The proposed architecture aims to reproduce writing style in the generated text by modelling the way of expression at a sentence level across all the works of an author. Extensive experiments were run in different embedding settings on a variety of tasks including conditional text generation and language generation. The model outperforms baseline text generation networks across several automated evaluation metrics like BLEU-n, METEOR and ROUGE. Further, wide applicability and effectiveness in real life tasks are demonstrated through human judgement scores.

1 Introduction

Inducing a particular style in generated text is a promising development which can lead to producing acceptable responses in dialogue generation, image captioning and artificial chat bot systems. In unsupervised text generation, estimating the distribution of real text from a corpus is a challenging task. Recent approaches using adversarial training have addressed this issue by trying to overcome the exposure bias that models trained for maximum likelihood suffer from. This work proposes an approach for text generation using a Generative Adversarial Network (GAN) with Skip-Thought vectors (STGAN). GANs (Goodfellow et al., 2014) are a class of neural networks that explicitly train a generator to produce high-quality samples by pitting the generator against an adversarial discriminative model. GANs output differentiable values and the task of discrete text generation is challenging because of the non-differentiable nature of generating discrete symbols. Hence, in the present work, the GANs are trained with sentence embedding vectors as a differentiable input. The sentence embeddings are produced using Skip-Thought (Kiros et al., 2015), a neural network model for learning fixed length representations of sentences.

People’s way of expression and communication intention is more diverse across utterances than the vocabulary. To imitate this, the proposed STGAN architecture models the variability at the utterance level in a corpus rather than at word or character level. The effectiveness of this approach is evaluated on automated corpus-based metrics: BLEU-n (Papineni et al., 2002), METEOR (Banerjee and Lavie, 2005) and ROUGE (Lin, 2004) using different embeddings: Average GloVe (Pennington et al., 2014), Vector Extrema GloVe (Pennington et al., 2014) and Skip-Thought (Kiros et al., 2015). We perform an empirical study with human judgements to assess both the quality and the style reproduction in the generated text.

2 Related Works

Deep neural network architectures have demonstrated strong results on natural language generation tasks such as dialogue response generation and machine translation. Early techniques for generating text conditioned on some input information were template or rule-based engines (McRoy et al., 2000), or probabilistic models such as n-gram. In the recent past, state-of-the-art results on these tasks have been achieved by recurrent (Press et al., 2017; Mikolov et al., 2010) and con-
volutional neural network models trained for likelihood maximization. Very recently, attempts have been made to generate text using purely generative adversarial training (Arjovsky et al., 2017).

Unsupervised learning with deep neural networks in the framework of encoder-decoder models has become the state-of-the-art methods for approaching NLP problems (Young et al., 2017). Recent text generation models have used a wide variety of GANs such as policy-gradient based sequence generation framework (Yu et al., 2016). Fedus et al. (2018) have used an actor-critic conditional GAN to fill in missing text conditioned on the surrounding text for natural language generation tasks. GANs have also been used for text style transfer by Yang et al. (2018) where language models act as the discriminator and by Chen et al. (2018) with the introduction of a new f-measure termed as feature-mover’s distance.

Using adversaries of word and character level embeddings for text generation has been explored by Rajeswar et al. (2017). Models trained using generative adversarial networks or variational autoencoders have been shown to learn representations of continuous structures by leveraging deep latent variables such as text embeddings (Zhao et al., 2017). This work explores injecting sentence embeddings produced using the Skip Thought architecture (Kiros et al., 2015) into GANs in different setups.

3 Skip-Thought Generative Adversarial Network

In literature corpora such as fantasy and science fiction novels, the vocabulary does not vary significantly across the authors, but the manner of expression does, which is intuitively best captured at the level of sentences than words. The approach, hence, that this work takes in generating sentences with the writing style of one author is to make the adversarial model approximate the distribution of all sentences (rather than words or characters) in a latent space using skip-thought architecture. Previous attempts on text generation have used the character and word-level embeddings instead with GANs (Rajeswar et al., 2017).

This section introduces Skip-Thought Generative Adversarial Network with a background on neural network models that it is based on. The Skip-Thought model (Kiros et al., 2015) produces embedding vectors for sentences present in training corpus. These vectors constitute the real distribution for the discriminator network. The generator network produces sentence vectors similar to those from the encoded real distribution. The generated vectors are sampled over the course of training and then decoded to produce sentences using a Skip-Thought decoder conditioned on the same text corpus.

3.1 Skip-Thought Vectors

Skip-Thought is an encoder-decoder framework with an unsupervised approach to train a generic, distributed sentence encoder. The encoder maps sentences sharing semantic and syntactic properties to similar vector representations and the decoder reconstructs the surrounding sentences of an encoded passage. The sentence encoding approach draws inspiration from the skip-gram model in producing vector representations using previous and next sentences.

The Skip-Thought model uses an RNN encoder with GRU activations (Chung et al., 2014) and an RNN decoder with conditional GRU. This combination is identical to the RNN encoder-decoder of Cho et al. (2014) used in neural machine translation.

Skip-Thought Architecture

For a given sentence tuple \((s_{i-1}, s_i, s_{i+1})\), let \(w_i\) denote the \(t\)-th word in sentence \(s_i\) and \(x_i\) denote its word embedding. The model is described as:

**Encoder.** Encoded vectors for a sentence \(s_i\) with \(N\) words \(w_i, w_{i+1}, ..., w^n\) are computed by iterating over the following sequence of equations:

\[
\begin{align*}
  r^t &= \sigma(W_r x^t + U_r h^{t-1}) \\
  z^t &= \sigma(W_z x^t + U_z h^{t-1}) \\
  h^t &= \tanh(W x^t + U (r^t \odot h^{t-1})) \\
  h^t &= (1 - z^t) \odot h^{t-1} + z^t \odot h^t
\end{align*}
\]

where \(h^t\) is a hidden state at each time step and interpreted as a sequence of words \(w^n_i, ..., w^1_i\), \(h^t\) is the proposed state update at time \(t\), \(z^t\) is the update gate and \(r^t\) is the reset gate. Both update gates take values between zero and one.

**Decoder.** A neural language model conditioned on the encoder output \(h_t\) serves as the decoder. Bias matrices \(C_s, C_r, C\) are introduced for the update gate, reset gate and hidden state computation by the encoder. Two decoders are used in parallel,
Figure 1: Skip-Thought Generative Adversarial Network model architecture

one each for sentences $s_{i+1}$ and $s_{i-1}$. The following equations are iterated over for decoding:

$$r^t = \sigma(W^d r^t + U^d h^{t-1} + C_r h_i)$$

$$z^t = \sigma(W^d z^t + U^d h^{t-1} + C_z h_i)$$

$$h^t = \tanh(W^d x^{t-1} + U^d (r^t \circ h^{t-1}) + C_h h_i)$$

$$h_{i+1} = (1 - z^t) \circ h^{t-1} + z^t \circ h^t$$

**Objective.** For the same tuple of sentences, objective function is the sum of log-probabilities for the forward and backward sentences conditioned on the encoder representation:

$$\sum_t \log P(w_{i+1}^t|w_{i+1}^t, h_i) + \sum_t \log P(w_{i-1}^t|w_{i-1}^t, h_i)$$

### 3.2 Generative Adversarial Networks

Generative Adversarial Networks (Goodfellow et al., 2014) are deep neural net architectures comprised of two networks, competing with each other in a zero-sum game framework. For a given data, GANs can mimic learning the underlying distribution and generate artificial data samples similar to those from the real distribution. Generative Adversarial Networks consists of two players: a Generator and a Discriminator. The generator $G$ tries to produce data close to the real distribution $P(x)$ from some stochastic distribution $P(z)$ termed as noise. The discriminator $D$’s objective is to differentiate between real and generated data $G(z)$.

The two networks - generator and discriminator compete against each other in a zero-sum game. The minimax strategy dictates that each network plays optimally with the assumption that the other network is optimal. This leads to Nash equilibrium which is the point of convergence for GAN model.

The STGAN architecture (Figure 1) has two components: Skip Thought encoder-decoder and a generative adversarial network. The model uses a deep convolutional generative adversarial network, similar to the one used in DCGAN (Radford et al., 2015). During the training, the generator network is updated twice for each discriminator network update to prevent fast convergence of the discriminator network.

The Skip-Thought encoder for the model encodes sentences using 2400 GRU units (Chung et al., 2014) with a word vector dimensionality of 620. The encoder combines the sentence embeddings to produce 4800-dimensional combine-skip vectors with the first 2400 dimensions being uni-skip model and the last 2400 bi-skip model. This work uses the 4800-dimensional vectors as they have been found to be the best performing in experiments.

For training of the STGAN, the discriminator network update to prevent fast convergence of the GAN discriminator.

The decoder uses greedy decoding by taking the argmax over the softmax output distribution for a given time-step which also acts as input for next time-step. It reconstructs sentences conditioned on a sentence vector by randomly sampling

---

Goodfellow et al. (2014) have formulated the minimax game for a generator $G$, discriminator $D$ adversarial network with value function $V(G, D)$ as:

$$\min_G \max_D V(G, D) = E_{x \sim p_{data}(x)}[\log D(x)] + E_{z \sim p_z(z)}[\log (1 - D(G(z)))]$$

### 3.3 Model Architecture

The STGAN architecture (Figure 1) has two components: Skip Thought encoder-decoder and a generative adversarial network. The model uses a deep convolutional generative adversarial network, similar to the one used in DCGAN (Radford et al., 2015). During the training, the generator network is updated twice for each discriminator network update to prevent fast convergence of the discriminator network.

For training of the STGAN, the discriminator network update to prevent fast convergence of the GAN discriminator.

The decoder uses greedy decoding by taking the argmax over the softmax output distribution for a given time-step which also acts as input for next time-step. It reconstructs sentences conditioned on a sentence vector by randomly sampling
from the predicted distributions with a preset beam width. A 620 dimensional RNN word embedding is used for the decoder with 1600 hidden GRU decoding units. All experiments are performed using the Adam optimizer (Kingma and Ba, 2014) with gradient clipping and a batch size of 16.

Each sentence is appended with a start token $<$s$>$ and an end token $<$ /s$>$ before encoding. During the process of training generator network with these embeddings, some generated vectors are randomly sampled. The sampled vectors are decoded using pretrained Skip-Thought decoder to produce a probability distribution over the vocabulary in order to reconstruct sentences. The decoding is terminated when the stop token $<$ /s$>$ is encountered during reconstruction.

3.4 Improving Training and Loss
The training process of a GAN is notably difficult (Salimans et al., 2016) and several improvement techniques such as batch normalization, feature matching, historical averaging (Salimans et al., 2016) and unrolling GAN (Metz et al., 2016) have been suggested for making the training more stable. Training the Skip-Thought GAN often results in mode dropping (Arjovsky and Bottou, 2017; Srivastava et al., 2017) with a parameter setting where it outputs a very narrow distribution of points. To overcome this, it uses minibatch discrimination by looking at an entire batch of samples and modeling the distance between a given sample and all the other samples present in that batch.

The minimax formulation for an optimal discriminator in a vanilla GAN is Jensen-Shannon Distance between the generated distribution and the real distribution. Arjovsky et al. (2017) used Wasserstein distance or earth mover’s distance to demonstrate how replacing distance measures can improve training loss for a GAN. Gulrajani et al. (2017) have incorporated a gradient penalty regularizer term in WGAN objective for discriminator’s loss function. The experiments in this work use the above $f$-measures to improve performance of Skip-Thought GAN on text generation.

4 Results and Discussion
4.1 Conditional Generation of Sentences.
GANs can be conditioned on certain attributes to generate real valued data (Mirza and Osindero, 2014; Radford et al., 2015). In this experiment, both the generator and discriminator are conditioned on the Skip-Thought encoded vectors.

The data used for this setup consists of 250,000 sentences chosen from the BookCorpus dataset (Zhu et al., 2015) with a training/test/validation split of 5/1/1. All the sentences belong to one series of fantasy novels by a particular author of English language. This selection implies that the author’s word choice, sentence structure, figurative language, and sentence arrangement are consistent and well-represented across the dataset. Conditioning on this high-level outline gives more robustness to the model in terms of generated samples.

The decoded sentences form the evaluation set for measuring performance of different models under corpus level BLEU-n (Papineni et al., 2002), METEOR (Banerjee and Lavie, 2005) and ROUGE (Lin, 2004) metrics. Table 1 compares these results for the proposed STGAN against standard LSTM and Attention Bidirectional LSTM models in two settings - using Skip Thought vectors and using tied GloVe (Pennington et al., 2014) embeddings. For this experiment, GloVe Average is obtained by averaging GloVe embeddings of all the words composing a given sentences while GloVe Extreme is computed by taking the most extreme value for each dimension across embeddings of all the words. All the three models used in the experiment: LSTM, Attention BiLSTM and Wasserstein GAN take the above three embeddings as input in separate runs and output vectors which are decoded to reconstruct sentences using the corresponding embedding’s decoder.

Table 2 shows improvements in metric scores when using Wasserstein distance and gradient penalty regularizer as discussed in section 3.4. WGAN-GP gives the strongest across-the-board performance in both the GloVe and Skip-Thought settings, so we use this as the basis for the rest of our experiments. The METEOR scores are reportedly better for other models because though it does not rely on embeddings but it includes notions of synonymy and paraphrasing to compute alignment between hypothesis and reference sentences (Sharma et al., 2017).

4.2 Language Generation.
Language generation is performed on a dataset comprising simple English sentences referred to
### Table 1: Evaluation of models on word-overlap based automated metrics when trained with different embeddings.

<table>
<thead>
<tr>
<th>Model</th>
<th>Embedding</th>
<th>BLEU-1</th>
<th>BLEU-2</th>
<th>BLEU-3</th>
<th>BLEU-4</th>
<th>METEOR</th>
<th>ROUGE</th>
</tr>
</thead>
<tbody>
<tr>
<td>LSTM</td>
<td>GloVe Average</td>
<td>0.874</td>
<td>0.792</td>
<td>0.621</td>
<td>0.582</td>
<td>0.681</td>
<td>0.692</td>
</tr>
<tr>
<td></td>
<td>GloVe Extreme</td>
<td>0.874</td>
<td>0.791</td>
<td>0.616</td>
<td>0.580</td>
<td>0.677</td>
<td>0.685</td>
</tr>
<tr>
<td></td>
<td>Skip Thought</td>
<td>0.885</td>
<td>0.807</td>
<td>0.633</td>
<td>0.585</td>
<td>0.683</td>
<td>0.692</td>
</tr>
<tr>
<td>Attention BiLSTM</td>
<td>GloVe Average</td>
<td>0.904</td>
<td>0.836</td>
<td>0.645</td>
<td>0.583</td>
<td>0.695</td>
<td>0.698</td>
</tr>
<tr>
<td></td>
<td>GloVe Extreme</td>
<td>0.886</td>
<td>0.827</td>
<td>0.643</td>
<td>0.581</td>
<td>0.689</td>
<td>0.696</td>
</tr>
<tr>
<td></td>
<td>Skip Thought</td>
<td>0.900</td>
<td>0.827</td>
<td>0.651</td>
<td>0.589</td>
<td>0.692</td>
<td>0.715</td>
</tr>
<tr>
<td>WGAN-GP</td>
<td>GloVe Average</td>
<td>0.879</td>
<td>0.807</td>
<td>0.668</td>
<td>0.585</td>
<td>0.694</td>
<td>0.702</td>
</tr>
<tr>
<td></td>
<td>GloVe Extreme</td>
<td>0.853</td>
<td>0.799</td>
<td>0.666</td>
<td>0.579</td>
<td>0.689</td>
<td>0.697</td>
</tr>
<tr>
<td></td>
<td>Skip Thought</td>
<td>0.903</td>
<td>0.836</td>
<td>0.682</td>
<td>0.594</td>
<td>0.692</td>
<td>0.731</td>
</tr>
</tbody>
</table>

Skip-Thought gives better results than GloVe for BLEU-n and ROUGE metrics, while the METEOR scores are comparable to that when using averaged GloVe embedding with Attention BiLSTM generator.

### Table 2: BLEU-2, BLEU-3 METEOR and ROUGE metric scores across GAN models with different f-measures.

<table>
<thead>
<tr>
<th>Model</th>
<th>BLEU-2</th>
<th>BLEU-3</th>
<th>METEOR</th>
<th>ROUGE</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>GloVe</td>
<td>ST</td>
<td>GloVe</td>
<td>ST</td>
</tr>
<tr>
<td>GAN</td>
<td>0.710</td>
<td>0.745</td>
<td>0.593</td>
<td>0.607</td>
</tr>
<tr>
<td>WGAN</td>
<td>0.786</td>
<td>0.833</td>
<td>0.645</td>
<td>0.669</td>
</tr>
<tr>
<td>WGAN-GP</td>
<td>0.807</td>
<td>0.836</td>
<td>0.668</td>
<td>0.682</td>
</tr>
</tbody>
</table>

### 4.3 Human Scores and Correlations

The performance of this approach to generate new sentences has been evaluated in reproducing writing style of a particular author. The participant group consisted of 14 individuals, who were familiar with writing style of the said author by having read all but a few of the literary works of the author. The setup prevents them from being certain whether a sentence in question has or has not appeared in any work of the author that they have already read. To form the evaluation set of sentences, as CMU-SE\(^2\) (Rajeswar et al., 2017). The CMU-SE dataset consists of 44,016 sentences with a vocabulary of 3,122 words. The vectors are extracted in batches of same-lengthed sentences for encoding. The samples represent how mode collapse is manifested when using least-squares distance (Mao et al., 2016) f-measure without minibatch discrimination. Table 3(a) contains sentences generated from a vanilla STGAN which mode collapse is observed, while 3(b) contains examples wherein it is not observed when using minibatch discrimination. Table 3(c) shows generated samples from STGAN when using Wasserstein distance f-measure as WGAN (Arjovsky et al., 2017)) and 3(d) contains samples when using a gradient penalty regularizer term as WGAN-GP (Gulrajani et al., 2017). The two models generate longer human-like sentences and over a more diverse vocabulary.

---

\(^2\)https://github.com/clab/sp2016.11-731/tree/master/hw4/data

**Figure 2:** Pearson’s correlation coefficient between automated computed metrics and human scores. Human scores correlate well with BLEU-3 and ROUGE scores.
Table 3: Sentences sampled from STGAN when training on CMU-SE Dataset; mode collapse is overcome by using minibatch discrimination. Sample quality in terms of length and diversity further improved by using Wasserstein distance f-measure with gradient penalty regularizer. **WGAN:** Wasserstein GAN, **GP:** Gradient Penalty

<table>
<thead>
<tr>
<th>Model Generated Samples</th>
<th></th>
</tr>
</thead>
<tbody>
<tr>
<td>a. GAN (Mode collapse)</td>
<td>1. it ?</td>
</tr>
<tr>
<td></td>
<td>2. it ?</td>
</tr>
<tr>
<td></td>
<td>3. it ?</td>
</tr>
<tr>
<td></td>
<td>4. it ? how would it ?</td>
</tr>
<tr>
<td></td>
<td>5. it ? how would it ?</td>
</tr>
<tr>
<td>b. GAN (minibatch)</td>
<td>1. it a bottle ?</td>
</tr>
<tr>
<td></td>
<td>2. a glass bottle ?</td>
</tr>
<tr>
<td></td>
<td>3. a glass bottle it ?</td>
</tr>
<tr>
<td></td>
<td>4. it my hand a bottle ?</td>
</tr>
<tr>
<td></td>
<td>5. the phone my hand it</td>
</tr>
<tr>
<td>c. Skip Thought WGAN</td>
<td>1. we have new year’s holidays, always.</td>
</tr>
<tr>
<td></td>
<td>2. here you can nt see your suitcase,</td>
</tr>
<tr>
<td></td>
<td>3. please show me how much is a transfer?</td>
</tr>
<tr>
<td></td>
<td>4. i had a police take watch out of my wallet .</td>
</tr>
<tr>
<td></td>
<td>5. here i collect my telephone card and telephone number</td>
</tr>
<tr>
<td>d. Skip Thought WGAN-GP</td>
<td>1. my passport and a letter card with my card , please</td>
</tr>
<tr>
<td></td>
<td>2. here on my telephone, mr. kimuras registration cards address.</td>
</tr>
<tr>
<td></td>
<td>3. i can nt see some shopping happened .</td>
</tr>
<tr>
<td></td>
<td>4. get him my camera found a person s my watch .</td>
</tr>
<tr>
<td></td>
<td>5. delta airlines flight six zero two from six p.m. to miami, please?</td>
</tr>
</tbody>
</table>

Table 4: Weighted human scores for sentences. $|\text{rating} – 3|$ is weight given to each sentence’s rating. 39.02% of the generated samples were marked as real.

<table>
<thead>
<tr>
<th></th>
<th>Real</th>
<th>Fake</th>
<th>% real</th>
<th>% fake</th>
</tr>
</thead>
<tbody>
<tr>
<td>Real</td>
<td>30</td>
<td>51</td>
<td>37.04%</td>
<td>62.96%</td>
</tr>
<tr>
<td>Fake</td>
<td>48</td>
<td>75</td>
<td>39.02%</td>
<td>60.98%</td>
</tr>
</tbody>
</table>

5 Conclusion

This work presents a simple and effective model for text generation based on adversarial training using sentence embeddings. It shows how the use of sentence-level embeddings allows modelling the way of expression of an author in generated text in a better way than when using word-level embeddings. A performance comparison across several metrics is made between different GAN architectures with improved training stability and attention augmented LSTM models. Finally, it discusses how the automated corpus-based evaluations correlate with human judgements. In future, this work aims to be applied for synthesizing images from text, exploring complementary architectures to projects like neural-storyteller$^3$ where skip-thought embeddings are already used to perform image captioning with story-style transfer.

$^3$https://github.com/ryankiros/neural-storyteller
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A Partially Rule-Based Approach to AMR Generation

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Abstract

This paper presents a new approach to generating English text from Abstract Meaning Representation (AMR). In contrast to the neural and statistical MT approaches used in other AMR generation systems, this one is largely rule-based, supplemented only by a language model and simple statistical linearization models, allowing for more control over the output. We also address the difficulties of automatically evaluating AMR generation systems and the problems with BLEU for this task. We compare automatic metrics to human evaluations and show that while METEOR and TER arguably reflect human judgments better than BLEU, further research into suitable evaluation metrics is needed.

1 Introduction

Abstract Meaning Representation, or AMR, is a representation of a sentence as a rooted, labeled graph. It provides a representation of the sentence’s semantics while abstracting away from morphosyntactic details such as tense, number, word order, and part of speech (Banarescu et al., 2013).

Because of these abstractions, it can be very difficult to generate from an AMR back to a fluent English sentence which preserves the original meaning. It is also difficult to accurately evaluate the quality of generation results, since there is typically only one reference sentence available to compare results to, but one of the basic principles of AMR is that the same AMR is used to represent many possible sentences; for example, “he described her as a genius”, “his description of her: genius”, and “she was a genius, according to his description” would all correspond to the same AMR (Banarescu et al., 2013).

The following represents an AMR graph for the sentence “A key European arms control treaty must be maintained.”:

(o / obligate-01
 :ARG2 (m / maintain-01
 :ARG1 (t / treaty
 :ARG0-of (c2 / control-01
 :ARG1 (a / arms))
 :ARG1-of (k / key-02)
 :mod (c / continent
 :wiki "Europe"
 :name (n / name
 :opl "Europe"))))

This example demonstrates several of the challenges faced by an AMR generation system. These include properly addressing constructions that do not correspond closely to the words in the reference, such as the use of the frame obligate-01 to express ‘must’ and the specific construction used for named entities such as ‘Europe’, as well as word order and the passive construction ‘be maintained’. In fact, this system successfully addresses some but not all of these challenges, producing the output “Must maintain Europe key arms control treaty.”

While most previous work in AMR generation has used statistical and neural techniques, the current work approaches the task with a combination of rules and statistical methods; the rules are intended to constrain possibilities, particularly the possible realizations of concepts and which information from the AMR is expressed. This allows for greater control over the output; even if the overall results do not score as well, on average, as those of other approaches, this approach has the potential to minimize the chances of significant adequacy errors such as omission of key information or addition of information not contained in the AMR, which are possible in machine-learning-based systems. Another advantage of a partially rule-based approach is that it can work without large amounts
of AMR-annotated data; it could thus be adapted to a new language or an altered AMR scheme in situations where there is insufficient data for a machine-learning-based system to achieve satisfactory performance.

2 Related Work

2.1 AMR Generation Systems

Flanigan et al. (2016) introduced the first AMR generator (JAMR), which transforms an AMR graph into a tree before using a weighted tree-to-string transducer to generate the string. While most of its rules are automatically extracted, these are supplemented with handwritten rules for some phenomena including dates, conjunctions, and some special concepts. An ablation experiment showed that these handwritten rules contributed significantly to the results.

AMR generation was included as a shared task at SemEval-2017 (May and Priyadarshi, 2017). The winner of the task, as determined by human judgments, was the RIGOTRIO system (Gruzitis et al., 2017), which uses handcrafted rules to convert AMRs to Abstract Syntax Trees, which are then realized as strings using existing Grammatical Framework resources. However, this approach has limited coverage, and is only used for about 12% of sentences, while the system defaults to using JAMR for other sentences.

Other submissions to the shared task included FORGe, which uses graph transducers (Mille et al., 2017); Sheffield, which treats AMR generation as an inverse of transition-based parsing, transforming the AMR graph into a syntactic dependency tree before realizing it as a sentence (Lampouras and Vlachos, 2017); and ISI, which uses phrase-based machine translation (PBMT) methods (Pourdamghani et al., 2016).

Beyond the shared task, other work in AMR generation has approached the task as a Traveling Salesman Problem (Song et al., 2016), with synchronous node replacement grammar (Song et al., 2017), and using a transition-based approach to transform the AMR into a syntactic dependency tree (Schick, 2017). Castro Ferreira et al. (2017) compare the effect of different types of preprocessing on the performance of AMR generation systems based on PBMT and NMT.

The best results to date have been obtained with neural methods, which excel when the small amount of manually-annotated training data is augmented with millions of unlabeled sentences which have been automatically parsed. Konstas et al. (2017) first used this approach to train a sequence-to-sequence model, and and Song et al. (2018) later adapted it to a graph-to-sequence model.

2.2 Evaluation

Most previous work in AMR generation has reported results exclusively using BLEU scores (Papineni et al., 2002), with the original sentence as the only reference. A notable exception is the five systems included in the SemEval-2017 shared task, which were additionally compared by human judgments. The human evaluations were shown not to correlate well with BLEU scores, raising questions about the suitability of the metric for this task (May and Priyadarshi, 2017). In particular, BLEU as used for AMR generation is intuitively inappropriate because it strictly measures similarity to one reference sentence, while by design, a single AMR can correspond to many different English sentences. Thus, BLEU is in practice more of a measure of how closely a system can replicate the exact wording used in the original sentence than of how adequately and fluently it expresses the meaning of the AMR.

Ideally, evaluation of AMR generation would be performed using human judgments or task-based evaluations; unfortunately, however, it is sometimes necessary to rely on the practicality of automatic metrics. We thus follow Castro Ferreira et al. (2017) in reporting two additional automatic metrics alongside BLEU, which may provide slightly more insight into system performance. The first is METEOR, which has been shown to correlate more strongly with human judgments of machine translation quality than BLEU does (Lavie and Agarwal, 2007; Denkowski and Lavie, 2014). It is a particularly appealing alternative to BLEU for AMR generation because, instead of only giving credit to exact word matches, METEOR also allows matching based on stems, synonyms, and paraphrases. This mitigates the issues associated with having a single reference sentence in AMR, because it does not penalize systems as harshly for not correctly guessing the forms of morphological and syntactic variants that are usually not specified within the AMR. The final evaluation metric used is Translation Edit Rate (TER), which has been shown to require only one reference sentence in order to correlate as well with human judgments for ma-
chine translation as BLEU does with four refer-
cences (Snover et al., 2006). This robustness against
lack of extra references makes it, too, likely to
be better suited to the AMR generation task than
BLEU is.

These metrics were all designed for evaluation
of machine translation; it may also be useful in
the future to explore evaluating AMR generation
with metrics from other NLG-related tasks, such as
referenceless measures developed for grammatical
error correction (e.g. Napoles et al., 2016; Choshen
and Abend, 2018).

3 Methods

The system introduced in this paper uses rules to
generate realization hypotheses, which are ranked
and combined by statistical methods.

We used the LDC2015E86 version of the AMR
corpus for training the linearization models, tuning
hyperparameters, and analyzing errors.

3.1 Algorithm

The system uses an algorithm based on cube prun-
ing (Huang and Chiang, 2007). Hypotheses are
generated by recursing down the AMR graph, ign-
oring subsequent mentions of variables that have
already been processed (reentrancies) and therefore
treating the AMR as a tree. The roles :wiki and
:mode are also ignored: :wiki provides extra in-
formation about named entities that does not need
to be expressed in the English realization. The
:mode relation could be used in future versions
of the system to generate particular sentence types
such as questions and imperatives; however, this
syntactic manipulation would require more compi-
lcated rules than are used in this algorithm, and so
the relation is ignored for now.

At each node, a priority queue of scored hypothe-
eses is generated.

In particular, at each leaf node, one or more
hypotheses are created for the realization of the
node, according to the process described in 3.2,
and each of these partial hypotheses is scored by a
language model.

At each non-terminal node, priority queues are
recursively generated containing hypotheses for the
realizations of each of the node’s children, as well
as one for the node itself. Each of these is pri-
oritized by language model score. An additional
priority queue represents possible linearizations of
the current node and each of its children, scored as
discussed in 3.4. Thus, for a node with \( n \) children,
a total of \( n + 2 \) priority queues are created, simu-
lating an \( n + 2 \)-dimensional hypercube. \( k \) nodes of
the cube are then expanded and rescored by the
language model.

When the root node of the AMR is reached, a
period is added to the end and the \( k \) hypotheses are
rescored by the language model, this time treated
as a complete sentence. The realization associated
with the best-scored hypothesis is postprocessed to
capitalize the first letter, then returned.

3.2 Realization

For each node, one or more possible strings are
generated to realize the node’s associated concept
or constant and, in some cases, its relation. The
system uses specific rules for some special cases,
and more generalized rules for most nodes.

Special Cases: Special rules are used for a few
constructions. In particular, the constructions for
named entities and for people with relationships
and roles in organizations (have-rel-role-91
and have-org-role-91) are represented in
AMR with some concepts that do not typically
align with words in the text; rules prevent these and
certain other AMR-specific concepts from being
realized.

In addition, a ‘–’ representing negative polarity
is realized as ‘not’ or as one of several negative con-
tractions; numbers in ordinal or month-name con-
structions are realized accordingly, and pronouns
are realized as their possessive form in possessive
constructions and may be realized as their subject-
ive or objective forms otherwise. Finally, a handful
of concepts whose names don’t correspond to the
English strings they typically represent are mapped
to more likely English translations. These repre-
sent some conjunctions, modals, and other rela-
tionships that are associated with particular concepts
in AMR; for example, contrast-01 is realized
as ‘but’ and obligate-01 as ‘can’.

Frames: Frames not dealt with in the special
rules are likely to correspond to verbs, or occasion-
ally adjectives, adverbs, and verb-derived nouns,
and are treated as such. These are represented
in English as they appear in the concept name,
with frame numbers removed and words joined by
spaces rather than hyphens. In addition to this base

---

1See the code, available at https://github.com/
emmann/emmAHR, for full details on the realization
rules that could not be explained exhaustively due to space
constraints.
concept name, several variations are generated, corresponding to different possible verbal, adjectival, and nominal forms. Verbal and adjectival forms are created by combining optional auxiliaries with verb forms such as those ending in ‘-ing’, ‘-s’, and ‘-ed’, (with several variations depending on the base form, such as removing a final ‘e’ before adding an ending when appropriate). Nominal variants are similarly created with variations on the suffixes ‘-ment’ and ‘-tion’, and adverbs with ‘-ly’.

In many cases this generates forms that are implausible; this is not a problem because they are ranked by language model score, so forms that are unattested or very infrequent in the language model’s training corpus receive poor scores and will be pruned out at a later stage. Nevertheless, this strategy inevitably misses some valid forms, such as those of many irregular verbs; this could be addressed in the future by integrating external resources that can produce inflections and derivations of a given word.

**Non-Frame Concepts:** Remaining non-frame concepts are given a treatment similar to the frames discussed above, except that they are assumed more likely to be nouns or elements of noun phrases (such as adjectives), and are given corresponding realizations. They may also represent other parts of speech, such as adverbs; in these cases, again, the language model can rule out any implausible forms that are spuriously generated.

The plain concept name is formed by simply replacing any hyphens in the original concept name with spaces, although concepts at this stage are usually already a single word. The hypotheses created are for the plain concept name, as well as variations which append ‘the’, ‘a’, or ‘an’ before the name or plural suffix ‘s’ at its end, and a hypothesis which adds both ‘the’ and the plural ‘s’.

**Relations:** In addition to the variable realizations, a handful of relations are realized by strings attached before or after the realization of their associated variable. These are given in Table 1. Many of these represent prepositions, including the general :prep-X relation, which is realized as the given preposition.

### 3.3 Language Model

A 5-gram language model was created to score hypothesis strings. It was trained on the English Gigaword Fifth Edition Corpus (LDC2011T07) using KenLM (Heafield, 2011; Heafield et al., 2013). For time and space efficiency, the model was pruned: singleton 2-grams were removed, as well as 3-, 4-, and 5-grams that appear 3 or fewer times.

<table>
<thead>
<tr>
<th>Concept</th>
<th>Realization</th>
<th>Type</th>
</tr>
</thead>
<tbody>
<tr>
<td>:prep-X</td>
<td>X</td>
<td>prefix (+space)</td>
</tr>
<tr>
<td>:accompanyer</td>
<td>with</td>
<td>prefix (+space)</td>
</tr>
<tr>
<td>:destination</td>
<td>to</td>
<td>prefix (+space)</td>
</tr>
<tr>
<td>:purpose</td>
<td>to</td>
<td>prefix (+space)</td>
</tr>
<tr>
<td>:condition</td>
<td>if</td>
<td>prefix (+space)</td>
</tr>
<tr>
<td>:compared-to</td>
<td>than</td>
<td>prefix (+space)</td>
</tr>
<tr>
<td>:poss</td>
<td>‘s</td>
<td>suffix (+space)</td>
</tr>
<tr>
<td></td>
<td>of</td>
<td>prefix (+space)</td>
</tr>
<tr>
<td>:domain</td>
<td>is</td>
<td>suffix (+space)</td>
</tr>
<tr>
<td>:location</td>
<td>in</td>
<td>prefix (+space)</td>
</tr>
<tr>
<td></td>
<td>at</td>
<td>prefix (+space)</td>
</tr>
<tr>
<td></td>
<td>by</td>
<td>prefix (+space)</td>
</tr>
</tbody>
</table>

Table 1: Realizations for relations.

### 3.4 Linearization

The linearization model contains two components, the pair-order model and the coreness model. These may be used in combination, in which case the scores they assign are averaged, or either one may be used alone. There is also a simpler baseline, described below, which is used by default when both models are disabled. Each of the models is trained on the alignments provided with the training data.

When non-baseline linearization scoring is used, all permutations of a node and its children are scored, assuming the node has no more than 5 children (producing at most $6! = 720$ combinations to score). This covers the vast majority of cases, but on the occasion that a node has more children, only the three orderings generated by the baseline linearization are considered. Unlike baseline linearization, these three candidates are still scored by the model(s) rather than being assigned identical scores. This limiting of possibilities serves a practical purpose in limiting the maximum number of permutations that must be calculated and scored; however, it is likely that it does not hurt performance, since nodes with a large number of children often represent lists, where preserving the original order of the children is likely to match the original sentence.

**Pair-Order Model:** The pair-order model is designed to capture the intuition that particular relations are likely to be realized to either the left or the right of other particular relations, or of their parent, represented here as the special relation \textsc{root}. For example, :\textsc{arg0}, which usually represents an agent, occurs before its \textsc{root} 77% of the time, while :\textsc{arg1}, usually a patient, precedes its \textsc{root}...
in only 25% of cases.

The model is trained by counting ordered pairs of relations when it can be determined that one is realized before the other based on the provided alignments. Counts are then normalized into the probability of a particular order for any pair of relations, using add-1 smoothing to avoid assigning probabilities of 0 to unattested orderings. The model scores a hypothesized ordering by combining the scores of each pair of relations in the ordering.

Focusing only on relations allows for a small, simple model and avoids data sparseness; however, there are situations where a lexicalized version of this model may provide useful information that is lost here, such as the fact that some frames are more likely than others to have :ARG1 realized to their left. A version of this model that could capture such information is left for future work.

Coreness Model: The second component to the linearization model is the coreness model, which attempts to capture the intuition that in addition to the left-to-right ordering represented by the pair-order model, some relations are more ‘core’ and are more likely to be realized closer to their parent than others. For example, whether :ARG1 appears before or after its parent, it is likely to be close to it, while a relation like :time or :purpose will usually appear farther away, either very early or very late in a sentence. Thus, the model stores a single score for each relation representing its average absolute distance from the root, as a proportional distance relative to other children. A hypothesis’s score is penalized based on the difference between each child’s observed and expected distance from the root.

Baseline Linearization: When neither of the statistical linearization models are available, only three orderings are considered. The children are kept in the order they appear in the penman representation of the AMR, with the parent inserted either initially, penultimately, or finally. The penultimate option is considered because this is typically the appropriate place when the parent is a conjunction like ‘and’. This is particularly useful in cases where the baseline is used as a default due to a node having a large number of children, since these often represent a long list of conjuncts. These hypotheses are given equal scores, meaning that all will be combined with realization hypotheses to create new hypotheses, and only the language model determines which are best.

4 Experiments

4.1 System Evaluation

First, several variations on this system with different hyperparameters were tested on the 1368 AMRs in the dev data. As discussed in 3.4, the linearization model contains two separate components, each of which is optional, resulting in four different linearization configurations. The pruning parameter $k$ was tested at values of 5, 10, and 100. In total, 12 different versions of the system were tested on dev data. Based on these results, an optimal version of the system was chosen, which was then evaluated quantitatively on test data with automatic metrics. This system’s output on dev data was also evaluated qualitatively by reviewing a small sample of its sentences, and quantitatively by comparing the number of tokens and frequency of parts of speech to those of the references.

4.2 Evaluation Metrics for AMR Generation

As discussed in 2.2, AMR Generation is usually evaluated only with BLEU, but one shared task obtained human judgments of five systems which were shown not to correlate well with BLEU scores (May and Priyadarshi, 2017). We tested the system outputs from this task with BLEU\(^2\) as well as METEOR\(^3\) and TER\(^4\) to determine whether these metrics would correspond more closely to human judgments; results are presented and discussed in 5.5.

5 Results

5.1 Intra-System Variation

Table 2 shows the performance of each variation of the system on the dev data.

Systems using the pair-order model for linearization always perform better than their counterparts without it. While the coreness model’s results are more mixed, it seems overall to hurt more than it helps. Increasing the stack size $k$ always improved BLEU and METEOR scores, and doesn’t hurt TER scores except in the +P+C condition, where the

\(^2\)Using the multi-bleu-detok.perl script provided with the MOSES toolkit (Koehn et al., 2007)
\(^3\)Using version 1.5, downloaded from http://www.cs.cmu.edu/~alavie/METEOR/
\(^4\)Downloaded from http://www.cs.umd.edu/~snover/tercom/
Table 2: System performance on dev data with various hyperparameter combinations. ‘+P’ is used to designate systems using the pair-order model and ‘-P’ to designate those without it; similarly, ‘+C’ and ‘-C’ are used to designate whether or not the coreness model was used. In the ‘-P-C’ configuration, baseline linearization is used. The best score for each metric is shown in bold. The shaded row represents the configuration of the system selected for further evaluation.

<table>
<thead>
<tr>
<th>Model</th>
<th>BLEU</th>
<th>METEOR</th>
<th>TER</th>
</tr>
</thead>
<tbody>
<tr>
<td>+P+C,k=5</td>
<td>7.51</td>
<td>27.9</td>
<td>76.1</td>
</tr>
<tr>
<td>+P+C,k=10</td>
<td>8.12</td>
<td>28.1</td>
<td><strong>75.9</strong></td>
</tr>
<tr>
<td>+P+C,k=100</td>
<td>8.8</td>
<td>28.3</td>
<td>76.1</td>
</tr>
<tr>
<td>+P-C,k=5</td>
<td>7.76</td>
<td>28.0</td>
<td>76.0</td>
</tr>
<tr>
<td>+P-C,k=10</td>
<td>8.13</td>
<td>28.1</td>
<td>76.0</td>
</tr>
<tr>
<td>+P-C,k=100</td>
<td><strong>9.03</strong></td>
<td>28.4</td>
<td>76.0</td>
</tr>
<tr>
<td>-P+C,k=5</td>
<td>6.18</td>
<td>26.9</td>
<td>81.4</td>
</tr>
<tr>
<td>-P+C,k=10</td>
<td>6.48</td>
<td>27.0</td>
<td>81.4</td>
</tr>
<tr>
<td>-P+C,k=100</td>
<td>7.70</td>
<td>27.7</td>
<td>80.4</td>
</tr>
<tr>
<td>-P-C,k=5</td>
<td>5.99</td>
<td>26.6</td>
<td>79.1</td>
</tr>
<tr>
<td>-P-C,k=10</td>
<td>6.87</td>
<td>27.0</td>
<td>78.3</td>
</tr>
<tr>
<td>-P-C,k=100</td>
<td>7.71</td>
<td>27.4</td>
<td>77.9</td>
</tr>
</tbody>
</table>

Because it performs best according to BLEU and METEOR, and achieves barely short of its best TER score, the system using only the pair-order linearization model and $k=100$ was selected as the best version. These are the hyperparameter values that are used for the following quantitative and qualitative analyses.

### 5.2 Comparison to Other Systems

Table 3 summarizes the best BLEU and (when available) METEOR and TER scores reported for this system and various others. This table excludes the results of participants in the SemEval shared task, which were evaluated on a separate test set and whose scores are given below in Table 6.

The best BLEU score of this system is substantially lower than that of others. However, the METEOR score does appear a little more competitive, outperforming Castro Ferreira et al.’s NMT system by a point, and it is currently unknown how most other system’s METEOR and TER scores compare with this one.

While automatic metrics are not ideal for comparing systems, they do seem to indicate that this system is not currently competitive with state-of-the-art systems. While it could certainly be improved in many ways given more time, this may indicate that rules augmented only with limited statistical models are not as well-suited to this task as other approaches, particularly the state-of-the-art neural models (Konstas et al., 2017; Song et al., 2018). Still, as discussed in Section 1, the partially rule-based approach has potential to be useful in situations where greater control over output is necessary or where training data is particularly limited.

### 5.3 Error Analysis

The system’s output on a random sample of 25 sentences from the dev data was analyzed. Sentences were subjectively coded for quality into four categories: ‘excellent’, ‘good’, ‘fair’, and ‘poor’. Descriptions, counts, and examples of each of these categories are given in Table 4. Crucially, unlike the automatic evaluations, this evaluation was based on how fluently and adequately the system output expresses the meaning of the AMR, not on how closely it matches the reference string. For example, a reference-based automatic metric would count ‘Kinkel’ in the third example as incorrect, because it does not match the misspelling ‘Kinkerl’ in the reference set; this evaluation, based on the AMR itself, recognizes the system’s output of ‘Kinkel’ as correct.

Most sentences, especially those classified as fair and poor, include both linearization and realization errors. An example of a linearization error is in the fourth example in the table, where the linearization in the system output incorrectly implies that jcboy was the speaker, rather than the addressee. However, linearization that differs from the reference is not always considered incorrect: in the second example, the system places the ‘even if...’ clause before the main clause of the sentence, which is a valid ordering even though it differs from that of the reference.

Further insight into the limitations of this system can be gained by looking at the lengths of output sentences and the frequency of different parts of speech. To analyze these discrepancies, the system output and references were both automatically tagged, using the NLTK tagger (Bird et al., 2009) and part of speech tags from the Penn Treebank (Santorini, 1990). Table 5 summarizes the counts of each part-of-speech tag found in system output and in the references.

One striking finding is that the system output has noticeably fewer tokens in total than the reference. Part of this is due to punctuation: while the system output contains more periods than the references due to adding a period to every sentence, it has no rules to output other punctuation such as commas, colons, parentheses, and quotation marks.
Further investigation into adding punctuation might not only improve scores from automatic metrics, but also increase readability of more complicated sentences if done correctly.

Beyond punctuation, the system under-produces many types of function words, particularly determiners (DT), existentials (EX), prepositions and subordinating conjunctions (IN), possessives (POS, PRPS), the word ‘to’ (TO), and all types of WH-words (WDT, WP, WPS, WRB). In the nominal domain, the system realizes too many nouns as singular rather than plural for both common and proper nouns, which is probably due to a combination of some irregular plurals being missed by the realization rules, and the language model perhaps preferring singular forms more than is appropriate for these reference sentences. In the verbal domain, the system outputs more ‘-ing’ forms (VBG) than are found in the references, and under-produces all other verb types, especially base forms (VB). This last is somewhat surprising, since the base form should always be accurately generated as an option by the realization rules; it is likely that its probability is underestimated by the language model, perhaps because the infinitival ‘to’ is not usually generated in hypotheses.

5.4 Future Work

The prevalence of linearization errors in the output sampled indicates that improving the linearization models would substantially improve system performance. As mentioned in 3.4, a lexicalized version of the linearization model would likely improve performance, especially if the data sparseness is mitigated by using augmented training data, similar to the approaches used by Konstas et al. (2017) and Song et al. (2018). Additionally, it may help to base the linearization models on the alignments of Szubert et al. (2018), which provide more complete coverage of the AMRs.

Realization errors mostly arise from the fact that the restrictive realization rules do not allow for all valid possibilities; for example, there is no rule to allow relative clauses such as the ‘the ones who...’ construction used in the reference of the second example sentence. The analysis of part of speech and overall token frequency also shows the need for more realization options involving function words, such as prepositions, which are currently not produced in many situations where they should be. These problems can be improved by the addition of more realization options to the handcrafted rules, although this is a time-consuming process.

5.5 Evaluation Metrics

In the SemEval shared task, only BLEU scores were originally reported alongside measures of human rankings. To explore how well each of the three automatic metrics used in this paper correlate with human judgments, we tested the output of all the systems that participated in the task with each of these metrics. Table 6 shows these new results, alongside the results of the human rankings.

All four measures agree that RIGOTRIO and CMU are the best two systems out of the task participants, but METEOR and TER both agree with humans in rating RIGOTRIO highest, while CMU obtains a higher BLEU score. This provides some evidence for the claim that METEOR and TER may be better suited than BLEU to evaluating AMR generation, especially when it comes to distinguishing among stronger systems. However, none of the metrics fully match the ranking given by humans—in particular, while humans considered FORGe the third-best system, all of the automatic metrics rank it lower. Thus, while these alternatives may be an improvement over BLEU, more research is necessary to determine a more accurate way to auto-

---

Table 3: Comparison of test scores achieved by this system and others.

<table>
<thead>
<tr>
<th>System</th>
<th>LDC2014T12 BLEU</th>
<th>LDC2015E86 BLEU</th>
<th>LDC2016E25 BLEU</th>
</tr>
</thead>
<tbody>
<tr>
<td>Pourdamghani (2016)</td>
<td>26.9</td>
<td>-</td>
<td>-</td>
</tr>
<tr>
<td>Schick (2017)</td>
<td>28.9</td>
<td>-</td>
<td>-</td>
</tr>
<tr>
<td>Flanigan (2016)</td>
<td>22.1</td>
<td>-</td>
<td>-</td>
</tr>
<tr>
<td>Song (2016)</td>
<td>22.4</td>
<td>-</td>
<td>-</td>
</tr>
<tr>
<td>Song (2017)</td>
<td>25.2</td>
<td>-</td>
<td>-</td>
</tr>
<tr>
<td>Castro Ferreira (2017)</td>
<td>33.8</td>
<td>37.5</td>
<td>46.9</td>
</tr>
<tr>
<td>Konstas (2017)</td>
<td>31.1</td>
<td>37.5</td>
<td>46.9</td>
</tr>
<tr>
<td>Song (2018)</td>
<td>33.0</td>
<td>-</td>
<td>-</td>
</tr>
<tr>
<td>This System</td>
<td>8.7</td>
<td>28.2</td>
<td>76.0</td>
</tr>
</tbody>
</table>

Footnotes:
-\(^5\) Castro Ferreira et al. report results for 8 versions of each of system (NMT and PBMT), using different combinations of preprocessing steps. For each system, we select the version that performs best according to two of the three metrics to represent here.
-\(^6\) Data provided to me by Konstas in personal communication; scores determined by the same tests used for my system’s data.
Table 4: Description and example for each of the four quality categories.

| Excellent (1): Human-like quality; fluently presents all meaning |
| (t / tool |
| mod (m / machine) |
| purpose (s / set-up-03 |
| :ARG1 (f2 / facility) |
| :location-of (f / fabricate-01))) |
| REF: Machine Tools for setting up a fabrication facility. |
| SYS: Machine tools to set up fabrication facility. |

| Good (5): Adequately and intelligibly expresses all or nearly all meaning with minor disfluencies |
| (m / multi-sentence |
| :snt1 (s / suffer-01 |
| :ARG0 (p / person |
| :mod (o2 / ordinary))) |
| :snt2 (f / fish |
| :ARG1-of (s2 / salt-01) |
| :mod (s3 / still) |
| :domain f2 |
| :concession (e / even-if |
| :opl1 (t / turn-01 |
| :ARG1 (b / body |
| :poss (f2 / fish |
| :ARG1-of (s4 / salt-01))) |
| :direction (o3 / over)))) |
| REF: the ones who are suffering are the ordinary people: even if the body of a salted fish is turned over, it is still a salted fish ... |
| SYS: An ordinary person suffer. salted fish is still salted fish even if the body turns over. |

| Fair (6): Meaning is partially intelligible |
| (s / say-01 |
| :ARG0 (p / person :wiki "Klaus_Kinkel" |
| :name (n2 / name :op1 "Kinkel")) |
| :ARG1 (a / and |
| :opl1 (c / correct-02 |
| :ARG1 (t / thing |
| :ARG1-of (d / decide-01 |
| :ARG0 (m / military |
| :wiki "NATO" |
| :name (n3 / name |
| :op1 "NATO"))) |
| :op2 (n4 / need-01 |
| :ARG1 t)) |
| REF: Kinkerl said that NATO’s decision was “correct and necessary.” |
| SYS: Said Kinkel NATO decided things correctly and needs. |

| Poor (13): Meaning is barely or not at all intelligible |
| (s / say-01 |
| :ARG1 (c / correct-02 |
| :ARG1 (y / you) |
| :ARG2 (p / person :wiki - |
| :name (n / name :op1 "jcboy")) |
| REF: @jcboy, You are correct. |
| SYS: You correct jcboy said. |

Table 5: Comparison of counts of part-of-speech tags in system output vs. references.

<table>
<thead>
<tr>
<th>System</th>
<th>Trueskill</th>
<th>BLEU</th>
<th>METEOR</th>
<th>TER</th>
</tr>
</thead>
<tbody>
<tr>
<td>RIGOTRIO</td>
<td>1.03</td>
<td>18.6</td>
<td>32.3</td>
<td>80.1</td>
</tr>
<tr>
<td>CMU</td>
<td>0.819</td>
<td>19.0</td>
<td>31.4</td>
<td>82.4</td>
</tr>
<tr>
<td>FORGe</td>
<td>0.458</td>
<td>2.8</td>
<td>20.0</td>
<td>92.0</td>
</tr>
<tr>
<td>ISI</td>
<td>-1.172</td>
<td>10.9</td>
<td>28.9</td>
<td>98.7</td>
</tr>
<tr>
<td>Sheffield</td>
<td>-2.132</td>
<td>1.2</td>
<td>20.0</td>
<td>87.5</td>
</tr>
<tr>
<td>This System</td>
<td>-</td>
<td>6.7</td>
<td>28.2</td>
<td>79.4</td>
</tr>
</tbody>
</table>

Table 6: Comparison of evaluation metrics to Trueskill (measure of human rankings) for shared task data and systems. This system’s performance on the same data according to automatic metrics is provided for comparison.

6 Conclusion

Given that the relatively small amount of available data and the difficulty of the task have made it difficult for statistical and neural approaches to achieve truly satisfactory results in AMR generation, we hypothesized that a partially rule-based system, combined with some simple statistical methods, might be able to effectively harness human linguistic knowledge to achieve comparable results.

Judging by automatic metrics, this method does not seem able to compete well with state-of-the-art systems—although this system’s scores could doubtless be improved somewhat with further development, especially if the primary goal were to optimize toward metrics like BLEU by providing more realization candidates that might better match the reference sentence’s n-grams. However, we also argue that BLEU scores are a poor metric for

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These numbers differ slightly from the previously reported BLEU scores, presumably due to differences in BLEU configuration, but the ranking of systems is the same.
the AMR generation task. While METEOR and TER appear to do slightly better than BLEU at least at distinguishing among better-performing systems, none of these metrics fully reflect human rankings, making it difficult to fully determine how this system compares to others without human evaluation.

As research moves forward in AMR generation, it is essential to ensure that we are truly moving in a direction that will help us generate English realizations that both adequately and fluently express the meaning represented in an AMR. It is clear that the automatic metrics that have been used for this task fail to achieve these goals. More research is necessary to develop new metrics that are better suited to this task.

Acknowledgments

I would like to thank Nathan Schneider for his guidance, Ioannis Konstas and Jonathan May for sharing their data, and the anonymous reviewers for their comments and suggestions.

References


Republic. Association for Computational Linguistics.


Computational Investigations of Pragmatic Effects in Natural Language

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Abstract
Semantics and pragmatics are two complementary and intertwined aspects of meaning in language. The former is concerned with the literal (context-free) meaning of words and sentences, the latter focuses on the intended meaning, one that is context-dependent. While NLP research has focused in the past mostly on semantics, the goal of this thesis is to develop computational models that leverage this pragmatic knowledge in language that is crucial to performing many NLP tasks correctly. In this proposal, we begin by reviewing the current progress in this thesis, namely, on the tasks of definiteness prediction and adverbial presupposition triggering. Then we discuss the proposed research for the remainder of the thesis which builds on this progress towards the goal of building better and more pragmatically-aware natural language generation and understanding systems.

1 Introduction
The past several years have seen growing trends in relying on intelligent systems to carry out day-to-day tasks. From interactions with your virtual personal assistant to carrying out a simple conversation with a chatbot to asking for directives or getting restaurant recommendations, the ability of said intelligent systems to properly understand its user is becoming increasingly important and crucial to their effective functioning.

For a proper understanding, these systems ought to rely on two different but complementary aspects of language: semantics and pragmatics. At a very high level, semantics is concerned with the literal meaning of words and sentences that is context-free, while pragmatics is concerned with the intended meaning, one that is context-dependent.

On the frontier of semantics, the past few years have witnessed a remarkable progress in language modeling and other tasks. Research on neural vector representations (word embeddings) has particularly exploded starting with the Word2vec model (Mikolov et al., 2013) and GloVe (Pennington et al., 2014). These neural models learn high-quality vector representations of words which are empirically shown to have state-of-the-art performance on semantic similarity tasks. Many variations have been proposed to account for varying textual structures (words vs sentences vs paragraphs vs documents), multiple languages (Luong et al., 2015), varying topics (Liu et al., 2015), etc.

Pragmatic reasoning, on the other hand, is arguably one of the major milestones on the road to general AI simply because for a machine to fully understand individuals, it has to understand the nuances and subtleties of the language that is being conveyed and which often goes beyond the literal meaning of the utterances being made. And while correctly performing pragmatic reasoning is at the core of many NLP tasks such as information extraction, automatic summarization, and machine translation, there has been not much focus on it in NLP research – at least not as much as its semantic counterpart.

The goal of this thesis is to develop computational models where pragmatics is a first-class citizen both in terms of natural language understanding and generation. We have already made progress toward this goal by (1) developing a neural model for definiteness prediction the task of determining whether a noun phrase should be definite or indefinite in contrast to prior work relying on heavily-engineered linguistic features and (2) by introducing the new task of adverbial presupposition triggering detection which focuses on detecting contexts where adverbs (e.g. “again”) trigger presuppositions (“John came again” presupposes “he came before”).

Moving forward, we propose to examine the
role of pragmatics, particularly presuppositions, in language understanding and generation. We will develop computational models and corpora that incorporate this understanding to improve: (1) summarization systems e.g. in a text rewriting step to learn how to appropriately allocate adverbs in generated sentences to make them more coherent, and (2) reading comprehension systems where pragmatic effects are crucial for the proper understanding of texts. By the end, this thesis would present the first study on presuppositional effects in language to enable pragmatically-empowered natural language understanding and generation systems.

In what follows, we will give a summary of our research effort thus far, briefly discussing the two tasks mentioned above and that can be seen as testbeds for pragmatic reasoning followed by exciting current and future research avenues for further exploration.

2 Definiteness Prediction

In (Kabbara et al., 2016), we focus on definiteness prediction, the task of determining whether a noun phrase should be definite or indefinite. In English, one instantiation of this task is to predict whether to use a definite article (the), indefinite article (a(n)), or no article at all. This task has applications in machine translation, and in L2 grammatical error detection and correction.

Definiteness prediction is an interesting testbed for pragmatic reasoning, because both contextual and local cues are crucial to determining the acceptability of a particular choice of article. Consider the following example: A/#the man entered the room. The/#a man turned on the TV. Factors such as discourse context, familiarity, and information status play a role in determining the choice of articles. Here, man is introduced into the discourse context by an indefinite article, then subsequently referred to by a definite article. On the other hand, non-context-dependent factors such as local syntactic and semantic restrictions may block the presence of an article. For example, demonstratives (e.g., this, that), certain quantifiers (e.g., no), and mass nouns (e.g., money) do not permit articles.

We present in this work a recurrent neural network model that employs an attention mechanism. The primary motivation for the use of an attention mechanism is to investigate whether LSTM models focus on certain parts of the sentence when making predictions, and if so, to gain more insight into what parts of the sentence affect the model’s prediction.

Our model achieves state-of-the-art performance on definiteness prediction, outperforming a previous logistic regression classifier (De Felice, 2008) that uses 10 types of hand-crafted linguistic features. Our best model achieves 96.63% accuracy on the WSJ/PTB corpus, representing a relative error reduction of 51% compared to the previous state of the art. Each of the factors we examined (initializing with pre-trained vectors, giving more context, giving POS tags, attention) contributes to the performance of the model, though in different degrees. We perform a number of analyses to understand the behavior of the models, and show in particular how the attention mechanism can be useful for interpreting the model predictions.

The most interesting contribution of this work is highlighting the suitability of LSTMs for tackling complex cases of article usage where there is no obvious local cue for prediction. We find evidence that LSTMs given an extended context window can resolve cases of article usage that seem to require reasoning about coreferent entities involving synonymy. Our results suggest that recurrent neural network models such as LSTMs are a promising approach to capturing pragmatic knowledge.

3 Adverbial Presupposition Triggering

In (Cianflone, Feng, Kabbara et al., 2018), we introduce the task of predicting adverbial presupposition triggers such as also and again. Solving such a task requires detecting recurring or similar events in the discourse context, and has applications in natural language generation tasks such as summarization and dialogue systems.

Presuppositions have been extensively studied in linguistics and philosophy of language with the earliest work dating back to (Frege, 1892). They can be viewed as assumptions or beliefs in the common ground between discourse participants when an utterance is made. The importance of presuppositions is that they underlie spoken statements and written sentences and understanding them facilitates smooth communication. We refer to expressions that indicate the presence of presuppositions as presupposition triggers. These include, among others, definite descriptions, factive verbs and certain adverbs.
Our focus in this work is on adverbial presupposition triggers such as again, also and still. These triggers indicate the recurrence, continuation, or termination of an event in the discourse context, or the presence of a similar event, and are the most commonly occurring presupposition triggers after existential triggers (Khaleel, 2010).

As a first step towards language technology systems capable of understanding and using presuppositions, we propose to investigate the detection of contexts in which these triggers can be used. This task constitutes an interesting testing ground for pragmatic reasoning, because the cues that are indicative of contexts containing recurring or similar events are complex and often span more than one sentence. Moreover, such a task has immediate practical consequences. For example, in language generation applications such as summarization and dialogue systems, adding presuppositional triggers in contextually appropriate locations can improve the readability and coherence of the generated output.

We create two datasets for the task based on the Penn Treebank corpus (Marcus et al., 1993) and the English Gigaword corpus (Graff et al., 2007), extracting contexts that include presupposition triggers as well as other similar contexts that do not, in order to form a binary classification task. In creating our datasets, we consider a set of five target adverbs: too, again, also, still, and yet. We focus on these adverbs in our investigation because these triggers are well known in the existing linguistic literature and commonly triggering presuppositions. We control for a number of potential confounding factors, such as class balance, and the syntactic governor of the triggering adverb, so that models cannot exploit these correlating factors without any actual understanding of the presuppositional properties of the context.

In addition, we investigate the potential of attention-based deep learning models for detecting adverbial triggers and introduce a new weighted pooling attention mechanism designed for predicting adverbial presupposition triggers. Our attention mechanism allows for a weighted averaging of our RNN hidden states where the weights are informed by the inputs, as opposed to a simple unweighted averaging. Our model uses a form of self-attention (Paulus et al., 2018; Vaswani et al., 2017), where the input sequence acts as both the attention mechanism’s query and key/value. Unlike other attention models, instead of simply averaging the scores to be weighted, our approach aggregates (learned) attention scores by learning a reweighting scheme of those scores through another level (dimension) of attention. Additionally, our mechanism does not introduce any new parameters when compared to our LSTM baseline, reducing its computational impact.

We compare our model using the novel attention mechanism against strong baseline classifiers, including a logistic regression model and RNN- and CNN-based deep learning models, in terms of prediction accuracy and show that it outperforms these baselines for most of the triggers on the two datasets without introducing additional parameters – achieving 82.42% accuracy on predicting the adverb “also” on the Gigaword dataset.

4 Proposed Research

The research work thus far in this thesis has explored two classes of function words, namely articles and adverbs, and investigated the suitability of deep learning models to pick up on complex contextual cues for handling pragmatic inferences in learning tasks involving these function words. The remainder of the research work to be carried in this thesis will build on this progress to explore how pragmatic effects in language such as presuppositional effects can be leveraged to improve natural language generation and understanding systems. Accordingly, the proposed research is divided among the following three main fronts:

1. Corpus construction
2. Language generation
3. Language understanding.

4.1 Corpus construction

Since our work is the first to explore presuppositional effects from a computational perspective, the research community currently lacks corpora that focus on these important pragmatic effects. For this reason, we are currently in the process of constructing a new corpus that focuses specifically on adverbial presuppositional effects in English. The goal of this corpus is to provide new resources that would push further the goal of understanding presuppositional effects in specific and research on computational pragmatics more generally. The corpus construction is motivated linguistically and will involve crowdsourcing data...
(e.g., through Amazon Mechanical Turk). Two approaches to the corpus construction are of interest:

1. One approach is to consider it from a generation problem perspective. Workers will be given sentences involving presuppositions and would be tasked with identifying the presupposition in context. For example, given the sentence “John went to the restaurant again”, the worker would optimally provide a simple explanation describing that the presupposition is that “John went to the restaurant before”. We envision the corpus to be useful in the context of language generation. Particularly, such corpus would be useful for designing learning models that can focus on contextual cues leading to pragmatic effects and accordingly generate what was presupposed in a sentence as a way of showing that it has a basic understanding of the relevant pragmatic effects in context.

2. The other approach is to consider it from the perspective of a presupposition-based entailment task which illustrates drawing conclusions from specific cues in text. Workers would be given a short passage involving a presupposition such as “John has been to the restaurant again”. They would be asked something along the line of: “Is it true that John has been to the restaurant before?” to which the correct answer would be “yes”. In this case, this would be an entailment setup where the presupposition is what leads to the correct conclusion.

It would be interesting to expand this to not just adverbs, but also to other kinds of presuppositions and possibly even implicature. A more general version of this crowdsourcing could involve, for example, asking the worker to qualify whether a certain statement seems to be true, or be suggested but not necessarily true or false.

We believe that such corpora are crucial for the development of learning models that can focus on the subtle pragmatic effects of language and will play an important role in improving language generation and language understanding systems.

4.2 Language Generation

Presuppositions are prevalent in language and they play a crucial role in shaping the conveyed meaning in a specific way. In summarization scenarios where information needs to be acquired from different parts of the text(s), this would be particularly more important.

Two pillars of effective summarization systems are (1) the ability to pinpoint key pieces of information and crucial parts of the text and (2) appropriately rewrite the original text in order to relay the information in those parts.

On the first front, we plan to investigate links between sentences that involve presuppositional effects and sentences occurring in the previous context. That is, if a sentence includes a presupposition trigger, it would be important – from a summarization point of view – to determine whether there is a specific sentence that occurred in the previous context and that plays a crucial role in how the meaning of that (presuppositional) sentence is understood. We believe investigating such links between sentences is crucial for designing better summarization systems.

On the latter front, we are interested in investigating how an understanding of presuppositional effects can lead to better summarization systems. Specifically, we will investigate how a learning model, in a text rewriting step, can make informed decisions on how to properly allocate adverbs with the goal of generating a more coherent output.

Framing the problem within a summarization setting enables to not have to rely on using the original document context but instead “manipulating” the summarized version. From that version, we can select groups of sentences and ask workers to add adverbs or to remove existing ones to create manipulated versions of summarized texts to be used for designing improved summarization systems. Indeed, oftentimes, missing one presupposition trigger such as an adverb while summarizing could lead to the presupposition context (and thus the overall meaning) not holding anymore. The goal in this case would be to design summarization systems that are also trained to fill/remove adverbs such that the final summarized version is more natural and more coherent. The idea would be for the summarization system to examine each relevant sentence in the summarized text and determine whether adding/removing a specific adverb would make the phrase more informative and so whether it should be included or not in the summarized version.
4.3 Language Understanding

Of interest to our pursuit is the task of machine/reading comprehension also referred to as text understanding or question answering (QA), where the goal is to determine if a learning system can answer basic questions about a passage in order to show some “comprehension” of the information in that passage.

Presuppositions play a crucial role in shaping the sentence meanings and extending what is explicitly conveyed by the semantics of the words making up the sentence. We believe that understanding their role and leveraging the pragmatic knowledge resulting from that role can improve text understanding systems.

The current QA datasets are not suitable however for the task of designing pragmatically-empowered QA systems. For example, among the popular datasets for this task, is the Stanford Question Answering Dataset (SQuAD) (Rajpurkar et al., 2016). SQuAD consists of questions posed by crowdworkers on a set of Wikipedia articles, with more than 100,000 question-answer pairs on 500 articles. A key feature of this dataset is that the answer to a question is always part of the context and also always appears as a continuous span of words. One simplified way to tackle the problem is then to find the start and end of the relevant span of words (in the given passage) that corresponds to the answer. Not only the answers in such datasets are explicitly present in the passage, they are typically of fact-based nature.

Our interest, on the other hand, is to design QA systems that can answer questions that tap implicit information in the text, one that is pragmatic in nature. The answers, unlike datasets like SQuAD, would not be explicitly stated in the text which makes the task more challenging. The constructed corpus that was discussed in Section 4.1 will be crucial to designing and testing such systems. A simple example would be a small passage with a sentence involving a presupposition such as “John has been to the restaurant yesterday”. One possible question would be: “Is it true that John has been to the restaurant before?” and the answer should be true. There would be challenging negative cases, e.g., “Mary has been to the restaurant again, but John wants to go to the restaurant tomorrow. The answer to the same question above would be false in that case.

Answering effectively such questions would tap into the models abilities to pick up on complex contextual cues and would be a strong hint that the model can have a basic understanding of the pragmatic effects in language.

4.4 Leveraging Pragmatic Knowledge in Multi-Task Scenarios

Correctly performing the pragmatic reasoning explored in this proposal, i.e. that which deals with presuppositions, is at the core of many NLP tasks such as discourse parsing, discourse segmentation and coherence modeling.

We believe that by training a model to learn to produce simple explanations of the presupposition effect in context (in the fashion described in Section 4.1), we could leverage the learned representations to “supplement” the learning of other NLP tasks as the ones mentioned above and for which such pragmatic knowledge is essential.

5 Conclusion

We have presented in this proposal the research progress that was accomplished thus far in this thesis, exploring two classes of function words, namely articles and adverbs, and investigating the suitability of deep learning models to pick up on complex contextual cues for handling pragmatic inferences in learning tasks involving these function words, namely, the tasks of definiteness prediction and adverbial presupposition triggering. We also discussed current and future research directions that will guide the research for the remainder of this thesis mainly in the areas of natural language generation and understanding. We believe that the research in this thesis has the potential to open up exciting research directions that are unexplored in the NLP community and that are crucial for designing improved and nuanced language generation and understanding systems that bring us closer to the vision of truly intelligent systems.

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References


Abstract

Event Detection has been one of the research areas in Text Mining that has attracted attention during this decade due to the widespread availability of social media data specifically twitter data. Twitter has become a major source for information about real-world events because of the use of hashtags and the small word limit of Twitter that ensures concise presentation of events. Previous works on event detection from tweets are either applicable to detect localized events or breaking news only or miss out on many important events. This paper presents the problems associated with event detection from tweets and a tweet-segmentation based system for event detection called SEDTWik, an extension to a previous work, that is able to detect newsworthy events occurring at different locations of the world from a wide range of categories. The main idea is to split each tweet and hash-tag into segments, extract bursty segments, cluster them, and summarize them. We evaluated our results on the well-known Events2012 corpus and achieved state-of-the-art results.

Keywords: Event detection, Twitter, Social Media, Microblogging, Tweet segmentation, Text Mining, Wikipedia, Hashtag.

1 Introduction

Microblogging, as a form of social media, is fast emerging in this decade. One of the best examples for this is Twitter which allows 280-character limit for a tweet. It is used not only to share and communicate with friends and family but also as a medium to share real-world events. An event according to the Topic Detection and Tracking (TDT) project (Allan et al., 1998), is “some unique thing that happens at some point in time”. Becker et al. (2011) defines an event as “a real-world occurrence $e$ with an associated time period $T_e$ and a time-ordered stream of Twitter messages $M_e$, of substantial volume, discussing the occurrence and published during time $T_e$. We borrow these definitions of an event in our work.

In Twitter, a user can not only publish about an event but can also propagate by retweeting the post by someone else. A user can also attach a hashtag with the tweet which can provide a significant amount of information about the event (e.g., #RIP to signify that the tweet is related to someone’s death). But some hashtags can also be used to promote ideas known as memes (Kotsakos et al., 2014). Event detection from tweets also faces other challenges like noisy data, informal writing, grammatical errors, and a large volume of data coming at very high velocity. According to Internet Live Stats\(^1\), on an average 6,000 tweets are published every second, which corresponds to nearly 500 million tweets per day. Moreover, nearly 40% of these tweets are just “pointless babbles”\(^2\) which are insignificant to the task of event detection.

To tackle the above-mentioned challenges, we present SEDTWik - a tweet segmentation-based event detection system that utilizes an external knowledge base like Wikipedia. The rest of the paper is organized as follows. Section 2 describes the working of SEDTWik in detail. Section 3 presents our experimental results. Section 4 presents some related works in event detection. We conclude in section 5 along with future work to be done.

2 SEDTWik

In this section, we present SEDTWik, an extension of a previous work by Li et al. (2012a) called Twevent. SEDTWik is an event detection

\(^1\)http://www.internetlivestats.com/twitter-statistics
framework that consists of four components: tweet segmentation, bursty segment extraction, bursty segment clustering, and event summarization. Figure 1 shows the architecture of SEDTWik. Events are detected from a time window $t$ of a fixed length during which all the tweets published are processed. In the tweet segmentation phase, all tweets coming from the Twitter stream within the current time window are segmented, and the segments along with the tweet details are indexed for use in next stages. Hashtags are given more weight as they contain more information. Based on the probability distribution of segments, retweet counts, user diversity, and user popularity, abnormally bursty segments are extracted and clustered in the next two stages. Finally, the clusters are summarized in the last step. In the rest of this section, we present all the four components in detail.

2.1 Tweet Segmentation

Tweet segmentation was introduced by Li et al. (2012b) and Li et al. (2015) for Named Entity Recognition (NER) in which they used a dynamic programming based approach to segment tweets based on a “stickiness” score of a segment. In this section, we present an alternative approach using Wikipedia Page Titles Dataset\(^3\) for segmentation of tweets and hashtags.

The task of tweet segmentation is to split a given tweet into non-overlapping meaningful segments. A segment can be unigram (a word) or multi-gram (a phrase). The reason why tweet segmentation is used is that a phrase contains much more specific information than the unigrams in it. So, a tweet segment makes the event more interpretable. For example, [vice presidential debate] is much more informative than [vice], [presidential], and [debate] separately that might be in any random order. While segmenting a tweet, we emphasize three components: tweet text, name mentions, and hashtags.

We consider tweet text as everything a user writes in a tweet except URL links, hashtags, and name mention. From tweet text, we only keep those segments that are present as a title of a Wikipedia page\(^3\). This ensures that only named entities (e.g., Barack Obama) or meaningful segments (e.g., new music) are kept from tweet text, and unnecessary words are removed that would otherwise increase noise in the event detection process.

Most Twitter users use a name mention in a tweet to mention a person by their username (e.g., @iamsrk for Shah Rukh Khan). So, we replace the username by their actual name and consider it as a segment.

The most important component in our event detection model is hashtags. Hashtags contain a lot of information in a concise form, and related tweets generally contain the same hashtag. Ozdikis et al. (2012a) used only hashtags for event detection as contrasted with Ozdikis et al. (2012b) and were able to get better results in the former. This motivated us to give more weight to hashtags in the segmentation process.

\(^3\)http://dumps.wikimedia.org/enwiki/latest/enwiki-latest-all-titles-in-ns0.gz
We use $\mathcal{H}$ as hashtag weight, so an $\mathcal{H}$ value of 2 means that all hashtags are duplicated in the segmentation process, resulting in a twice weight that would allow hashtags to become more bursty in the next stage. This also ensures that if a segment is not previously seen in the Wikipedia page titles, then its use in hashtag would still make the segment bursty. Since hashtags do not contain whitespace or punctuations, we consider the capitalization of letters to segment a hashtag. For example, #BreakingNews will be segmented as [breaking news]. Hashtags that do not contain any capitalization of letters in them would be considered as a unigram when segmenting them.

### 2.2 Bursty Segment Extraction

Since there are hundreds of thousands of unique segments within a day, clustering all of them for detecting events would be a computationally expensive task. So, once the tweets are segmented, we find out abnormally bursty segments that might be related to an event and discard the remaining ones.

Let $N_t$ denote the number of tweets within the current time window $t$ and $f_{s,t}$ be the number of tweets containing segment $s$ in $t$. The probability of observing $s$ with a frequency $f_{s,t}$ can be considered as a Binomial distribution $B(N_t, p_s)$ where $p_s$ is the expected probability of observing segment $s$ in any random time window. Since $N_t$ is very large in case of tweets, this probability distribution can be approximated to a Normal distribution with parameters $E[s|t] = N_t p_s$ and $\sigma^2[s|t] = \sqrt{N_t p_s (1 - p_s)}$.

If a segment has $f_{s,t} \geq E[s|t]$, it will be called a bursty segment, while a segment with $f_{s,t} < E[s|t]$ will not be considered bursty and will be discarded. We use a formula for the bursty probability $P_b(s,t)$ for segment $s$ in time window $t$ defined by Li et al. (2012a) as given in (1) that transfers the frequency of a bursty segment to the range $[0,1]$.

$$P_b(s,t) = S\left(10 \frac{f_{s,t} - (E[s|t] + \sigma^2[s|t])}{\sigma^2[s|t]}\right) \quad (1)$$

where $S(\bullet)$ is the sigmoid function, and since sigmoid function smooths well in the range $[-10,10]$, the constant 10 is introduced.

Instead of depending entirely on tweet frequency, to incorporate user diversity, user frequency $u_{s,t}$ is also used which denotes the number of distinct users using segment $s$ in time window $t$. A retweet is a copy of a tweet created by another user. According to Boyd et al. (2010), a retweet is “a conversational practice and can negotiate authorship, attribution, and communicative fidelity”. We find that a tweet retweeted by many users might be related to an important event and can be used to provide more weight to segments in retweets. We define $\text{segment retweet count}$ of a segment $s$ in $t$ as $src_{s,t}$ which is the sum of retweet counts of all tweets containing $s$ in $t$. A tweet by someone who has millions of followers (e.g., a celebrity or a news page) might also be more important as compared with someone who has very few followers. Giving more weight to such tweets will ensure that spam or self-promoting tweets are filtered out and do not harm the accuracy of the event detection process. So, we define $\text{segment follower count}$ of a segment $s$ in $t$ as $sfc_{s,t}$ which is the sum of follower count of all users using this segment in $t$. Combining all the above, the formula for bursty weight $w_b(s,t)$ for segment $s$ in $t$ is defined in (2).

$$w_b(s,t) = P_b(s,t) \log(u_{s,t})x$$

$$\log(src_{s,t})\log(\log(sfc_{s,t})) \quad (2)$$

Among all the segments, top $K$ segments are selected as bursty segments based on their bursty weight. A small value of $K$ would result in a very low recall of events detected, and a large value of $K$ may bring in more noise leading to higher computational cost. Therefore, an optimal value of $K$ is kept to be $\sqrt{N_t}$.

### 2.3 Bursty Segment Clustering

In this section, we cluster bursty segments and filter non-event clusters using the approach by Li et al. (2012a).

Since the topics in tweets are fast changing and extremely dynamic, the similarity of two segments is calculated from their temporal frequency and the contents of the tweets that contain the segment. Each time window is evenly split into $M$ subwindows $t = < t_1, t_2, ..., t_M >$. Let $f_{t}(s,m)$ be the tweet frequency of segment $s$ in the subwindow $t_m$ and $T_{1}(s,m)$ be the concatenation of all the tweets in the subwindow $t_m$ that contain segment $s$. The similarity $sim_{1}(s_a,s_b)$ between segments $s_a$ and $s_b$ in time window $t$ is calculated...
based on formula (3).

\[ \text{sim}(s_a, s_b) = \sum_{m=1}^{M} w_t(s_a, m)w_t(s_b, m) \times \text{sim}(T_t(s_a, m), T_t(s_b, m)) \]  

(3)

where \( w_t(s, m) \) is the fraction of frequency of segment \( s \) in the subwindow \( t_m \) as mentioned in (4) and \( \text{sim}(T_1, T_2) \) is the tf-idf similarity of the set of tweets \( T_1 \) and \( T_2 \).

\[ w_t(s, m) = \frac{f_t(s, m)}{f_t(s, t)} \]  

(4)

Using the similarity measure given in (3), all the bursty segments are clustered using a variation of Jarvis-Patrick algorithm (Jarvis and Patrick, 1973). In this, all segments are considered as nodes and initially, all nodes are disconnected. An edge is added between segments \( s_a \) and \( s_b \) if \( k \)-Nearest neighbors of \( s_a \) contains \( s_b \) and vice versa. After adding all possible edges, all the connected components of the graph are considered as candidate event clusters. Those segments that do not have any edges are discarded from further processing.

After clustering the bursty segments, we found that some clusters were not related to any event. For example, one of the candidate event clusters detected from tweets of Sunday, October 14, 2012 had segments like [sunday dinner], [sunday night], [every sunday], [sunday funday], and [next sunday]. This kind of events have segments that are bursty on specific days of the week. Thus, some filtering has to be done to eliminate these events. So, use of external knowledge base like Wikipedia is made.

The newsworthiness \( \mu(s) \) of a segment \( s \), is defined as given in (5) which ensures that if a sub-phrase of a segment is an important phrase then the segment is also considered newsworthy.

\[ \mu(s) = \begin{cases} e^{Q(s)} & \text{s is a word} \\ \max_{l \in s} e^{Q(l)} - 1 & \text{otherwise} \end{cases} \]  

(5)

where \( l \) is any sub-phrase of segment \( s \) and \( Q(l) \) is the probability of \( l \) appearing as anchor text in Wikipedia articles containing \( l \).

The newsworthiness \( \mu(e) \) of an event cluster \( e \), is defined in (6) that considers the newsworthiness of its constituent segments and the weight of edges of the event cluster in the form of segment similarity.

\[ \mu(e) = \sum_{s \in e} \mu(s) \sum_{g \in E_e} \text{sim}(g) \frac{|g_e|}{|e_s|} \]  

(6)

where \( e_s \) is the set of segments associated with event \( e \), \( E_e \) is the set of edges between segments of the event \( e \), and \( \text{sim}(g) \) is the similarity between nodes of the edge \( g \) which is calculated from (3).

Candidate events that are not likely to be realistic events are observed to have very small newsworthiness as compared to real events. So, if an event \( e \) satisfies the condition \( \frac{\mu_{\text{max}}(e)}{\mu(e)} < \mathcal{T} \) then only it is kept as a realistic event otherwise discarded. Here \( \mu_{\text{max}} \) is the highest newsworthiness among all candidate event clusters and \( \mathcal{T} \) is a threshold.

2.4 Event Summarization

A list of segments associated with an event cluster might not provide all the information related to an event. So, we used the LexRank algorithm (Erkan and Radev, 2004) to summarize the event clusters obtained in the previous step. The LexRank algorithm takes as input multiple documents and provides a summary of it by combining the top-ranking sentences. To summarize an event, we use all the tweets in current time window \( t \) that contain the segments in the event cluster obtained from the segment index created in the tweet segmentation phase and apply the algorithm to provide a summary of the event.

3 Experimental Results

In this section, we will mention the dataset and evaluation metrics we used, the statistics about tweet segmentation, and our results. Our model outperforms Twevent (Li et al., 2012a) with better precision, a greater number of events, and less duplicate events.

3.1 Dataset and Experimental Setting

The Wikipedia page titles dataset used in subsection 2.1 was a dump from March 2018 which contains 8,007,358 page titles. We used the Wikipedia keyphraseness values \( Q(s) \)\(^4\) used by Li et al. (2012a) which was based on a dump released on Jan 30, 2010, and contains 4,342,732 distinct entities that appeared as anchor text.

\(^4\)https://www.ntu.edu.sg/home/axsun/datasets.html
McMinn et al. (2013) created a Twitter corpus called Events2012 containing tweets from Oct 10 - Nov 7, 2012. They removed tweets containing more than 3 hashtags, 3 name mentions, or 2 URLs as they might be spam (Benevenuto et al., 2010). After all this filtering, the corpus contains over 120 million tweets. It also contains a list of 506 events detected in the corpus distributed among 8 categories. We used this corpus to estimate the segment probabilities $p_s$ used in subsection 2.2 and to evaluate the performance of our model. Both Wikipedia Page Titles and the tweets in the corpus were preprocessed using using pyTweetCleaner. Figure 2 shows a plot of average no. of tweets published within each hour of the day for this corpus.

There were several parameters that affect the performance of our model like time window size, number of subwindows $M$, hashtag weight $H$, number of neighbors $k$ while clustering, and threshold $T$. We set a time window to be of 24 hours which contains $M = 12$ subwindows of 2 hours each. We set $H = 3, k = 3$ neighbors and $T = 4$ in our work.

Allan et al. (1998) define precision as “the fraction of the detected events that are related to a realistic event”. Moreover, Li et al. (2012a) defines another measure called Duplicate Event Rate (DERate) as “the percentage of events that have been duplicately detected among all realistic events detected”. We use these definitions of precision and DERate in our evaluation. We did not use recall as a measure to evaluate the results found by our model because we find a lack of an exhaustive list of events in the Events2012 dataset (McMinn et al., 2013). Although they have provided a list of 506 events detected by their model within the period of Oct 10 - Nov 7, 2012, our model SEDTWik finds 48 events within a period of Oct 11 - Oct 17, 2012, that were not reported by them. Their later work (McMinn and Jose, 2015) also agrees with this. Table 1 shows some of the events detected by SEDTWik that were not detected by McMinn et al. (2013). Note that the event info is manually written since the summary generated is a set of tweets that is quite large to fit in the table. Instead of recall, we use No. of events, which is the number of realistic events detected, as a measure to evaluate the performance of SEDTWik.

### 3.2 Tweet Segmentation Statistics

We segmented tweets from Oct 11 - Oct 17, 2012. After removing all the retweets, this period contained 11,705,978 tweets containing 3,653,039 distinct segments. Figure 3 shows the length of the segment along with their frequency within this period. We found that many of the bigrams were named entities (e.g., [nicki minaj], [mitt romney]) or meaningful segments (e.g., passed away). Sample tweet segmentations with hashtag weight $H = 3$ are shown in Table 2. Notice that in the second row, the username @ddlovato corresponds to Demi Lovato, a pop singer. Thus,
Joe Biden and Paul Ryan will be seated at the debate tonight #VpDebate

My #TeenChoice for #ChoiceSnapchatter is @ddlovato

Amanda Todd took her own life due to cyber bullying #RipAmandaTodd #NoMoreBullying

Table 2: Sample tweet segmentations with $H = 3$. Note that “x3” in the segmentation column signifies that the segment is present 3 times in the segmentation.

<table>
<thead>
<tr>
<th>Tweet</th>
<th>Segmentation</th>
</tr>
</thead>
<tbody>
<tr>
<td>Joe Biden and Paul Ryan will be seated at the debate tonight #VpDebate</td>
<td>[joe biden], [paul ryan], [seated], [debate], [tonight], [vp debate]x3</td>
</tr>
<tr>
<td>My #TeenChoice for #ChoiceSnapchatter is @ddlovato</td>
<td>[teen choice]x3, [choice snapshot]x3, [demi lovato]</td>
</tr>
<tr>
<td>Amanda Todd took her own life due to cyber bullying #RipAmandaTodd #NoMoreBullying</td>
<td>[amanda todd], [cyber bullying], [rip amanda todd]x3, [no more bullying]x3</td>
</tr>
</tbody>
</table>

Table 3: Comparison of SEDTWik (our method) with Twevent for events detected during the period of Oct 11 - Oct 17, 2012.

<table>
<thead>
<tr>
<th>Method</th>
<th>No. of events</th>
<th>Precision</th>
<th>DERate</th>
</tr>
</thead>
<tbody>
<tr>
<td>SEDTWik</td>
<td>79</td>
<td>88.12%</td>
<td>14.10%</td>
</tr>
<tr>
<td>Twevent</td>
<td>42</td>
<td>80.32%</td>
<td>16.67%</td>
</tr>
</tbody>
</table>

Figure 3: Segment length distribution during the period of Oct 11 - Oct 17, 2012.

3.3 Event Detection Results

Twevent (Li et al., 2012a) performs event detection from tweets using tweet segmentation and outperformed EDCoW (Weng and Lee, 2011) which was the state-of-the-art method that time, in terms of more no. of events, higher precision and recall, and less duplication rate. Since our model SEDTWik is an extension of Twevent, we set Twevent’s results as a baseline for our model and compare both these models in this section.

Table 3 shows the comparison of SEDTWik with Twevent in terms of no. of events, precision, and DERate for events detected in the period of Oct 11 - Oct 17, 2012. Recall that we are not calculating recall of our model because of lack of an exhaustive list of events within this period. Note that for calculating the results of Twevent, instead of Microsoft Web N-gram service, we used our estimates of probability which was also used in our model for the same task. The reason is that Microsoft has discontinued providing this Web-service. After the event clusters and summary are generated, we manually annotate the clusters as realistic or non-realistic event and calculate the precision on them.

As shown in Table 3, SEDTWik achieved a precision of 88.12% as compared to 80.32% by Twevent. The no. of events detected by SEDTWik were significantly more than that by Twevent (79 vs 42). In terms of DERate, SEDTWik performs slightly better than Twevent (14.10% vs 16.67%). Thus, our model SEDTWik outperforms Twevent in all the three metrics.

Edouard et al. (2017) and TwitterNews+ (Hasan et al., 2016) also evaluated their models on the same Events2012 dataset (McMinn et al., 2013) but on a different period of tweets and were able to get precision values of 75.0% and 78.0% only. Since we have to manually annotate the results, we did not re-evaluate the results of our model on these tweets but we believe our model would outperform both these models in terms of precision.

Table 4 shows some of the events detected by SEDTWik for each day in the period Oct 11 - Oct 17, 2012, along with top segments in the event cluster. Note that the event information is manually written since the summary consists of several tweets that is quite large to fit in the table.

SEDTWik code, the data used, and the entire
Oct 11 • [mo yan], [chinese writer], [nobel prize literature] → Chinese author Mo Yan wins the Nobel Prize in Literature.
• [national coming out day], [national coming out day], [lgbt], [coming day], [ncod] → National Coming Out Day celebrated on this day.
• [steelers], [nfl], [titans], [tnf] → Pittsburgh Steelers vs. Tennessee Titans Thursday Night Football (TNF) game.

Oct 12 • [nobel peace prize], [nobel], [european union], [peace prize] → The European Union wins the 2012 Nobel Peace Prize.
• [nlds], [st louis cardinals], [cardinal nation], [washington nationals] → St. Louis Cardinals win their National League Divisional Series (NLDS) against Washington Nationals.

Oct 13 • [xfactor], [x factor], [james arthur], [rylan clark] → X Factor UK finalists James Arthur and Rylan Clark give a live show in London.
• [national no bra day], [no bra day], [th october] → National No Bra Day celebrated on 13th October.

Oct 14 • [arlen specter], [passed away], [sen arlen specter] → Former US Senator Arlen Specter, died at the age of 82.
• [taylor swift], [xfactor], [the x factor] → Pop singer Taylor Swift performs live at the X Factor UK.

Oct 15 • [justin bieber], [baabworldrecord], [vevo] → Justin Bieber’s music video Beauty and a Beat (BAAB) creates world record of most watched VEVO video in 24 hrs.
• [breast cancer awareness month], [breast cancer awareness], [cure cancer] → Every year, October is celebrated as Breast Cancer Awareness Month.

Oct 16 • [debate], [barack obama], [presidential debate] → 2nd US presidential debate between Barack Obama and Mitt Romney.
• [hilary mantel], [man booker prize], [booker prize] → Hilary Mantel wins the 2012 Man Booker Prize for her novel.

Oct 17 • [lance armstrong], [endorsement deal], [nike] → Nike ended the promotional agreements they had with Lance Armstrong when he was accused of using performance enhancing drugs.
• [emmerdale live], [emmerdalelive], [live love] → A live episode of the UK soap opera Emmerdale was broadcast, marking its 40th anniversary.

Table 4: Some of the events detected by SEDTWik for each day in the period Oct 11 - Oct 17, 2012, along with top segments in the event cluster.

3.4 Impact of $H$ and $T$

Recall that while performing tweet segmentation in subsection 2.1, we used $H$ as hashtag weight which signifies by how many times, the frequency of a hashtag is multiplied. As most users associate a hashtag with any important tweet and that hashtag is common among similar tweets, giving more weight to hashtags seems intuitive. A lower value of $H$ would cause noisy segments from tweet text to dominate and harm the accuracy of the event detection model. Similarly, a higher value of $H$ would not allow other frequently used segments in the tweet text to become bursty and again reduce the accuracy. We experimented with $H$ values 1,2,3, and 4, and found the best results at $H = 3$.

The threshold $T$ was used in deciding if a candidate event cluster is a realistic event or not in subsection 2.3. We observed that on increasing $T$, more event would be considered realistic that would increase the number of events detected, but reduce the precision of the model. On experimenting with different values of $T$ from 2,3,4, and 5, we found optimal results at $T = 4$.
4 Related Work

Event detection from tweets is not a new topic of research, but rather an area on which extensive research has been done over this decade. In this section, we will present some of the related works in event detection from tweets that have motivated us to research in this field.

Panagiotou et al. (2016), Weiler et al. (2016) and Farzindar and Khreich (2015) presented a survey of many approaches that have been used over the past few years for event detection from Twitter. They had also mentioned several open challenges for event detection from tweets. Our work has tried to address some of these challenges in this paper.

TwInsight (Valkanas and Gunopulos, 2013) identified events by monitoring surges in 6 emotional states (anger, fear, disgust, happiness, sadness, and surprise) and gave information about the location, timestamp, emotion, and description of the event.

EvenTweet (Abdelhaq et al., 2013) used a fixed historical usage of words for finding those that have a burstiness degree two standard deviation above mean and then clustered them. Their method was used to find localized events only. EventRadar (Boettcher and Lee, 2012) also used Twitter to detect local events like parties and art exhibitions.

McMinn et al. (2013) created a pool of events using Locality Sensitive Hashing (LSH), Cluster Summarization, and Wikipedia, and clustered them using category, temporal, and content-based features and found events distributed among 8 categories (e.g., Sports, Science & Technology, etc.). But as shown in Table 1, there were many events that were not detected by their model.

Phuvipadawat and Murata (2010) used features like hashtags, usernames, follower count, retweet count, and proper noun terms to cluster and rank breaking news detected from Twitter.

Twevent (Li et al., 2012a) used a segmentation-based event detection from tweets method. In this method, tweets were segmented using a “stickiness” score of segments, and bursty segments were selected based on the prior probability distribution of segments, and user diversity, and were clustered into events. Our work SEDTWik is extension of Twevent so we have compared these two methods in subsection 3.3.

Recently, ArmaTweet (Tonon et al., 2017) used semantic event detection on Tweets to detect events such as ‘politician dying’ and ‘militia terror act’.

5 Conclusion And Future Work

Twitter has experienced an explosive increase in both users and the volume of information in the recent time which has attracted great interests from both industry and academia. Tweets being short and containing noisy data in large volume poses challenges on event detection task. In this paper, we presented SEDTWik - a tweet segmentation-based event detection system in which hashtags, retweet count, user popularity, and follower count were key features. Giving more weight to hashtags significantly improved the model’s performance. Our model achieved outstanding results on event detection from tweets using Wikipedia. As part of future work, we will explore ways to improve the segmentation process and use URL links in the event detection process. We will try to find more efficient ways to estimate segment probabilities considering the days and months in which specific segments are observed. We also plan to apply more sophisticated methods for event summarization that also leverage the segment and cluster information instead of just using the them to find tweets to use for summarization.

Acknowledgments

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Multimodal Machine Translation with Embedding Prediction

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Abstract

Multimodal machine translation is an attractive application of neural machine translation (NMT). It helps computers to deeply understand visual objects and their relations with natural languages. However, multimodal NMT systems suffer from a shortage of available training data, resulting in poor performance for translating rare words. In NMT, pretrained word embeddings have been shown to improve NMT of low-resource domains, and a search-based approach is proposed to address the rare word problem. In this study, we effectively combine these two approaches in the context of multimodal NMT and explore how we can take full advantage of pretrained word embeddings to better translate rare words. We report overall performance improvements of 1.24 METEOR and 2.49 BLEU and achieve an improvement of 7.67 F-score for rare word translation.

1 Introduction

In multimodal machine translation, a target sentence is translated from a source sentence together with related nonlinguistic information such as visual information. Recently, neural machine translation (NMT) has superseded traditional statistical machine translation owing to the introduction of the attentional encoder-decoder model, in which machine translation is treated as a sequence-to-sequence learning problem and is trained to pay attention to the source sentence while decoding (Bahdanau et al., 2015).

Most previous studies on multimodal machine translation are classified into two categories: visual feature adaptation and data augmentation. In visual feature adaptation, multitask learning (Elliott and Kádár, 2017) and feature integration architecture (Caglayan et al., 2017a; Calixto et al., 2017) are proposed to improve neural network models. Data augmentation aims to deal with the fact that the size of available datasets for multimodal translation is quite small. To alleviate this problem, parallel corpora without a visual source (Elliott and Kádár, 2017; Grönoos et al., 2018) and pseudo-parallel corpora obtained using back-translation (Helcl et al., 2018) are used as additional learning resources.

Due to the availability of parallel corpora for NMT, Qi et al. (2018) suggested that initializing the encoder with pretrained word embedding improves the translation performance in low-resource language pairs. Recently, Kumar and Tsvetkov (2019) proposed an NMT model that predicts the embedding of output words and searches for the output word instead of calculating the probability using the softmax function. This model performed as well as conventional NMT, and it significantly improved the translation accuracy for rare words.

In this study, we introduce an NMT model with embedding prediction for multimodal machine translation that fully uses pretrained embeddings to improve the translation accuracy for rare words.

The main contributions of this study are as follows:

1. We propose a novel multimodal machine translation model with embedding prediction and explore various settings to take full advantage of word embeddings.
2. We show that pretrained word embeddings improve the model performance, especially when translating rare words.

2 Multimodal Machine Translation with Embedding Prediction

We integrate an embedding prediction framework (Kumar and Tsvetkov, 2019) with the multimodal
machine translation model and take advantage of pretrained word embeddings. To highlight the effect of pretrained word embeddings and embedding prediction architecture, we adopt IMAGINATION (Elliott and Kádár, 2017) as a simple multimodal baseline.

IMAGINATION jointly learns machine translation and visual latent space models. It is based on a conventional NMT model for a machine translation task. In latent space learning, a source sentence and the paired image are mapped closely in the latent space. We use the latent space learning model as it is, except for the preprocessing of images. The models for each task share the same textual encoder in a multitask scenario.

The loss function for multitask learning is the linear interpolation of loss functions for each task.

\[ J = \lambda J_T(\theta, \phi_T) + (1 - \lambda) J_V(\theta, \phi_V) \]  

where \( \theta \) is the parameter of the shared encoder; \( \phi_T \) and \( \phi_V \) are parameters of the machine translation model and latent space model, respectively; and \( \lambda \) is the interpolation coefficient\(^1\).  

### 2.1 Neural Machine Translation with Embedding Prediction

The machine translation part in our proposed model is an extension of Bahdanau et al. (2015). However, instead of using the probability of each word in the decoder, it searches for output words based on their similarity with word embeddings. Once the model predicts a word embedding, its nearest neighbor in the pretrained word embeddings is selected as the system output.

\[
\hat{e}_j = \text{tanh}(W_o s_j + b_o) \quad (2) \\
\hat{y}_j = \underset{w \in V}{\text{argmin}} \{d(\hat{e}_j, e(w))\} \quad (3)
\]

where \( s_j, \hat{e}_j, \) and \( \hat{y}_j \) are the hidden state of the decoder, predicted embedding, and system output, respectively, for each timestep \( j \) in the decoding process. \( e(w) \) is the pretrained word embedding for a target word \( w \). \( d \) is a distance function that is used to calculate the word similarity. \( W_o \) and \( b_o \) are parameters of the output layer.

We adopt margin-based ranking loss (Lazaridou et al., 2015) as the loss function of the machine translation model.

\[
J_T(\theta, \phi_T) = \sum_{j}^{M} \max\{0, \gamma + d(\hat{e}_j, e(w_j^-)) - d(\hat{e}_j, e(y_j))\} \quad (4)
\]

\[
 w_j^- = \arg\max_{w \in V} \{d(\hat{e}_j, e(w)) - d(\hat{e}_j, e(y_j))\} \quad (5)
\]

where \( M \) is the length of a target sentence and \( \gamma \) is the margin\(^2\). \( w_j^- \) is a negative sample that is close to the predicted embedding and far from the gold embedding as measuring using \( d \).

Pretrained word embeddings are also used to initialize the embedding layers of the encoder and decoder, and the output layer of the decoder. The embedding layer of the encoder is updated during training, and the embedding layer of the decoder is fixed to the initial value.

### 2.2 Visual Latent Space Learning

The decoder of this model calculates the average vector over the hidden states \( h_i \) in the encoder and maps it to the final vector \( \hat{v} \) in the latent space.

\[ \hat{v} = \text{tanh}(W_v \cdot \frac{1}{N} \sum_{i}^{N} h_i) \quad (6) \]

where \( N \) is the length of an input sentence and \( W_v \in \mathbb{R}^{N \times M} \) is learned parameter of the model.

We use max margin loss as the loss function; it learns to make corresponding latent vectors of a source sentence and the paired image closer.

\[
J_V(\theta, \phi_V) = \sum_{v', v \neq v} \max\{0, \alpha + d(\hat{v}, v') - d(\hat{v}, v)\} \quad (7)
\]

where \( v \) is the latent vector of the paired image; \( v' \), the image vector for other examples; and \( \alpha \), the margin that adjusts the sparseness of each vector in the latent space\(^3\).

### 3 Experiment

#### 3.1 Dataset

We train, validate, and test our model with the Multi30k (Elliott et al., 2016) dataset published in the WMT17 Shared Task.

We choose French as the source language and English as the target one. The vocabulary size of both the source and the target languages is 10,000.

\(^1\)We use \( \lambda = 0.01 \) in the experiment.

\(^2\)We use \( \gamma = 0.5 \) in the experiment.

\(^3\)We use \( \alpha = 0.1 \) in our experiment.
Following Kumar and Tsvetkov (2019), byte pair encoding (Sennrich et al., 2016) is not applied. The source and target sentences are preprocessed with lower-casing, tokenizing and normalizing the punctuation.

Visual features are extracted using pretrained ResNet (He et al., 2016). Specifically, we encode all images in Multi30k with ResNet-50 and pick out the hidden state in the pool5 layer as a 2,048-dimension visual feature. We calculate the centroid of visual features in the training dataset as the bias vector and subtract the bias vector from all visual features in the training, validation and test datasets.

3.2 Model

The model is implemented using nmtpytorch toolkit v3.0.0\(^4\) (Caglayan et al., 2017b).

The shared encoder has 256 hidden dimensions, and therefore the bidirectional GRU has 512 dimensions. The decoder in NMT model has 256 hidden dimension. The input word embedding size and output vector size is 300 each. The latent space vector size is 2,048.

We used the Adam optimizer with learning rate of 0.0004. The gradient norm is clipped to 1.0. The dropout rate is 0.3.

BLEU (Papineni et al., 2002) and METEOR (Denkowski and Lavie, 2014) are used as performance metrics. We also evaluated the models using the F-score of each word; this shows how accurately each word is translated into target sentences, as was proposed in Kumar and Tsvetkov (2019). The F-score is calculated as the harmonic mean of the precision (fraction of produced sentences with a word that is in the references sentences) and the recall (fraction of reference sentences with a word that is in model outputs). We ran the experiment three times with different random seeds and obtained the mean and variance for each model.

To clarify the effect of pretrained embeddings on machine translation, we also initialized the encoder and decoder of our models with random values instead of pretrained embeddings, and investigated the effect of fixing decoder embeddings.

3.3 Word Embedding

We use publicly available pretrained FastText (Bojanowski et al., 2017) embeddings (Grave et al., 2018). These word embeddings are trained on Wikipedia and Common Crawl using the CBOW algorithm, and the dimension is 300.

The embedding for unknown words is calculated as the average embedding over words that are a part of pretrained embeddings but are not included in the vocabularies. Both the target and the source embeddings are preprocessed according to Mu and Viswanath (2018), in which all embeddings are debiased to make the average embedding into a zero vector and the top five principal components are subtracted for each embedding.

4 Results

Table 1 shows the overall performance of the proposed and baseline models. Compared with randomly initialized models, our model outperforms the text-only baseline by +2.49 BLEU and +1.24 METEOR, and the multimodal baseline by +2.31 BLEU and +1.09 METEOR, respectively. While pretrained embeddings improve NMT/IMAGINATION models as well, the improved models are still beyond our model.

Table 2 shows the results of ablation experiments of the initialization and fine-tuning methods. The pretrained embedding models outperform other models by up to +2.77 BLEU and +1.37 METEOR.

5 Discussion

Rare Words Our model shows a great improvement for low-frequency words. Figure 1 shows a variety of F-score according to the word frequency in the training corpus. Whereas IMAGINATION improves the translation accuracy uniformly, our model shows substantial improvement.
Table 2: Results on test dataset with variations of model initialization and fine-tuning in decoder.

<table>
<thead>
<tr>
<th>Encoder</th>
<th>Decoder</th>
<th>Fixed</th>
<th>BLEU</th>
<th>METEOR</th>
</tr>
</thead>
<tbody>
<tr>
<td>fasttext</td>
<td>fasttext</td>
<td>Yes</td>
<td>53.49</td>
<td>43.89</td>
</tr>
<tr>
<td>random</td>
<td>fasttext</td>
<td>Yes</td>
<td>53.22</td>
<td>43.83</td>
</tr>
<tr>
<td>fasttext</td>
<td>random</td>
<td>No</td>
<td>51.53</td>
<td>43.07</td>
</tr>
<tr>
<td>random</td>
<td>random</td>
<td>No</td>
<td>51.42</td>
<td>42.77</td>
</tr>
<tr>
<td>fasttext</td>
<td>fasttext</td>
<td>No</td>
<td>51.42</td>
<td>42.88</td>
</tr>
<tr>
<td>random</td>
<td>fasttext</td>
<td>No</td>
<td>50.72</td>
<td>42.52</td>
</tr>
</tbody>
</table>

Table 3: Ablation experiments of visual features. “−Debias” denotes the result without subtracting the bias vector. “−Images” shows the result of text-only NMT with embedding prediction.

<table>
<thead>
<tr>
<th>Visual Feature</th>
<th>val BLEU</th>
<th>test BLEU</th>
<th>test METEOR</th>
</tr>
</thead>
<tbody>
<tr>
<td>−Debias</td>
<td>53.14</td>
<td>53.49</td>
<td>43.89</td>
</tr>
<tr>
<td>−Images</td>
<td>52.65</td>
<td>53.27</td>
<td>43.91</td>
</tr>
</tbody>
</table>

Figure 1: F-score of word prediction per frequency breakdown in training corpus.

6 Related Works

Most studies on multimodal machine translation are divided into two categories: visual feature adaptation and data augmentation.

First, in visual feature adaptation, visual features are extracted using image processing techniques and then integrated into a machine translation model. In contrast, most multitask learning models use latent space learning as their auxiliary task. Elliott and Kádár (2017) proposed the IMAGINATION model that learns to construct the corresponding visual feature from the textual hidden states of a source sentence. The visual model shares its encoder with the machine translation model; this helps in improving the textual encoder.

Second, in data augmentation, parallel corpora without images are widely used as additional train-
Grönroos et al. (2018) trained their multimodal model with parallel corpora and achieved state-of-the-art performance in the WMT 2018. However, the use of monolingual corpora has seldom been studied in multimodal machine translation. Our study proposes using word embeddings that are pretrained on monolingual corpora.

7 Conclusion

We have proposed a multimodal machine translation model with embedding prediction and showed that pretrained word embeddings improve the performance in multimodal translation tasks, especially when translating rare words.

In the future, we will tailor the training corpora for embedding learning, especially for handling the embedding for unknown words in the context of multimodal machine translation. We will also incorporate visual features into contextualized word embeddings.

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Deep Learning and Sociophonetics: Automatic Coding of Rhoticity Using Neural Networks

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Abstract
Automated extraction methods are widely available for vowels (Rosenfelder et al., 2014), but automated methods for coding rhoticity have lagged far behind. R-fulness versus r-lessness (in words like park, store, etc.) is a classic and frequently cited variable (Labov, 1966), but it is still commonly coded by human analysts rather than automated methods. Human-coding requires extensive resources and lacks replicability, making it difficult to compare large datasets across research groups (Yaeger-Dror et al., 2008; Heselwood et al., 2008). Can reliable automated methods be developed to aid in coding rhoticity? In this study, we use Neural Networks/Deep Learning, training our model on 208 Boston-area speakers.

1 Introduction
Despite advances in automation for phonetic alignment and extraction of vowel formants, there is still no reliable automated method for classifying r-dropping, that is, whether a given word is pronounced with an /r/ in words like park (pahk), start (staht), and so on. R-dropping, also known as non-rhotic speech, is an important sociolinguistic variable in modern dialect research. But unfortunately most researchers continue to depend on human judgments (Nagy and Irwin, 2010; Becker, 2009; Nagy and Roberts, 2004), which is an inconsistent and time-consuming method that lacks replicability. Turning to the field of machine learning, our deep learning approach investigates a new way to distinguish rhotic versus non-rhotic pronunciations in recorded data. This is the first study to use neural networks to classify rhotic versus non-rhotic speech.

Although human-coding requires extensive resources and lacks replicability and replicability (Yaeger-Dror et al., 2008; Heselwood et al., 2008), making it difficult to compare large datasets across different research groups, it is the only method we have right now. How soon will computers be able to quickly and reliably code rhoticity up to this standard? In terms of other machine learning approaches, McLarty, Jones, and Hall work on this challenge using Support Vector Machines (SVMs) (Mclarty et al., 2018). The present study uses Neural Networks/Deep Learning, one of the most effective and fastest-growing approaches in machine-learning. To our knowledge, this is the first attempt to use neural networks for automatic coding of any sociophonetic variable.

This new method was developed using audio recordings from over 200 New England speakers from Boston, Maine, and central New Hampshire (Stanford, forthcoming), and is here compared to other work on rhoticity (Heselwood et al., 2008; Mclarty et al., 2018). In what ways can neural networks be effective tools in assisting the coding of rhoticity? To what level can they perform compared to traditional coding methods and other approaches?

2 Background
The phoneme /r/ has been particularly difficult to pin down because it may be articulated in different ways, yet still produce the same acoustic signal. As most phoneticians have come to agree, F3 is one of the primary acoustic correlates of rhoticity (Espy-Wilson et al., 2000; Hagiwara, 1995; Thomas, 2011). The general consensus is that the F3 measurement for /r/ is lower than that of other non-rhotic vowels, but reliable standards for coding rhoticity are lacking.

In this paper, rhoticity will refer to post-vocalic realizations of the phoneme /r/ which do not occur before other vowels. For example, rhotic tokens of interest would include park and father but not...
marr y. British phonetician John Wells used the term “rhotic”, which has been subsequently considered in the field as one of the most defining traits of varieties of English (Wells, 1982).

Rhotic and non-rhotic dialects have been widely studied as they relate to sociolinguistic features of location, age, gender, and socioeconomic status. However, we are still reliant on human analysts to make judgements of rhotic vs. non-rhotic speech, which can require a lot of time and money. Despite advances in many areas of computational linguistics, there is still not an accurate way to determine rhoticity based on acoustic components alone; a human must judge for themselves whether or not an /r/ has been dropped. As expected, this is not highly replicable as different speakers may perceive things differently especially when it comes to dialects that are not so clear-cut (Yaeger-Dror et al., 2008). For this reason, an automated way to determine rhotic/non-rhotic tokens would be especially helpful in these contexts.

3 Other work

3.1 Heselwood, Plug, and Tickle

Heselwood et al. (2008) extracted formant data from the spectrograms on the Bark scale – usually, formant data F2/F3 is reported on the Hertz scale. The Bark scale more closely correlates to human perception of sounds, that is, on a logarithmic scale rather than absolute. After conversion, F2 was labeled Z2 and F3 was labeled Z3, and a series of perceptual experiments were performed to ascertain rhoticity thresholds. Note that it was conducted for the purposes of perceptual research rather than coding applications.

3.2 McLarty, Jones, and Hall

McLarty et al. (2018) trained a Support Vector Machine (SVM) on pre-vocalic /r/ and vowels, and their approach did quite well in classifying prevocalic /r/s. They then took this pre-trained model and applied it to classifying postvocalic /r/ tokens, which classified 84% as vowels, and 15% as /r/. As they describe, this is likely because all postvocalic segments still contain vowel-like properties; furthermore, their training set excluded postvocalic /r/ so the accuracy is expected to decrease.

However, their method did not perform as well in comparison to humans. On tokens where there was no ground truth, humans only agreed with the SVM classification about 55% of the time.

4 Methods

In this initial study, we used Boston-area field recordings of 208 speakers, 100 tokens per speaker (107 women/101 men, born 1915-1997). These on-the-street interviews (15-20 minutes each) are typical sociolinguistic recordings in terms of speech styles (word-list, sentences, reading passage, free speech) and occasional background noise. We chose to omit free speech because its token variability between speakers would present another challenging factor, leaving us with recordings where participants were reading (word-list, sentences, passage). Given word transcriptions, we used the Montreal Forced Aligner (McAuliffe et al., 2017) and modified Praat scripts (DiCanio, 2014; Koops, 2013) to align and extract vowel+(r) sequences, e.g., park, short. However, note that because non-rhotic dialects are less common, and some of our recordings had background noise, it could be possible that alignments were not perfect for all of our tokens.

Two human analysts listened to recordings and judged each vowel+(r) token as r-ful or r-less. The human analysts agreed on 89.9% of the tokens, similar to human agreement elsewhere (Nagy and Irwin, 2010). Like other studies, we omitted tokens when the human analysts disagreed (10%). So overall, 1700 tokens were discarded because of speaker disagreement, and 6500 rhotic tokens and 5300 non-rhotic tokens remained for analysis.

4.1 Preliminary Investigations

In early testing, we attempted classification into r-ful, r-less, and unknown, but this did not provide strong results so we simplified to a binary classification. From the beginning of this project, we knew we wanted to use a machine learning approach, so before using neural networks we tried some easier classifiers. However, we did not get encouraging results. For example, our Random Forest Classifier only gave about 54% accuracy. When we tried simpler neural networks, these gave much more promising results to we chose to pursue this method.

4.2 Data Extraction and Model Specifications

Following standard methods of Automatic Speech Recognition, we converted the audio to 12 Mel-Frequency-Cepstral-Coefficients (MFCCs). We used the 12 MFCCs, similar to McLarty et al.. For each vowel+(r) sequence, we normalized across
the length to extract 100 time-points per token, as shown in figure 1. In the training, MFCCs were more effective than traditional sociophonetic /r/ correlates F2 and F3 (Thomas, 2011). These samples were used in the model architecture as shown in figure 1, where there are 100 samples for each vowel + /r/ sequence. The Gated Recurrent Unit is shown in more detail in figure 2, where we can see the input from the previous timestep and layer, and how this is filtered through gates using \textit{tanh} and \textit{sigmoid} activation functions.

Figure 1: Model architecture.

Importantly, no work on coding rhoticity has made use of Recurrent Neural Networks, and we believe our methods are a promising step. We used Gated Recurrent Units (Cho et al., 2014; Chung et al., 2014) to train our system to classify vowel+(r) tokens as r-ful or r-less. Following standard methods in machine-learning, we split the data in order to train with 80% of the data and test with 20%.

We chose hyperparameters based on a grid search using 3-fold cross validation (only 3 due to the small dataset). We saved the test set to validate results. The hidden layer size was 50 nodes, and dense layer size was 200 nodes. For regularization we used a kernel L2 regularization for the dense layer and we used both activation L2 and Recurrent L2 for the GRU layer. All of the alphas for this regularization are 0.01. The optimization method was RMSprop, and the learning rate was 0.001.

5 Results

In figure 3, we see the Normalized Confusion Matrix, which summarizes our results by lining up true labels and predicted labels for our rhotic and non-rhotic tokens. We consider this binary classification either rhotic (positive) or non-rhotic (negative). In this way we can see the proportion of true positives (predicted to be rhotic and indeed truly rhotic), false positive (predicted to be rhotic but actually non-rhotic), true negative (predicted to be non-rhotic and actually non-rhotic), and false negative (predicted to be non-rhotic and actually rhotic). In deciding which model to use, we tried a few different configurations. We used the sampled MFCCs (as described earlier, figure 1) as well as Bark measurements that were extracted also at 100 time-points across the vowel. Because our MFCC data is multi-dimensional and time-dependent, we wanted to see how a Convolutional Neural Network would perform (table 1), but it turned out not to be as high in performance as our earlier model.

Figure 4 shows the Receiver Operating Characteristic (ROC) for our model (created using scikit-learn), which is fairly good by machine learning standards. The Area Under the Curve (AUC, as noted in Table 1) is 0.892, and as evident from the
graph, is much closer to 1. Our system had 81.1% accuracy with the human analysts in judging tokens as r-less or r-ful, scoring 0.829 for F-measure.

![Receiver Operating Characteristic](image)

Figure 4: Receiver Operating Characteristic.

<table>
<thead>
<tr>
<th>Model</th>
<th>Accuracy</th>
<th>Precision</th>
<th>Recall</th>
<th>F1</th>
<th>AUC</th>
</tr>
</thead>
<tbody>
<tr>
<td>GRU-MFCC</td>
<td>0.811</td>
<td>0.829</td>
<td>0.830</td>
<td>0.829</td>
<td>0.892</td>
</tr>
<tr>
<td>GRU-Bark</td>
<td>0.806</td>
<td>0.844</td>
<td>0.856</td>
<td>0.850</td>
<td>0.869</td>
</tr>
<tr>
<td>CNN-MFCC</td>
<td>0.746</td>
<td>0.796</td>
<td>0.815</td>
<td>0.805</td>
<td>0.808</td>
</tr>
</tbody>
</table>

Table 1: Metrics showing the performance of different models – our top performing model was using GRUs with MFCCs as input (as described previously).

We also used the Heselwood et al. approach (section 3.1) of classifying front or back vowels to see how accurately it would perform on the same test dataset. This classification gave an average speaker accuracy of 63.3% and an average token accuracy of 62.1% (Table 2), much lower than our best model’s overall accuracy (i.e. average across all tokens) of 81.1% (Table 1).

<table>
<thead>
<tr>
<th></th>
<th>Average Speaker Accuracy</th>
<th>Average Token Accuracy</th>
</tr>
</thead>
<tbody>
<tr>
<td>Heselwood et al.</td>
<td>63.3%</td>
<td>62.1%</td>
</tr>
</tbody>
</table>

Table 2: Heselwood et al. approach on test dataset (using Bark thresholds Z2 and Z3)

6 Discussion

The initial results of this study are promising. Our results are quite strong, as shown by the metrics in Table 1. When testing the Heselwood et al. approach (Table 2), it only predicted correctly approximately 60% of the time; our model performs significantly better, at an accuracy of 81.1% (Table 1). It seems that we are also slightly better at predicting rhotic tokens than non-rhotic (Figure 3), which likely has to do with the fact that we have more rhotic tokens in total.

We aimed to reach human levels – considering that analyst agreement is 89.9% for our dataset (as mentioned above), our accuracy of 81.1% is quite good. However, these numbers are not strictly comparable as we discarded tokens that proved difficult for human analysts.

In future development of this method, we want to consider any sources of error on our part. For example, some audio and text files could be misaligned so we might consider hand-correcting these alignments. However, the nature of the neural network could correct for this in that it learns to forget irrelevant or noisy data. By gathering more data, we would expect that our accuracy would improve and eventually reach a plateau where additional speakers would not affect anything.

Additionally, a study that involves cross-corpus analysis could provide greater insight into how this model might be applicable on a larger scale, and how well our model actually performs. Furthermore, if we had 3 analysts rather than 2, we could have used a majority vote for classifying tokens, and would not have to discard tokens where rhoticity was ambiguous.

A shortcoming of this study is that it only involves speech that is elicited through reading – ideally future studies would involve free speech in order to use more natural speech.

R-dropping is a crucial sociolinguistic variable for English dialect research in the US Northeast, Great Britain, Australia, New Zealand, Singapore, and other locations. Our neural network model takes a significant step toward automation of this key variable. In the future, we will continue optimizing and improving our model. Other groups have studied automated methods for coding sociolinguistic variables (Yuan and Liberman, 2011; Bailey, 2016), and there are great ideas to be found in these works. When automated methods for rhoticity reach the accuracy level of humans, along with consistency and full replicability, this will open the floodgates to large amounts of /r/ data and greatly expand sociolinguistic knowledge of dialect variation around the world, efficiently allowing studies to be replicated across research groups.
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Data Augmentation by Data Noising for Open-vocabulary Slots in Spoken Language Understanding

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Abstract
One of the main challenges in Spoken Language Understanding (SLU) is dealing with ‘open-vocabulary’ slots. Recently, SLU models based on neural network were proposed, but it is still difficult to recognize the slots of unknown words or ‘open-vocabulary’ slots because of the high cost of creating a manually tagged SLU dataset. This paper proposes data noising, which reflects the characteristics of the ‘open-vocabulary’ slots, for data augmentation. We applied it to an attention based bi-directional recurrent neural network (Liu and Lane, 2016) and experimented with three datasets: Airline Travel Information System (ATIS), Snips, and MIT-Restaurant. We achieved performance improvements of up to 0.57% and 3.25 in intent prediction (accuracy) and slot filling (f1-score), respectively. Our method is advantageous because it does not require additional memory and it can be applied simultaneously with the training process of the model.

1 Introduction
Dialog processing enables dialogue between humans and voice assistants such as ‘Siri’ and ‘Alexa’. In dialogue processing, spoken language understanding (SLU) is aimed at understanding and generating the user intention from an utterance. The user intention consists of an intent and slots, which are semantic entities, and it is generally defined variously according to the domain.

Neural networks (NNs) have been actively studied and applied to SLU. Xu and Sarikaya (2013) and Vu (2016) proposed models for SLU using Convolutional Neural Networks and many other researchers used recurrent neural networks (RNNs). Liu and Lane (2016) used bi-directional RNN models and applied the attention mechanism. They showed good results by using a joint learning method for both slot filling and intent prediction tasks. For considering the relationship between the two tasks, Wang et al. (2018) constructed two models for each task and Goo et al. (2018) used a ‘slot-gate’.

Training an SLU model using an NN requires a large amount of training data labeled with slots and intents, which is expensive to build. In particular, plenty of corpora or dictionaries are required to recognize the value of an ‘open-vocabulary’ slot, such as a song title. In addition, it is difficult to predict the slot type of words used in the ‘open-vocabulary’ slot because there is neither a semantic restriction nor a length limit.

Kim et al. (2018) presented the features and examples of ‘open-vocabulary’ slot and proposed a new model that could effectively predict this type of slot. They exploited a long-term aware attention structure and positional encoding with multi-task learning of a character-based language model and intent detection model to focus more on relatively global information within a sentence.

The objective is to recognize slots, including ‘open-vocabulary’ slots, and predict the intent effectively by data augmentation. We propose a data noising method that reflects the characteristics of the ‘open-vocabulary’ slots. This method is advantageous in that it does not require additional memory. Moreover, it is performed simultaneously with the training of the model.

2 Related works
2.1 Data noising as a form of data augmentation
Data augmentation is a technique to avoid overfitting by increasing the size of the training datasets. This technique is widely used in many machine learning tasks. A typical method in natural language processing (NLP) is generating a sentence/corpus by replacing its words with their
synonyms based on rules, dictionaries, or ontology constructed by a person. In recent years, Kobayashi (2018) proposed a method of modifying sentences by the analogy of words to be replaced using an NN-based language model to achieve data augmentation.

One of the methods of data augmentation is data noising. Data noising is an effective technique for normalizing a neural network, and has been widely applied in fields such as computer vision and speech recognition. Its application in the field of NLP is relatively limited because NLP is based on discrete values like words and the quality of the data generated is not guaranteed. Several studies have used data noising for data augmentation. Iyyer et al. (2015); Bowman et al. (2015); Kumar et al. (2016) used a method for randomly dropping input word embedding. Xie et al. (2017) improved the performance by replacing a word with another word sampled from the unigram distribution or by blanking out as interpolation in language modeling. Cheng et al. (2018) improved the robustness of neural machine translation models against noisy inputs by randomly adding Gaussian noise to the word embedding and maintaining consistent behavior of the encoder to normal and perturbed inputs through adversarial learning. They showed that data noising is effective in normalizing sequence models based on neural networks.

2.2 Data augmentation for SLU

The NN model for SLU requires numerous labeled training corpora; therefore, some studies have attempted to improve SLU performance by data augmentation. Kurata et al. (2016) proposed a method of using encoder-decoder long short-term memory (LSTM) to generate labeled data. Hou et al. (2018) generated diverse utterances of existing training data through a data augmentation framework based on sequence-to-sequence generation. Yoo et al. (2018) proposed a data augmentation method that sampled similar words using a variational autoencoder. These methods have shown good performance improvements with data augmentation, but they require new models or consume additional memory.

3 Proposed Method

3.1 Motivation

First, we would like to cite an example of an utterance in the Snips dataset as the motive for the proposed method. Consider the sentence ‘i d like a table for midday at the unseen bean’; ‘the unseen bean’ is the value of the ‘restaurant_name’ slot. Even when people read ‘the unseen bean,’ it is difficult to recognize it as the name of a restaurant without prior knowledge. However, people can infer the slot type from the context (i.e., ‘i d like a table at’).

‘Open-vocabulary’ slots, such as the name of a restaurant or the title of a song, have no restriction on the length or the specific patterns of content in the slot. Therefore, it is hard to recognize the slot type from only the words in the slots, but it is possible to predict them based on the surrounding words or the context.

Considering these linguistic features, the purpose of this study is to augment the data by transforming them into utterances with the same context/surrounding words but various slot values. It has the effect of creating utterance patterns.

3.2 Data noising for SLU

The flow of our proposed method is shown in Figure 1. It consists of two steps: data noising and using the context word window. When the training data are input, they become noised embedding vectors after data noising. Then, we use a context word window as the input to a layer of the neural network. These steps are performed per batch and the model trains with different noised data in each step. Because data augmentation is performed in the same embedding space, it does not need additional memory and is included in the training process.

Data noising: We propose a data noising method for SLU, which augments the training data by replacing the slot values with a random value while maintaining the surrounding context. The augmented data are used to train the NN for an utterance pattern. This method is shown in Figure 2 and follows the procedure described below.
1) Express an utterance \( w = \{w_1, ..., w_T \} \) as an embedding vector \( e(w) \in \mathbb{R}^{T \times k} \). \( T \) is the length of the word sequence and \( k \) is the size of the word embedding vector.

2) Extract the binary vector \( b(s) = \{b_0, ..., b_T \} \) for the slot sequence \( s = \{s_0, ..., s_T \} \) in the utterance \( w \). The value of \( b_i \) is set to 1 when the \( i \)-th word \( w_i \) is the slot value (e.g., \( s_i \) is ‘B-album’), and to 0 when it is not the slot value (e.g., \( s_i \) is ‘O’).

3) Multiply the binary vector \( b(s) \) by the randomly sampled vector \( v_{\text{noise}} \in \mathbb{R}^k \). Next, the noise is added only to the slot values, as shown in equation (1). This is to maintain unchanged the surrounding context of the slot value.

\[
e'(w) = e(w) + b(s) \cdot v_{\text{noise}}, e'(w) \in \mathbb{R}^{T \times k}
\]

The noised embedding vector \( e'(w_i) \) is located somewhere in the same embedding space as \( w_i \). Because ‘open-vocabulary’ slots can contain any word, we do not need to know the words that are replaced. Therefore, we augment the training data by replacing the words in the ‘open-vocabulary’ slots with the embedding vector of any unknown word, through data noising.

**Noised context word window**: Mesnil et al. (2015) and Zhang and Wang (2016) used a context word window to improve the performance of the RNNs in slot filling. This can reflect the context information well by examining the surrounding words together. We also use the \( d \)-context word window as an input to the recurrent layer to reflect the contextual information. In this study, the noised context word window \( e^d_i \) is the result of the concatenation of the noised embedding vector \( e^d_k(w_i) \) of the center word \( w_i \) and the noised embedding vectors of the \( d \) previous words and \( d \) next words, as shown in equation (2).

\[
e^d_i = [e^k_i(w_{i-d}), ..., e^k_i(w_i), ..., e^k_i(w_{i+d})]
\]

### 4 Experiments and Results

#### 4.1 Data

We experimented with three datasets: Airline Travel Information System (ATIS) (Hemphill et al., 1990), Snips\(^1\), and MIT-restaurant (MR)\(^2\). The statistics of each dataset are shown in Table 1. We calculated the length of each slot and identified the maximum slot length. This was to numerically confirm whether each dataset had the characteristic of an ‘open-vocabulary’ slot, namely, there was no limit on the length of the slot value.

- **ATIS**: It is used in many SLU researches (Liu and Lane, 2016; Wang et al., 2018; Kim et al., 2018), including those on utterances for flight reservations. The training set is from ATIS-2 and ATIS-3 corpora, and the testing set is from ATIS-3, NOV93, and DEC94 datasets.

- **Snips**: It is an open-sourced NLU dataset of custom-intent-engines by Snips. It is used in SLU studies (Goo et al., 2018; Yoo et al., 2018). The Snips dataset contains user utterances from various domains, such as playing music or searching a movie schedule. It has ‘open-vocabulary’ slots, such as movie title.

- **MR**: It is a single-domain dataset, which is associated with restaurant reservations. MR contains ‘open-vocabulary’ slots, such as restaurant names.

#### 4.2 Baseline and details of experiments

We set the attention based bi-directional LSTM model (Liu and Lane, 2016) without the label dependency as the baseline for the experiment. We

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\(^{1}\)https://github.com/snipsco/nlu-benchmark/tree/master/2017-06-custom-intent-engines

\(^{2}\)https://groups.csail.mit.edu/sls/downloads/restaurant/

<table>
<thead>
<tr>
<th>Dataset</th>
<th>ATIS</th>
<th>Snips</th>
<th>MR</th>
</tr>
</thead>
<tbody>
<tr>
<td>Train set</td>
<td>4,978</td>
<td>13,084</td>
<td>6,894</td>
</tr>
<tr>
<td>Evaluation set</td>
<td>-</td>
<td>700</td>
<td>766</td>
</tr>
<tr>
<td>Test set</td>
<td>893</td>
<td>700</td>
<td>1,521</td>
</tr>
</tbody>
</table>

Table 1: Statistics of ATIS, Snips, and MR datasets.
applied our data augmentation method to the baseline and evaluated the following two cases: *Just add noise* (+Noise), *Add noise and use context window* (+Noise, cw).

We followed the set-up in Liu and Lane (2016). We set the number of LSTM cells to 128, the batch size was 16, and the dropout rate was 0.5. We considered one layer and used the Adam optimizer (Kingma and Ba, 2015) for parameter optimization. The word embeddings were randomly initialized and then fine-tuned and their size was 128 for experiments with ATIS and MR, and 64 for experiments with Snips, for comparison with previous studies.

The noise vector was created with probability \( p \) and it defined the number of augmented data. In this paper, because we performed the data augmentation in batches during the training, the number of augmented utterances was defined as \((\text{the number of step} \times \text{batch size}) \times p\). The probability was set to 0.25, 0.5, 0.75, or 1.0. The noise vector was sampled randomly from the normal distribution or uniform distribution and its size was equal to the word embedding size. We set the mean of the normal distribution to 0.0, and the \( \sigma \) value to 0.1, 0.2, or 0.3. The range of the uniform distribution was set to [-0.2, 0.2] or [-0.5, 0.5].

As in previous SLU studies, we used the F1-score and the accuracy to evaluate the performance of the slot filling (SF) and the intent prediction (IP), respectively.

### 4.3 Performances of intent prediction and slot filling

Table 2 shows the performance improvements achieved by the proposed method versus the baseline. The proposed method showed clear improvements for the Snips and MR datasets. It is considered that the proposed method is effective in the two datasets because they have larger length of slots (as shown in Table 1) and more ‘open-vocabulary’ slots. Experimental results show that our approach improves slot filling of an unknown word or ‘open-vocabulary’ slots by learning the patterns of utterances. Examples illustrating the results can be found in Table 4. They show that the proposed method improves the slot filling of unknown words and ‘open-vocabulary’ slots by learning the utterance patterns.

Additionally, the proposed method is more effective when the size of the training data is small. Figure 3 shows the performance according to the training data size with ATIS.

### 4.4 Comparison with previous studies

Table 3 shows the comparison of the performance of previous studies with each data set. In the case of the ATIS dataset, our method shows better performance than the ‘Baseline’ study (Liu and Lane, 2016) and a study targeting ‘open-vocabulary’ slots (Kim et al., 2018), but it is not state-of-the-art. However, we achieve the best performance among studies using the Snips and MR datasets.
Table 4: Examples of slot filling with the Snips dataset. ‘Input’ is the input utterance, and ‘Baseline Pred.’ is the slot filling result of the baseline model. ‘Proposed pred.’ is the result of applying the proposed method and is the ground truth. Unknown words are represented as italicized text. The slot filling errors are marked with underline and instances of correct slot filling are represented in bold text.

5 Conclusion

This paper focuses on data augmentation by reflecting the characteristics of ‘open-vocabulary’ slot in order to achieve better SLU. The experiments show that the proposed method outperforms the baseline, especially with datasets that include more ‘open-vocabulary’ slots. The proposed method has the following three advantages. Data augmentation can be performed during training. It does not need additional memory because it utilizes the input embedding space. It is straightforward and intuitive; hence, it can be easily applied to any model. In this paper, we added noise in all slots without classifying types of slots, i.e., without determining whether they were open-vocabulary slots. In our future work, we will perform additional experiments by classifying the types of slots and adding noise depending on the types. In addition, we plan to apply this method to Named Entity Recognition because it also has the problem of ‘open-vocabulary’ tags.

Acknowledgments

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Expectation and Locality Effects in the Prediction of Disfluent Fillers and Repairs in English Speech

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Abstract

This study examines the role of three influential theories of language processing, viz., Surprisal Theory, Uniform Information Density (UID) hypothesis and Dependency Locality Theory (DLT), in predicting disfluencies in speech production. To this end, we incorporate features based on lexical surprisal, word duration and DLT integration and storage costs into logistic regression classifiers aimed to predict disfluencies in the Switchboard corpus of English conversational speech. We find that disfluencies occur in the face of upcoming difficulties and speakers tend to handle this by lessening cognitive load before disfluencies occur. Further, we see that reparationums behave differently from disfluent fillers possibly due to the lessening of the cognitive load also happening in the word choice of the reparationandum, i.e., in the disfluency itself. While the UID hypothesis does not seem to play a significant role in disfluency prediction, lexical surprisal and DLT costs do give promising results in explaining language production. Further, we also find that as a means to lessen cognitive load for upcoming difficulties speakers take more time on words preceding disfluencies, making duration a key element in understanding disfluencies.

1 Introduction

In contrast to written text which can be rewritten or edited, speech happens spontaneously making it more prone to mistakes. Speakers tend not to speak fluently and take pauses or even repeat words. Such errors where speakers interrupt their flow of speech are known as disfluencies. One of the primary reasons for speech disfluencies is difficulties in language production (Tree and Clark, 1997; Clark and Wasow, 1998). In this study, we aim to understand the role of disfluencies and classify disfluencies into two categories namely, disfluent fillers and reparationums. Disfluent fillers are utterances like *uh*, *um* which break fluency by interjecting and creating an interruption between words. For example, suppose a speaker says “thinking about the *uh* day when I”. Here, there is a break of fluency between the words *the* and *day* due to the interjection of the filler *uh*. Reparationums involve cases where speakers break fluency by making corrections in their speech. For example, when a speaker says “*Go to the right* to the left”. Here, the speaker makes a correction to *to the right* by restarting with the intended (corrected) speech *to the left*. We call the words to be corrected as the reparationandum (*to the right*) and the correction the speaker follows with as the repair (*to the left*).

In order to study disfluencies, we use transcribed data from the Switchboard corpus (Godfrey et al., 1992), a corpus of fully spontaneous speech of American English. We focus on testing the role of three influential linguistic theories, viz., Surprisal Theory (Levy, 2008; Hale, 2001), Uniform Information Density (UID) hypothesis (Jaeger and Levy, 2007) and Dependency Locality Theory (Gibson, 2000) in accounting for disfluencies. Surprisal Theory defines an information-theoretic measure of comprehension difficulty *viz.*, surprisal. Recently, Demberg et al. (2012) showed that syntactic surprisal is a significant predictor of word duration in spontaneous speech even amidst the presence of competing controls like lexical frequency. Thus surprisal can be used to model language production as well, with words with high surprisal associated with speech disfluencies *i.e.*, fillers and repairs. The UID hypothesis predicts that in language production, speakers prefer to minimize variation of information density (mathematically same as surprisal) across the speech signal. Thus based on the UID hypothesis, it is plausible to assume that disfluencies are associated with higher informa-
tion density variation. Finally, DLT posits integration and storage costs as measures of comprehension difficulty. Scontras et al. (2015) showed that for English relative clause production, locality results in greater speech disfluencies and starting time for object relatives compared to subject relatives. Thus, we conceive higher values of integration and storage costs leading to disfluencies in language production.

We predict disfluencies in the Switchboard corpus using a one-vs-all logistic regression classifier containing features based on lexical surprisal, UID, DLT-inspired costs and duration. Further, by looking into the classifier’s regression weights and accuracies, we get an insight behind how these theories affect disfluencies in speech. Our results do not uncover evidence to indicate UID hypothesis plays a significant role in disfluency prediction; however, lexical surprisal and DLT costs do give promising results in explaining language production. The latter two theories indicate that disfluencies tend to be followed with upcoming difficulties and speakers lower cognitive load on words preceding these disfluencies to ease this difficulty. Apart from these three theories, we look into how duration behaves in disfluent contexts and find that speakers take more time in words preceding disfluencies which we explain as a means to lower cognitive load for upcoming difficulties by buying more processing time. Further, we see that reparandums do not occur prior to words with lower surprisal like in the case of disfluent fillers. This effect may be due to the lessening of the cognitive load also happening in the word choice of the reparandum, i.e., in the disfluency itself.

2 Background

In the context of disfluency detection, disfluent fillers tend to be easier to identify as they mostly consist of a closed set of fixed utterances (e.g. um, uh). Reparandums on the other hand are more difficult to identify because they tend to resemble fluent words a lot more. One of the effective feature types for detecting these reparandums are distance and pattern matching based features that look into the similarity of words and POS tags with their neighbours (Honnibal and Johnson, 2014; Zayats et al., 2016; Wang et al., 2017). The reason for their effectiveness could stem from how the repair that follows the reparandum is usually a “rough copy” of the reparandum, i.e., it incorpo-rates the same or very similar words in roughly the same word order as the reparandum. Apart from this, disfluency detection has also been shown to be effective with other features like language models and lexical features (Zwarts and Johnson, 2011; Zayats et al., 2016); prosody (Shriberg et al., 1997; Kahn et al., 2005; Tran et al., 2017) and dependency based features (Honnibal and Johnson, 2014).

Seeing how disfluency detection in the past has collected features from lexical language models, dependency grammar and prosody, we are motivated to test influential linguistic theories in these domains and study whether disfluencies can be explained by these theories, viz., Surprisal Theory (Levy, 2008), the UID hypothesis (Jaeger and Levy, 2007) and DLT (Gibson, 2000). Further, to examine the effects of prosody we look into duration as a feature to explain disfluencies.

Building on Shannon’s (1948) definition of information, it has been shown in recent work formalized as Surprisal Theory (Hale, 2001; Levy, 2008) that the information content of a word is a measure of human sentence comprehension difficulty. The surprisal of a word is defined as the negative log of its conditional probability in a given context (either lexical or syntactic). The second theory we examine, the Uniform Information Density (henceforth UID) hypothesis also relates to information density and states that language production exhibits a preference for distributing information uniformly across a linguistic signal. Our third theory is the Dependency Locality Theory (henceforth DLT) proposed by Gibson (2000). This theory defines processing costs that have successfully accounted for the comprehension difficulty associated with many constructions (subject and object relative clauses for example).

3 Experiments and Results

Our study focuses on 3 classes of words in the Switchboard corpus: reparandum, disfluent filler and fluent word. We use the corpus provided by the switchboard NXT project (Calhoun et al., 2010) and base our features for machine learning from the fluent words that immediately follow or precede these disfluencies (for reparandum based disfluencies, these are taken as the words that immediately follow repair and precede the reparandum, this was done out of uniformity as disfluencies such as a disfluent filler uh do not posses the
<table>
<thead>
<tr>
<th>Features</th>
<th>Accuracy</th>
<th>Fluent</th>
<th>Filler</th>
<th>Repar.</th>
</tr>
</thead>
<tbody>
<tr>
<td>Baseline</td>
<td>37.18%</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Preceding surprisal*</td>
<td>38.12%</td>
<td>0.014</td>
<td>-0.0664</td>
<td>0.0525</td>
</tr>
<tr>
<td>Following surprisal**</td>
<td>38.78%</td>
<td>-0.1204</td>
<td>0.0915</td>
<td>0.0389</td>
</tr>
<tr>
<td>Both surprisals***</td>
<td>40.02%</td>
<td></td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

Table 1: Accuracy and weights for lexical surprisal. Here * denotes p-value < 0.05 and ** denotes p-value < 0.01 for McNemar’s test relative to the baseline.

same linguistic features as fluent words. All the cases where the surrounding words have unclear POS tags or non-aligned duration have been excluded from this dataset. This results in a total of 14923 cases of reparandum, 12183 cases of disfluent filler and 558361 cases of a fluent word. For uniformity in classes we randomly sample 12183 cases from each class. We setup a one-vs-all logistic regression classifier to classify between our 3 categories. To set up a baseline performance for this multi-classification, we train the classifier on features pertaining to the speaker and listener particularly the gender, age and rate of speech. The change in accuracy relative to the baseline on adding features, viz., lexical surprisal, UID, DLT costs and duration, tells us whether these theories explain the presence of disfluent contexts. Further, using the regression weights we look at whether the correlations are indeed as the theory expects. The next three subsections will describe these results in depth.

3.1 Lexical Surprisal

We deploy lexical surprisal as measure of predicting disfluencies. We use the definition proposed by Hale (2001) which states that the lexical surprisal of the $k^{th}$ word $w_k$ in a sentence is $S_k = -\log P(w_k | w_{k-1}, w_{k-2})$. Where $P(w_k | w_{k-1}, w_{k-2})$ refers to the conditional probability of $k^{th}$ word in the sentence given the previous two words. We calculate lexical surprisal of each word in our corpus by training a simple trigram model over words on the Open American National Corpus (Ide and Suderman, 2004) using the SRILM toolkit (Stolcke, 2002). The lexical surprisal feature of the surrounding words turns out to be significant with a p-value < 0.05 and we can note from Table 1 that these classifiers with surprisal give a significant improvement from baseline (McNemar’s test). Further, using surprisal of the word following the disfluency gives a 1.6% boost in accuracy and the regression coefficients from Table 1 indicate that the words that follow disfluencies (this would be the word following the repair in the case of reparands) show a higher surprisal, suggesting that disfluencies occur in the presence of an upcoming difficulty. Previous studies have shown similar results that disfluencies occur in the presence of production difficulties due to new information (Arnold et al., 2000; Barr, 2001; Arnold et al., 2003; Heller et al., 2015). Examples from the corpus illustrated such a behaviour where disfluent sentences such as “for the uh scud missiles” or “imagine thats a - thats a pillsbury plant?” have high surprisal difficulties like scud or pillsbury following the disfluency.

We also note that lexical surprisal of the word preceding the disfluency leads to an accuracy increase of 0.94%. There is a low surprisal of the preceding words in the case of disfluent fillers, as seen from Table 1, suggestive that speakers use easier words (lesser cognitive load due to low surprisal) to handle the production problems better. However, the other type of disfluency, reparandum shows a higher preceding surprisal which might be attributed to the fact that unlike fillers, reparands consist of words in themselves and these words may be the ones that hold the low surprisal rather than the preceding word to the reparandum.

3.2 Uniform Information Density (UID)

In order to quantify the uniformity in information density spread, we use two types of UID measures proposed in previous works by Collins (2014) and Jain et al. (2018). The two measures are as follows:

- $UID_{glob} = \frac{-1}{N} \sum_{i=1}^{N} (id_i - \mu)^2$
- $UID_{globNorm} = \frac{-1}{N} \sum_{i=1}^{N} (\frac{id_i}{\mu} - 1)^2$

Here, N is the number of words in the sentence, $id_i$ refers to the information density, i.e., lexical surprisal of $i^{th}$ word and $\mu$ is the average information density of the sentence. We note from the equation above that the uniformity measure, UIDglob is defined as the negative of variance of lexical surprisal. Further, our second measure UIDglobNorm is nothing but the normalized version of the first measure UIDglob.

We calculate the UID measures for the surrounding words, i.e., the immediate preceding and following words to the class, and the UID measures for that sentence. From the accuracies that are listed in Table 2, we can see that the best accuracy is an increase of 0.42% from UIDglob of the
sentence. However, this UIDglob measure for sentence when normalized (UIDglobNorm) shows a net decrease in accuracy. Though these UID measures on the sentence are significant features with a p-value $< 0.05$, the UID measures on the surrounding words are not. Further, these improvements in accuracy upon using sentence level UID features are significant (McNemar’s test) only with p-value 0.05 but not with 0.01. We see that the UID measure for the surrounding words causes an increase of 0.13% which is far less compared to the increase in preceding (0.94%) and following (1.6%) surprisal noted earlier. The UID hypothesis we’ve examined has hence failed to bring about any notable improvements in our model in comparison to competing explanations like surprisal theory. It may also be possible that the UID hypothesis is limited only to syntactic reduction in English (Jaeger and Levy, 2007). Previous work by Jain et al. (2018) in Hindi word order choices has shown that the UID measures does not outperform lexical surprisal.

### 3.3 DLT: Integration and Storage Costs

The central notion of DLT revolves around two costs: integration cost (IC) and storage cost (SC) as proposed by Gibson (2000). We compute DLT costs as follows: For a word to be integrated into the structure built so far, its integration cost, a backward-looking cost, would be the sum of the dependency lengths of all dependencies that include the word to be integrated and its previously encountered head/dependent word (grammatical link provided by dependency grammar). In contrast, the storage cost is a forward-looking cost and corresponds to the number of incomplete dependencies in our integrated structure thus far. To calculate these costs, the dependency relations for our corpus were extracted by removing disfluencies, i.e., both disfluent fillers and reparandum based disfluencies but contrarily is lower in the context of disfluent fillers which goes against the expectation of an upcoming difficulty in the case of disfluencies. Recent work by Demberg and Keller (2008) has shown integration cost to behave anomalously and act in the expected direction only for high values of dependency length. With preceding word’s integration cost getting lowered in the context of disfluent fillers and higher in the context of reparandum based disfluencies, we can see that apart from the anomalous result for following integration cost in fillers, integration cost functions gets explained similar to lexical surprisal. We also see that disfluencies, i.e., both disfluent fillers and reparandum based disfluencies have a lower preceding storage cost and higher following storage cost. This makes sense as a lower preceding storage cost lowers the cognitive load and helps process the upcoming difficulty better (indicated by high following storage cost).

### 3.4 Duration

Looking into how duration affects disfluencies, from Table 4, we can note that the duration of the preceding word gives a huge accuracy increase of

<table>
<thead>
<tr>
<th>Features</th>
<th>Accuracy</th>
</tr>
</thead>
<tbody>
<tr>
<td>Baseline</td>
<td>37.18%</td>
</tr>
<tr>
<td>UIDglob surrounding</td>
<td>37.22%</td>
</tr>
<tr>
<td>UIDglob sentence*</td>
<td>37.60%</td>
</tr>
<tr>
<td>UIDglob both*</td>
<td>37.49%</td>
</tr>
<tr>
<td>UIDglobNorm surrounding</td>
<td>37.31%</td>
</tr>
<tr>
<td>UIDglobNorm sentence*</td>
<td>36.94%</td>
</tr>
<tr>
<td>UIDglobNorm both*</td>
<td>37.28%</td>
</tr>
</tbody>
</table>

Table 2: Accuracy for UID measures.

<table>
<thead>
<tr>
<th>Features</th>
<th>Accuracy</th>
<th>Fluent</th>
<th>Filler</th>
<th>Repar.</th>
</tr>
</thead>
<tbody>
<tr>
<td>Preceding IC**</td>
<td>37.65%</td>
<td>-0.042</td>
<td>-0.069</td>
<td>-0.027</td>
</tr>
<tr>
<td>Following IC**</td>
<td>39.06%</td>
<td>-0.0334</td>
<td>-0.0375</td>
<td>0.0709</td>
</tr>
<tr>
<td>Preceding SC**</td>
<td>39.48%</td>
<td>0.2895</td>
<td>-0.0575</td>
<td>-0.232</td>
</tr>
<tr>
<td>Following SC**</td>
<td>37.56%</td>
<td>-0.1731</td>
<td>0.0098</td>
<td>0.1633</td>
</tr>
<tr>
<td>Both ICs &amp; SCs**</td>
<td>39.76%</td>
<td>0.0709</td>
<td>0.0027</td>
<td>0.1633</td>
</tr>
<tr>
<td>Both SCs**</td>
<td>41.61%</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Both ICs and SCs**</td>
<td>48.57%</td>
<td></td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

Table 3: Accuracy and weights for DLT costs.

parses and in this way is inspired from the original DLT that makes use of constituency parsing. We see from the McNemar’s test (Table 3) that the increase in accuracy for DLT costs are all significant improvements w.r.t baseline. The integration costs and storage costs, all significant features with p-value $< 0.05$, give an increase of 2% and 4.43% individually (from Table 3). Further, combining all these four features gives a far larger increase of 11.39% from the baseline. This can indeed be explained as the two costs serve complementary functions (can be seen from their negative correlation of -0.26) as forward and backing looking costs, and a combination would possess greater information on the whole. We see that DLT seems to perform well for our disfluency prediction task and so we will proceed to examining the correlations of the DLT costs in disfluent contexts.

From Table 3 we see that following integration cost is expectantly high for reparandum based disfluencies but contrarily is lower in the context of disfluent fillers which goes against the expectation of an upcoming difficulty in the case of disfluencies. We see from the McNemar’s test (Table 3) that the increase in accuracy for DLT costs are all significant improvements w.r.t baseline. The integration costs and storage costs, all significant features with p-value $< 0.05$, give an increase of 2% and 4.43% individually (from Table 3). Further, combining all these four features gives a far larger increase of 11.39% from the baseline. This can indeed be explained as the two costs serve complementary functions (can be seen from their negative correlation of -0.26) as forward and backing looking costs, and a combination would possess greater information on the whole. We see that DLT seems to perform well for our disfluency prediction task and so we will proceed to examining the correlations of the DLT costs in disfluent contexts.
9.64\% from baseline. From McNemar’s test it is indicated that the increase in accuracy from using duration features w.r.t baseline are all significant. The duration features are also significant features with p-value < 0.05 and are higher in the context of disfluencies as can be seen from the regression weights in Table 4. These results are in concert with Bell et al’s (2003) study of duration in disfluent contexts. The higher duration of words preceding disfluencies suggests that speakers try to buy time in order to better process for the upcoming production difficulties that follow these disfluencies.

### 3.5 Correlations between features

We observe the maximum correlation from the feature correlations is between duration and surprisal. This positive correlation of 0.49 between surprisal and duration can be expected as higher information density for a word would take the speaker a longer duration to process. Given the performance of duration in disfluency prediction, this correlation could also explain the significant performance of disfluency prediction with surprisal. Further, recent work by Demberg et al. (2012) has shown how syntactic surprisal is a significant predictor of word duration. This is suggestive of the fact that surprisal can possibly be used to model language production, despite being an information-theoretic measure of comprehension difficulty. For further correlation values between features refer to Table 1 in Appendix A.

### 4 Discussion

Our results indicate that disfluencies occur when speaker has upcoming difficulties, as evinced from high storage cost and lexical surprisal at words following disfluencies. Speakers also seem to want to lower their cognitive load before disfluencies to help in planning, as suggested by low values of duration, storage cost, surprisal and integration cost on the preceding word. Ease of production is often attributed to ease of retrieval of words from memory. More accessible words (more salience, more predictability) are known to be easier to retrieve compared words with low accessibility. Since surprisal quantifies contextual predictability, the low surprisal prior to fillers indicate the ease of retrievability of words prior to fillers. Though words preceding reparandums do not show a lowering in surprisal and integration cost, it could be attributed to the fact that reparandums in itself consists of words, which may be the ones that hold low surprisal or integration costs, rather than the word preceding to the reparandum. This difference in the context of fillers and reparandums indicates the presence of distinct memory operations in language production.

DLT-based costs hitherto used to explain language comprehension gave the best increase in accuracy to 48.57\%, showing it has promise in explaining language production too. We did however note that the integration costs behaved in contrary directions, and so further detailed research is needed. The anomalous DLT effects need to be investigated more thoroughly in future work. In the comprehension literature Vasishth and Lewis (2006) proffer a unified explanation for both locality and anti-locality effects in Hindi verb-final constructions by resorting to either decay or interference (on account of similar intervening words) at a verbal head while integrating a previously encountered argument head. In a survey of dependency distance, Liu et al. (2017) state that long dependencies might not be difficult to process due to the presence of mitigating factors like frequency, contextual familiarity and positional salience.

Given that DLT costs bring about a large increase in accuracy, incorporating other syntax-based features like syntactic surprisal and UID (based on syntactic surprisal) might confer further insights on the role of syntax in disfluency prediction and language production. Despite seeing how DLT, duration and lexical surprisal behave individually, we have not as such compared these features against each other and studied if they account for explaining same parts of the data. We leave these steps for future work. Our modelling presupposes linear dependence between individual predictors. In future inquiries we plan to use other non-linear classifiers like decision trees and KNN models.

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*We thank the anonymous reviewers, Micha Elsner and Sidharth Ranjan for insightful comments and feedback.*

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### Table 4: Accuracy and weights for duration features

<table>
<thead>
<tr>
<th>Features</th>
<th>Accuracy</th>
<th>Fluent</th>
<th>Filler</th>
<th>Repar.</th>
</tr>
</thead>
<tbody>
<tr>
<td>Baseline</td>
<td>37.18%</td>
<td>-</td>
<td>-</td>
<td>-</td>
</tr>
<tr>
<td>Preceding duration**</td>
<td>46.82%</td>
<td>-2.9315</td>
<td>2.7944</td>
<td>0.1371</td>
</tr>
<tr>
<td>Following duration**</td>
<td>37.71%</td>
<td>-0.4512</td>
<td>0.0046</td>
<td>0.4466</td>
</tr>
<tr>
<td>Both durations**</td>
<td>46.89%</td>
<td>-</td>
<td>-</td>
<td>-</td>
</tr>
</tbody>
</table>

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We thank the anonymous reviewers, Micha Elsner and Sidharth Ranjan for insightful comments and feedback.
References


Marie-Catherine De Marneffe, Bill MacCartney, Christopher D. Manning, et al. Generating typed dependency parses from phrase structure parses.


Gating Mechanisms for Combining Character and Word-level Word Representations: An Empirical Study

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Abstract

In this paper we study how different ways of combining character and word-level representations affect the quality of both final word and sentence representations. We provide strong empirical evidence that modeling characters improves the learned representations at the word and sentence levels, and that doing so is particularly useful when representing less frequent words. We further show that a feature-wise sigmoid gating mechanism is a robust method for creating representations that encode semantic similarity, as it performed reasonably well in several word similarity datasets. Finally, our findings suggest that properly capturing semantic similarity at the word level does not consistently yield improved performance in downstream sentence-level tasks. Our code is available at https://github.com/jabalazs/gating.

1 Introduction

Incorporating sub-word structures like substrings, morphemes and characters to the creation of word representations significantly increases their quality as reflected both by intrinsic metrics and performance in a wide range of downstream tasks (Bojanowski et al., 2017; Luong and Manning, 2016; Wu et al., 2016; Ling et al., 2015).

The reason for this improvement is related to sub-word structures containing information that is usually ignored by standard word-level models. Indeed, when representing words as vectors extracted from a lookup table, semantically related words resulting from inflectional processes such as surf, surfing, and surfed, are treated as being independent from one another1. Further, word-level embeddings do not account for derivational processes resulting in syntactically-similar words with different meanings such as break, breakable, and unbreakable. This causes derived words, which are usually less frequent, to have lower-quality (or no) vector representations.

Previous works have successfully combined character-level and word-level word representations, obtaining overall better results than using only word-level representations. For example Luong and Manning (2016) achieved state-of-the-art results in a machine translation task by representing unknown words as a composition of their characters. Botha and Blunsom (2014) created word representations by adding the vector representations of the words’ surface forms and their morphemes (perfectly $= \text{perfectly} + \text{perfect} + \text{ly}$), obtaining significant improvements on intrinsic evaluation tasks, word similarity and machine translation. Lample et al. (2016) concatenated character-level and word-level representations for creating word representations, and then used them as input to their models for obtaining state-of-the-art results in Named Entity Recognition on several languages.

What these works have in common is that the models they describe first learn how to represent subword information, at character (Luong and Manning, 2016), morpheme (Botha and Blunsom, 2014), or substring (Bojanowski et al., 2017) levels, and then combine these learned representations at the word level. The incorporation of information at a finer-grained hierarchy results in higher-quality modeling of rare words, morphological processes, and semantics (Avraham and Goldberg, 2017).

There is no consensus, however, on which combination method works better in which case, or how the choice of a combination method affects downstream performance, either measured intrinsically at the word level, or extrinsically at the sen-

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1 Unless using pre-trained embeddings with a notion of subword information such as fastText (Bojanowski et al., 2017)
tence level.
In this paper we aim to provide some intuitions about how the choice of mechanism for combining character-level with word-level representations influences the quality of the final word representations, and the subsequent effect these have in the performance of downstream tasks. Our contributions are as follows:

- We show that a feature-wise sigmoidal gating mechanism is the best at combining representations at the character and word-level hierarchies, as measured by word similarity tasks.
- We provide evidence that this mechanism learns that to properly model increasingly infrequent words, it has to increasingly rely on character-level information.
- We finally show that despite the increased expressivity of word representations it offers, it has no clear effect in sentence representations, as measured by sentence evaluation tasks.

2 Background

We are interested in studying different ways of combining word representations, obtained from different hierarchies, into a single word representation. Specifically, we want to study how combining word representations (1) taken directly from a word embedding lookup table, and (2) obtained from a function over the characters composing them, affects the quality of the final word representations.

Let \( W \) be a set, or vocabulary, of words with \(|W|\) elements, and \( C \) a vocabulary of characters with \(|C|\) elements. Further, let \( x = w_1, \ldots, w_n \); \( w_i \in W \) be a sequence of words, and \( c^i = c^i_1, \ldots, c^i_m \); \( c^i_j \in C \) be the sequence of characters composing \( w_i \). Each token \( w_i \) can be represented as a vector \( v^{(w)}_i \in \mathbb{R}^d \) extracted directly from an embedding lookup table \( E^{(w)} \in \mathbb{R}^{|W| \times d} \), pre-trained or otherwise, and as a vector \( v^{(c)}_i \in \mathbb{R}^d \) built from the characters that compose it; in other words, \( v^{(c)}_i = f(c^i) \), where \( f \) is a function that maps a sequence of characters to a vector.

The methods for combining word and character-level representations we study, are of the form \( G(v^{(w)}_i, v^{(c)}_i) = v_i \) where \( v_i \) is the final word representation.

2.1 Mapping Characters to Character-level Word Representations

The function \( f \) is composed of an embedding layer, an optional context function, and an aggregation function.

The embedding layer transforms each character \( c^i_j \) into a vector \( r^i_j \) of dimension \( d_r \), by directly taking it from a trainable embedding lookup table \( E^{(c)} \in \mathbb{R}^{|C| \times d_r} \). We define the matrix representation of word \( w_i \) as \( C^i = [r^i_1, \ldots, r^i_m] \), \( C^i \in \mathbb{R}^{m \times d_r} \).

The context function takes \( C^i \) as input and returns a context-enriched matrix representation \( H^i = [h^i_1, \ldots, h^i_m] \), \( H^i \in \mathbb{R}^{m \times d_h} \), in which each \( h^i_j \) contains a measure of information about its context, and interactions with its neighbors. In particular, we chose to do this by feeding \( C^i \) to a Bidirectional LSTM (BiLSTM) (Graves and Schmidhuber, 2005; Graves et al., 2013)\(^2\).

Informally, we can think of a Long Short-Term Memory Network (LSTM) (Hochreiter and Schmidhuber, 1997) as a function \( \mathbb{R}^{m \times d_r} \rightarrow \mathbb{R}^{m \times d_h} \) that takes a matrix \( C = [r^i_1, \ldots, r^i_m] \) as input and returns a context-enriched matrix representation \( H = [h^i_1, \ldots, h^i_m] \), where each \( h^i_j \) encodes information about the previous elements \( h^i_1, \ldots, h^i_{j-1} \).\(^3\)

A BiLSTM is simply composed of 2 LSTMs, one that reads the input from left to right (forward), and another that does so from right to left (backward). The output of the forward and backward LSTMs are \( \vec{H} = [\vec{h}^i_1, \ldots, \vec{h}^i_m] \) and \( \hat{H} = [\hat{h}^i_1, \ldots, \hat{h}^i_m] \) respectively. In the backward case the LSTM reads \( r^i_m \) first and \( r^i_1 \) last, therefore \( \hat{h}^i_j \) will encode the context from \( \hat{h}^i_{j+1}, \ldots, \hat{h}^i_m \).

The aggregation function takes the context-enriched matrix representation of word \( w_i \) for both directions, \( \vec{H}^i \) and \( \hat{H}^i \), and returns a single vector \( v^{(c)}_i \in \mathbb{R}^{d_h} \). To do so we followed Miyamoto and Cho (2016), and defined the character-level representation \( v^{(c)}_i \) of word \( w_i \) as the linear combination of the forward and backward last hidden states re-

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\(^2\)Other methods for encoding the characters’ context, such as CNNs (Kim et al., 2016), could also be used.

\(^3\)In terms of implementation, the LSTM is applied iteratively to each element of the input sequence regardless of dimension \( m \), which means it accepts inputs of variable length, but we will use this notation for the sake of simplicity.

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course by the context function:

\[ v_i^{(c)} = W^{(c)} [h_i^c; h_i^w] + b^{(c)} \]  

where \( W^{(c)} \in \mathbb{R}^{d_{h} \times 2d_{h}} \) and \( b^{(c)} \in \mathbb{R}^{d_{h}} \) are trainable parameters, and \([c; c]\) represents the concatenation operation between two vectors.

### 2.2 Combining Character and Word-level Representations

We tested three different methods for combining \( v_i^{(c)} \) with \( v_i^{(w)} \): simple concatenation, a learned scalar gate (Miyamoto and Cho, 2016), and a learned vector gate (also referred to as feature-wise sigmoidal gate). Additionally, we compared these methods to two baselines: using pre-trained word vectors only, and using character-only features for representing words. See fig. 1 for a visual description of the proposed methods.

**word-only (w)** considers only \( v_i^{(w)} \) and ignores \( v_i^{(c)} \):

\[ v_i = v_i^{(w)} \]  

**char-only (c)** considers only \( v_i^{(c)} \) and ignores \( v_i^{(w)} \):

\[ v_i = v_i^{(c)} \]  

**concat (cat)** concatenates both word and character-level representations:

\[ v_i = [v_i^{(c)}; v_i^{(w)}] \]  

**scalar gate (sg)** implements the scalar gating mechanism described by Miyamoto and Cho (2016):

\[ g_i = \sigma (w^T v_i^{(w)} + b) \]  

\[ v_i = g_i v_i^{(c)} + (1 - g_i) v_i^{(w)} \]  

where \( w \in \mathbb{R}^d \) and \( b \in \mathbb{R} \) are trainable parameters, \( g_i \in (0, 1) \), and \( \sigma \) is the sigmoid function.

**vector gate (vg)**:

\[ g_i = \sigma (W v_i^{(w)} + b) \]  

\[ v_i = g_i \odot v_i^{(c)} + (1 - g_i) \odot v_i^{(w)} \]  

where \( W \in \mathbb{R}^{d \times d} \) and \( b \in \mathbb{R}^d \) are trainable parameters, \( g_i \in (0, 1)^d \), \( \sigma \) is the element-wise sigmoid function, \( \odot \) is the element-wise product for vectors, and \( 1 \in \mathbb{R}^d \) is a vector of ones.

The vector gate is inspired by Miyamoto and Cho (2016) and Yang et al. (2017), but is different to the former in that the gating mechanism acts upon each dimension of the word and character-level vectors, and different to the latter in that it does not rely on external sources of information for calculating the gating mechanism.

Finally, note that **word only** and **char only** are special cases of both gating mechanisms: \( g_i = 0 \) (scalar gate) and \( g_i = 0 \) (vector gate) correspond to **word only**; \( g_i = 1 \) and \( g_i = 1 \) correspond to **char only**.

### 2.3 Obtaining Sentence Representations

To enable sentence-level classification we need to obtain a sentence representation from the word vectors \( v_i \). We achieved this by using a BiLSTM with max pooling, which was shown to be a good universal sentence encoding mechanism (Conneau et al., 2017).

Let \( x = w_1, \ldots, w_n \), be an input sentence and \( V = [v_1, \ldots, v_n] \) its matrix representation, where each \( v_i \) was obtained by one of the methods described in section 2.2. \( S = [s_1, \ldots, s_n] \) is the
context-enriched matrix representation of \( x \) obtained by feeding \( V \) to a BiLSTM of output dimension \( d_s \). Lastly, \( s \in \mathbb{R}^{d_s} \) is the final sentence representation of \( x \) obtained by max-pooling \( S \) along the sequence dimension.

Finally, we initialized the word representations \( v_i^{(w)} \) using GloVe embeddings (Pennington et al., 2014), and fine-tuned them during training. Refer to appendix A for details on the other hyperparameters we used.

3 Experiments

3.1 Experimental Setup

We trained our models for solving the Natural Language Inference (NLI) task in two datasets, SNLI (Bowman et al., 2015) and MultiNLI (Williams et al., 2018), and validated them in each corresponding development set (including the matched and mismatched development sets of MultiNLI).

For each dataset-method combination we trained 7 models initialized with different random seeds, and saved each when it reached its best validation accuracy\(^5\). We then evaluated the quality of each trained model’s word representations \( v_i \) in 10 word similarity tasks, using the system created by Jastrzkebski et al. (2017)\(^6\).

Finally, we fed these obtained word vectors to a BiLSTM with max-pooling and evaluated the final sentence representations in 11 downstream transfer tasks (Conneau et al., 2017; Subramanian et al., 2018).

3.2 Datasets

Word-level Semantic Similarity A desirable property of vector representations of words is that semantically similar words should have similar vector representations. Assessing whether a set of word representations possesses this quality is referred to as the semantic similarity task. This is the most widely-used evaluation method for evaluating word representations, despite its shortcomings (Faruqui et al., 2016).

This task consists of comparing the similarity between word vectors measured by a distance

\[ s_i = \frac{\langle s_i^w; \bar{s}_i \rangle}{\|s_i^w\| \|\bar{s}_i\|} \text{ for each } i, \text{ and both } s_i^w \text{ and } \bar{s}_i \in \mathbb{R}^{d_w}. \]

\(^5\)We found that models validated on the matched development set of MultiNLI, rather than the mismatched, yielded best results, although the differences were not statistically significant.

\(^6\)https://github.com/facebookresearch/SentEval/tree/906b34a

4 Word-level Evaluation

4.1 Word Similarity

Table 1 shows the quality of word representations in terms of the correlation between word similarity metric (usually cosine distance), with a similarity score obtained from human judgements. High correlation between these similarities is an indicator of good performance.

A problem with this formulation though, is that the definition of “similarity” often confounds the meaning of both similarity and relatedness. For example, \( \text{cup} \) and \( \text{tea} \) are related but dissimilar words, and this type of distinction is not always clear (Agirre et al., 2009; Hill et al., 2015).

To face the previous problem, we tested our methods in a wide variety of datasets, including some that explicitly model relatedness (WS353R), some that explicitly consider similarity (WS353S, SimLex999, SimVerb3500), and some where the distinction is not clear (MEN, MTurk287, MTurk771, RG, WS353). We also included the RareWords (RW) dataset for evaluating the quality of rare word representations. See appendix B for a more complete description of the datasets we used.

Sentence-level Evaluation Tasks Unlike word-level evaluations, there is no consensus on the desirable properties sentence representations should have. In response to this, Conneau et al. (2017) created SentEval\(^7\), a sentence representation evaluation benchmark designed for assessing how well sentence representations perform in various downstream tasks (Conneau and Kiela, 2018).

Some of the datasets included in SentEval correspond to sentiment classification (CR, MPQA, MR, SST2, and SST5), subjectivity classification (SUBJ), question-type classification (TREC), recognizing textual entailment (SICK E), estimating semantic relatedness (SICK R), and measuring textual semantic similarity (STS16, STSB). The datasets are described by Conneau et al. (2017), and we provide pointers to their original sources in the appendix table B.2.

To evaluate these sentence representations SentEval trained a linear model on top of them, and evaluated their performance in the validation sets accompanying each dataset. The only exception was the STS16 task, in which our representations were evaluated directly.

\(^7\)https://github.com/facebookresearch/SentEval/tree/50Gb34a
scores obtained by the proposed models and word similarity scores defined by humans.

First, we can see that for each task, character only models had significantly worse performance than every other model trained on the same dataset. The most likely explanation for this is that these models are the only ones that need to learn word representations from scratch, since they have no access to the global semantic knowledge encoded by the GloVe embeddings.

Further, bold results show the overall trend that vector gates outperformed the other methods regardless of training dataset. This implies that learning how to combine character and word-level representations at the dimension level produces word vector representations that capture a notion of word similarity and relatedness that is closer to that of humans.

Additionally, results from the MNLI row in general, and underlined results in particular, show that training on MultiNLI produces word representations better at capturing word similarity. This is probably due to MultiNLI data being richer than that of SNLI. Indeed, MultiNLI data was gathered from various sources (novels, reports, letters, and telephone conversations, among others), rather than the single image captions dataset from which SNLI was created.

Exceptions to the previous rule are models evaluated in MEN and RW. The former case can be explained by the MEN dataset\(^8\) containing only words that appear as image labels in the ESP-Game\(^9\) and MIRFLICKR-1M\(^{10}\) image datasets (Bruni et al., 2014), and therefore having data that is more closely distributed to SNLI than to MultiNLI.

More notably, in the RareWords dataset (Luong et al., 2013), the word only, concat, and scalar gate methods performed equally, despite having been trained in different datasets \((p > 0.1)\), and the char only method performed significantly worse when trained in MultiNLI. The vector gate, however, performed significantly better than its counterpart trained in SNLI. These facts provide evidence that this method is capable of capturing linguistic phenomena that the other methods are unable to model.

### 4.2 Word Frequencies and Gating Values

Figure 2 shows that for more common words the vector gate mechanism tends to favor only a few dimensions while keeping a low average gating value across dimensions. On the other hand, values are greater and more homogeneous across dimensions in rarer words. Further, fig. 3 shows this mechanism assigns, on average, a greater gating value to less frequent words, confirming the findings by Miyamoto and Cho (2016), and Yang et al. (2017).

In other words, the less frequent the word, the more this mechanism allows the character-level representation to influence the final word representation, as shown by eq. (8). A possible interpretation of this result is that exploiting charac-

\(^8\)\url{https://staff.fni.uva.nl/e.bruni/MEN}

\(^9\)\url{http://www.cs.cmu.edu/~biglou/resources/}

\(^{10}\)\url{http://press.liacs.nl/mirflickr/}

<table>
<thead>
<tr>
<th></th>
<th>MEN</th>
<th>MTurk287</th>
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Table 1: Word-level evaluation results. Each value corresponds to average Pearson correlation of 7 identical models initialized with different random seeds. Correlations were scaled to the \([-100; 100]\) range for easier reading. Bold values represent the best method per training dataset, per task; underlined values represent the best-performing method per task, independent of training dataset. For each task and dataset, every best-performing model was trained in SNLI. These facts provide evidence that this method is capable of capturing linguistic phenomena that the other methods are unable to model.
ter information becomes increasingly necessary as word-level representations’ quality decrease.

Another observable trend in both figures is that gating values tend to be low on average. Indeed, it is possible to see in fig. 3 that the average gating values range from 0.26 to 0.56. This result corroborates the findings by Miyamoto and Cho (2016), stating that setting \( g = 0.25 \) in eq. (6), was better than setting it to higher values.

In summary, the gating mechanisms learn how to compensate the lack of expressivity of underrepresented words by selectively combining their representations with those of characters.

5 Sentence-level Evaluation

Table 2 shows the impact that different methods for combining character and word-level word representations have in the quality of the sentence representations produced by our models.

We can observe the same trend mentioned in section 4.1, and highlighted by the difference between bold values, that models trained in MultiNLI performed better than those trained in SNLI at a statistically significant level, confirming the findings of Conneau et al. (2017). In other words, training sentence encoders on MultiNLI yields more general sentence representations than doing so on SNLI.

The two exceptions to the previous trend, SICKE and SICKR, benefited more from models trained on SNLI. We hypothesize this is again due to both SNLI and SICK (Marelli et al., 2014) having similar data distributions\(^1\).

Additionally, there was no method that significantly outperformed the word only baseline in classification tasks. This means that the added expressivity offered by explicitly modeling characters, be it through concatenation or gating, was not significantly better than simply fine-tuning the pre-trained GloVe embeddings for this type of task. We hypothesize this is due to the conflation of two effects. First, the fact that morphological processes might not encode important information for solving these tasks; and second, that SNLI and MultiNLI belong to domains that are too dissimilar to the domains in which the sentence representations are being tested.

On the other hand, the vector gate significantly outperformed every other method in the STSB task when trained in both datasets, and in the STS16 task when trained in SNLI. This again hints at this method being capable of modeling phenomena at the word level, resulting in improved semantic representations at the sentence level.

6 Relationship Between Word- and Sentence-level Evaluation Tasks

It is clear that the better performance the vector gate had in word similarity tasks did not trans-
late into overall better performance in downstream tasks. This confirms previous findings indicating that intrinsic word evaluation metrics are not good predictors of downstream performance (Tsvetkov et al., 2015; Chiu et al., 2016; Faruqui et al., 2016; Gladkova and Drozd, 2016).

Figure 4(b) shows that the word representations created by the vector gate trained in MultiNLI had positively-correlated results within several word-similarity tasks. This hints at the generality of the word representations created by this method when modeling similarity and relatedness.

However, the same cannot be said about sentence-level evaluation performance; there is no clear correlation between word-similarity tasks and sentence-evaluation tasks. This is clearly illustrated by performance in the STSBenchmark, the only in which the vector gate was significantly superior, not being correlated with performance in any word-similarity dataset. This can be interpreted simply as word-level representations capturing word-similarity not being a sufficient condition for good performance in sentence-level tasks.

In general, fig. 4 shows that there are no general correlation effects spanning both training datasets and combination mechanisms. For example, fig. 4(a) shows that, for both word-only and concat models trained in SNLI, performance in word similarity tasks correlates positively with performance in most sentence evaluation tasks, however, this does not happen as clearly for the same models trained in MultiNLI (fig. 4(b)).

7 Related Work

7.1 Gating Mechanisms for Combining Characters and Word Representations

To the best of our knowledge, there are only two recent works that specifically study how to combine word and subword-level vector representations.

Miyamoto and Cho (2016) propose to use a trainable scalar gating mechanism capable of learning a weighting scheme for combining character-level and word-level representations. They compared their proposed method to manually weighting both levels; using characters only; words only; or their concatenation. They found that in some datasets a specific manual weighting scheme performed better, while in others the learned scalar gate did.

Yang et al. (2017) further expand the gating concept by making the mechanism work at a finer-grained level, learning how to weight each vector’s dimensions independently, conditioned on external word-level features such as part-of-speech and named-entity tags. Similarly, they compared their proposed mechanism to using words only, characters only, and a concatenation of both, with and without external features. They found that their vector gate performed better than the other methods in all the reported tasks, and beat the state of the art in two reading comprehension tasks.

Both works showed that the gating mechanisms assigned greater importance to character-level rep-

<table>
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resentations in rare words, and to word-level representations in common ones, reaffirming the previous findings that subword structures in general, and characters in particular, are beneficial for modeling uncommon words.

7.2 Sentence Representation Learning

The problem of representing sentences as fixed-length vectors has been widely studied. Zhao et al. (2015) suggested a self-adaptive hierarchical model that gradually composes words into intermediate phrase representations, and adaptively selects specific hierarchical levels for specific tasks. Kiros et al. (2015) proposed an encoder-decoder model trained by attempting to reconstruct the surrounding sentences of an encoded passage, in a fashion similar to Skip-gram (Mikolov et al., 2013). Hill et al. (2016) overcame the previous model’s need for ordered training sentences by using autoencoders for creating the sentence representations. Jernite et al. (2017) implemented a model simpler and faster to train than the previous two, while having competitive performance. Similar to Kiros et al. (2015), Gan et al. (2017) suggested predicting future sentences with a hierarchical CNN-LSTM encoder.

Conneau et al. (2017) trained several sentence encoding architectures on a combination of the SNLI and MultiNLI datasets, and showed that a BiLSTM with max-pooling was the best at producing highly transferable sentence representations. More recently, Subramanian et al. (2018) empirically showed that sentence representations created in a multi-task setting (Collobert and Weston, 2008), performed increasingly better the more tasks they were trained in. Zhang et al. (2018) proposed using an autoencoder that relies on multi-head self-attention over the concatenation of the max and mean pooled encoder outputs for producing sentence representations. Finally, Wieting and Kiela (2019) show that modern sentence embedding methods are not vastly superior to random methods.

The works mentioned so far usually evaluate the quality of the produced sentence representations in sentence-level downstream tasks. Common benchmarks grouping these kind of tasks include SentEval (Conneau and Kiela, 2018), and GLUE (Wang et al., 2019). Another trend, however, is to probe sentence representations to understand what linguistic phenomena they encode (Linzen et al., 2016; Adi et al., 2017; Conneau et al., 2018; Perone et al., 2018; Zhu et al., 2018).

7.3 General Feature-wise Transformations

Dumoulin et al. (2018) provide a review on feature-wise transformation methods, of which the mechanisms presented in this paper form a part of. In a few words, the $g$ parameter, in both scalar gate and vector gate mechanisms, can be understood as a scaling parameter limited to the $(0,1)$ range and conditioned on word representations, whereas adding the scaled $v_i^{(c)}$ and $v_i^{(w)}$ representations can be seen as biasing word representations conditioned on character representations.

The previous review extends the work by Perez et al. (2018), which describes the Feature-wise Linear Modulation (FiLM) framework as a generalization of Conditional Normalization methods, and apply it in visual reasoning tasks. Some of the reported findings are that, in general, scaling has greater impact than biasing, and that in a setting similar to the scalar gate, limiting the scaling parameter to $(0,1)$ hurt performance. Future decisions involving the design of mechanisms for combining character and word-level representations should be informed by these insights.
8 Conclusions

We presented an empirical study showing the effect that different ways of combining character and word representations has in word-level and sentence-level evaluation tasks.

We showed that a vector gate performed consistently better across a variety of word similarity and relatedness tasks. Additionally, despite showing inconsistent results in sentence evaluation tasks, it performed significantly better than the other methods in semantic similarity tasks.

We further showed through this mechanism, that learning character-level representations is always beneficial, and becomes increasingly so with less common words.

In the future it would be interesting to study how the choice of mechanism for combining subword and word representations affects the more recent language-model-based pretraining methods such as ELMo (Peters et al., 2018), GPT (Radford et al., 2018, 2019) and BERT (Devlin et al., 2018).

Acknowledgements

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References


Peter Young, Alice Lai, Micah Hodosh, and Julia Hockenmaier. 2014. From image descriptions to visual denotations. Transactions of the Association for Computational Linguistics, 2:67–78.


**A Hyperparameters**

We only considered words that appear at least twice, for each dataset. Those that appeared only once were considered UNK. We used the Treebank Word Tokenizer as implemented in NLTK\(^\text{12}\) for tokenizing the training and development datasets.

In the same fashion as Conneau et al. (2017), we used a batch size of 64, an SGD optimizer with an initial learning rate of 0.1, and at each epoch divided the learning rate by 5 if the validation accuracy decreased. We also used gradient clipping when gradients where > 5.

We defined character vector representations as 50-dimensional vectors randomly initialized by sampling from the uniform distribution in the \((-0.05; 0.05)\) range.

The output dimension of the character-level BiLSTM was 300 per direction, and remained of such size after combining forward and backward representations as depicted in eq. 1.

Word vector representations where initialized from the 300-dimensional GloVe vectors (Pennington et al., 2014), trained in 840B tokens from the Common Crawl\(^\text{13}\), and finetuned during training. Words not present in the GloVe vocabulary where randomly initialized by sampling from the uniform distribution in the \((-0.05; 0.05)\) range.

The input size of the word-level LSTM was 300 for every method except concat in which it was 600, and its output was always 2048 per direction, resulting in a 4096-dimensional sentence representation.

**B Datasets**

**B.1 Word Similarity**

Table B.1 lists the word-similarity datasets and their corresponding reference. As mentioned in section 3.2, all the word-similarity datasets contain pairs of words annotated with similarity or relatedness scores, although this difference is not always explicit. Below we provide some details for each.

**MEN** contains 3000 annotated word pairs with integer scores ranging from 0 to 50. Words correspond to image labels appearing in the ESP-Game\(^\text{14}\) and MIRFLICKR-1M\(^\text{15}\) image datasets.

**MTurk287** contains 287 annotated pairs with scores ranging from 1.0 to 5.0. It was created from words appearing in both DBpedia and in news articles from The New York Times.

\(^{12}\)http://www.nltk.org/

\(^{13}\)https://nlp.stanford.edu/projects/glove/

\(^{14}\)http://www.cs.cmu.edu/~biglou/resources/

\(^{15}\)http://press.liacs.nl/mirflickr/
Words include nouns, adjectives and verbs. of datasets that do not, such as MEN and WS353.

and not relatedness, addressing the shortcomings case the authors explicitly considered similarity with similarity scores ranging from 0 to 10. In this following classes: synonyms, antonyms, identical, classify each WS353 word pair into one of the

This dataset was created by asking humans to word pairs annotated with relatedness scores.

SimLex999 contains 999 word pairs annotated with similarity scores ranging from 0 to 10. In this case the authors explicitly considered similarity and not relatedness, addressing the shortcomings of datasets that do not, such as MEN and WS353. Words include nouns, adjectives and verbs.

SimVerb3500 contains 3500 verb pairs annotated with similarity scores ranging from 0 to 10. Verbs were obtained from the USF free association database (Nelson et al., 2004), and VerbNet (Kipper et al., 2008). This dataset was created to address the lack of representativity of verbs in SimLex999, and the fact that, at the time of creation, the best performing models had already surpassed inter-annotator agreement in verb similarity evaluation resources. Like SimLex999, this dataset also explicitly considers similarity as opposed to relatedness.

WS353 contains 353 word pairs annotated with similarity scores from 0 to 10.

WS353R is a subset of WS353 containing 252 word pairs annotated with relatedness scores. This dataset was created by asking humans to classify each WS353 word pair into one of the following classes: synonyms, antonyms, identical, hyperonym-hyponym, hyperonym-hyperonym, holonym-meronym, meronym-holonym, and none-of-the-above. These annotations were later used to group the pairs into: similar pairs (synonyms, antonyms, identical, hyperonym-hyponym, and hyponym-hyperonym), related pairs (holonym-meronym, meronym-holonym, and none-of-the-above with a human similarity score greater than 5), and unrelated pairs (classified as none-of-the-above with a similarity score less than or equal to 5). This dataset is composed by the union of related and unrelated pairs.

WS353S is another subset of WS353 containing 203 word pairs annotated with similarity scores. This dataset is composed by the union of similar and unrelated pairs, as described previously.

B.2 Sentence Evaluation Datasets

Table B.2 lists the sentence-level evaluation datasets used in this paper. The provided URLs correspond to the original sources, and not necessarily to the URLs where SentEval\textsuperscript{16} got the data from\textsuperscript{17}.

The version of the CR, MPQA, MR, and SUBJ datasets used in this paper were the ones preprocessed by Wang and Manning (2012)\textsuperscript{18}. Both SST2 and SST5 correspond to preprocessed versions of the Stanford Sentiment Treebank (SST) dataset by Socher et al. (2013)\textsuperscript{19}. SST2 corresponds to a subset of SST used by Arora et al. (2017) containing flat representations of sentences annotated with binary sentiment labels, and SST5 to another subset annotated with more fine-grained sentiment labels (very negative, negative, neutral, positive, very positive).

\begin{table}[h]
\centering
\begin{tabular}{|l|l|l|}
\hline
Dataset & Reference & URL \\
\hline
MEN & Bruni et al. (2014) & https://staff.fnwi.uva.nl/e.bruni/MEN \\
MTurk771 & Radinsky et al. (2011) & https://git.io/fhQAB (Unofficial) \\
RG & Rubenstein and Goodenough (1965) & https://git.io/fhQAB (Unofficial) \\
RareWords (RW) & Luong et al. (2013) & https://nlp.stanford.edu/˜lmthang/morphoNLM/ \\
SimVerb3500 & Gerz et al. (2016) & http://people.ds.cam.ac.uk/dsg40/simverb.html \\
\hline
\end{tabular}
\caption{Word similarity and relatedness datasets.}
\end{table}

\textsuperscript{16}https://github.com/facebookresearch/SentEval/tree/906b34a
\textsuperscript{17}A list of the data used by SentEval can be found in its data setup script: https://git.io/fhqpq
\textsuperscript{18}https://nlp.stanford.edu/~sidaw/home/projects:nbsem
\textsuperscript{19}https://nlp.stanford.edu/sentiment/
Table B.2: Sentence representation evaluation datasets. SST5 was obtained from a GitHub repository with no associated peer-reviewed work.

<table>
<thead>
<tr>
<th>Dataset</th>
<th>Reference</th>
<th>URL</th>
</tr>
</thead>
<tbody>
<tr>
<td>SST5</td>
<td>See caption.</td>
<td><a href="https://git.io/fhQAV">https://git.io/fhQAV</a></td>
</tr>
<tr>
<td>SICKR</td>
<td>Marelli et al. (2014)</td>
<td><a href="http://clic.cimec.unitn.it/composes/sick.html">http://clic.cimec.unitn.it/composes/sick.html</a></td>
</tr>
<tr>
<td>STS16</td>
<td>Agirre et al. (2016)</td>
<td><a href="http://ixa2.si.ehu.es/stswiki/index.php/Main_Page">http://ixa2.si.ehu.es/stswiki/index.php/Main_Page</a></td>
</tr>
<tr>
<td>STSb</td>
<td>Cer et al. (2017)</td>
<td><a href="http://ixa2.si.ehu.es/stswiki/index.php/STSbenchmark">http://ixa2.si.ehu.es/stswiki/index.php/STSbenchmark</a></td>
</tr>
</tbody>
</table>
A Pregroup Representation of Word Order Alternation using Hindi Syntax

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Abstract

Pregroup calculus has been used for the representation of free word order languages (Sanskrit and Hungarian), using a construction called precyclicity. However, restricted word order alternation has not been handled before. This paper aims at introducing and formally expressing three methods of representing word order alternation in the pregroup representation of any language. This paper describes the word order alternation patterns of Hindi, and creates a basic pregroup representation for the language. In doing so, the shortcoming of correct reductions for ungrammatical sentences due to the current apparatus is highlighted, and the aforementioned methods are invoked for a grammatically accurate representation of restricted word order alternation. The replicability of these methods is explained in the representation of adverbs and prepositional phrases in English.

1 Introduction

Categorial grammars are one of the frameworks for the representation of syntactic structures of languages (Oehrle et al., 2012). A foundational problem in such formalisms, including the well established lexical formalism combinatory categorial grammars (CCG), is the representation of free word order in light of syntactic or semantic constraints presented by the language. Extensive resources following the different formalisms, such as CCG Banks (Hockenmaier and Steedman, 2007), have been developed. Development in pregroup calculus, however, has been more focused on developing formal constraints in the calculus for the representation of syntactic phenomena. In that vein, this paper aims at presenting the problem of restricted word alternation (constituent scrambling) by using the example of Hindi syntax, and uses that to develop three formal approaches to represent word alternation in the pregroup calculus framework.

Pregroups are mathematical structures which were developed initially by Lambek and can be used to analyze sentences in English algebraically (Lambek, 1997). Pregroup calculus was a revision of Lambek’s previous categorial grammar called Syntactic Calculus (Lambek, 1958). Various languages have since adopted the use of pregroup calculus as the formal representation of syntax of a fragment or a particular property of the language, including Arabic (Bargelli and Lambek, 2001b), French (Bargelli and Lambek, 2001a), German (Lambek, 2000), Japanese (Cardinal, 2002), Persian (Sadrzadeh, 2007), Polish (Kislan-Malinowska, 2008), and Sanskrit (Casadio and Sadrzadeh, 2014), among others. Coecke et al. (2010)’s compositional distributional model of meaning also uses pregroup calculus as the compositional theory for grammatical types.

In the study of free word order languages (or languages with clitics), such as Italian (Casadio, 2010), Latin (Casadio and Lambek, 2005) and Hungarian (Sadrzadeh, 2011), a transformation was introduced, known as the precyclic transformation (section 3.1). This transformation was also used for Sanskrit (Casadio and Sadrzadeh, 2014), where generation of ungrammatical sentences was restricted by disallowing certain transformations.

In this paper, we first establish a preliminary pregroup grammar of Hindi and highlight the syntactic constraints of the language. We then show the shortcomings of the current word order alternation mechanism. We also formally define the
Table 1: Case/Role Marking in Hindi

<table>
<thead>
<tr>
<th>karaka</th>
<th>vibhakti</th>
<th>Equivalent Case</th>
</tr>
</thead>
<tbody>
<tr>
<td>karta</td>
<td>φ</td>
<td>Nominative</td>
</tr>
<tr>
<td></td>
<td>ne</td>
<td>Ergative</td>
</tr>
<tr>
<td>karam</td>
<td>ko</td>
<td>Accusative</td>
</tr>
<tr>
<td></td>
<td>dative</td>
<td>Dative</td>
</tr>
<tr>
<td>karan</td>
<td>se</td>
<td>Instrumental</td>
</tr>
<tr>
<td>sampradan</td>
<td>ko, ke liye</td>
<td>Purpose/Reason</td>
</tr>
<tr>
<td>apadaan</td>
<td>se</td>
<td>Source</td>
</tr>
<tr>
<td>adhikaran</td>
<td>me, par</td>
<td>Locative</td>
</tr>
</tbody>
</table>

The genitive case (called *sambandh*) in Hindi is not a *karaka*, as it shows the relationship of one noun to another noun. These are gender marked, to reflect the gender of the following noun. For instance the phrase "Neha’s ball" will be translated as *Neha kI gend*.

The verb phrase consists of either a single verb or a conjunct verb, complex verb or a light verb complex, which usually follows the construction "noun/adjective/verb + verbalizer" (Ahmed et al., 2012). The tense markers and aspect markers are separate lexical items, while the future marker is a suffix on the trailing verb or verbalizer. The verbalizer interactions can be further classified based on their behavior with aspect markers such as infinitive + forms of *lagnaa* (to begin) (Spencer et al., 2005; Chakrabarti et al., 2008).

### 2.1 Restricted Word Order Alternation

Hindi follows a general SOV word order (more specifically, S-IO-DO-V) (Seddah et al., 2010).

In the default word order, Hindi is a head-final with a relatively free word order (Patil et al., 2008). However, constituents are often "mixed up", which may be done for focus or emphasis, but this is not always the case (Butt and King, 1996; Kidwai, 2000).

We take an example of the sentence: *raam ne sitaa ko kitaab dii* Ram erg. Sita dat. book gave-fem.

"Ram gave the book to Sita" to explain the possible word orders (Ambati et al., 2010).

- Default word order (S-IO-DO-V) used above
- S-DO-IO-V: *raam ne kitaab sitaa ko dii*
- IO-S-DO-V: *sitaa ko raam ne kitaab dii*
- IO-DO-S-V: *sitaa ko raam ne kitaab dii*
- DO-S-IO-V: *kitaab raam ne sitaa ko dii*
- DO-IO-S-V: *kitaab sitaa ko raam ne dii*

Note that due to its isolating nature, the word order in the constituents remains intact, which is difficult to represent. The pregroup representation should allow only for valid constituent alternation while keeping the word order in the constituents in agreement with the grammar. The example provided in Section 4.3 explains this in detail.

### 3 Pregroup Calculus and Precyclicity

In this section, we define pregroups and pregroup grammars. We also explore the mathematical apparatus required to define word order change, based upon the concept of precyclicity.

A pregroup is a partially ordered monoid \((P, \cdot, 1, \to, \langle -, \rangle^l, \langle -, \rangle^r)\), which has two unary operators, the left and the right adjoint such that \(\forall x \in P: \langle x \rangle^l \cdot x \to 1 \to x \cdot \langle x \rangle^r\).  

\(^2S = \text{subject}, O = \text{object}, IO = \text{indirect object}, DO = \text{direct object}, V = \text{verb}\)

\(^3A\) monoid is a set closed under an associative binary operation. Partial order indicates that the binary relation has to be reflexive, antisymmetric and transitive.
where 1 is the identity element, the · operator is a concatenation operator (usually not explicitly mentioned) and the → operator indicates partial order.

Some other properties of pregroups include:

\[ 1^l = 1 = 1^r \]
\[ (a \cdot b)^l = b^l \cdot a^l \quad (a \cdot b)^r = b^r \cdot a^r \]
\[ (a^l)^l = a^{ll} \quad (a^r)^r = a^{rr} \]
\[ a^{rl} = a = a^{lr} \]

Adjoints are switching in nature, which means that:

\[ a \rightarrow b \iff b^l \rightarrow a^l \quad a \rightarrow b \iff b^r \rightarrow a^r \]

The operations \( x^l \cdot x \rightarrow 1 \) and \( x \cdot x^r \rightarrow 1 \) are called contractions and the operations \( 1 \rightarrow x \cdot x^l \) and \( 1 \rightarrow x^r \cdot x \) are called expansions. The monoids used for pregroup grammars are free pregroups, which have the property that without loss of generality, contractions precede expansions. This is called the switching lemma (Lambek, 1999).

Pregroup calculus defines a basic type as an element \( a \in P \). A simple type can be obtained by basic types as:

\[ a^{ll}, a^l, a, a^r, a^{rr}, a^{rrr} \]

A compound type is a concatenation of simple types. Pregroup grammars, much like other categorial grammars, assigns compound types to words in a sentence. A sentence is considered grammatical if it reduces (by concatenation with adjoints) to the simple type of the main verb in the sentence.

Therefore, in English a simple transitive verb will be represented as follows:

subject \quad verb \quad object \quad \ll
\[ \eta \quad p^r \quad s \quad d \quad o \quad \rightarrow s \]

The reductions are shown by the arcs, and the sentence reduces to type \( s \), the type of the main verb.

A pregroup grammar is a quintuple \( G = (\Sigma, P, \rightarrow, s, I) \) such that \( \Sigma \) is a nonempty, finite alphabet, \( (P, \rightarrow) \) is a finite poset, \( s \in P \), and \( I \) is a finite relation between symbols from \( \Sigma \) and non-empty types (on \( P \)) (Buszkowski, 2001). Contemporary literature symbolizes the set of all basic types as \( B \), and the set of all compound types as \( T(B) \). \( B \) is a partially ordered set, while \( T(B) \) is a free, proper pregroup over the set \( B \).

### 3.1 Precyclicity in Pregroups

A detailed understanding of cyclic properties of a pregroup can be derived from Lambek’s syntactic calculus, and using a translation between residuated monoids (the structure used in syntactic calculus) and pregroups (Casadio and Lambek, 2002). Since pregroups used for language formalism are free, proper pregroups (Buszkowski, 2001), the classical definition of cyclicity \( a \cdot b \rightarrow c \Rightarrow b \cdot a \rightarrow c \) does not hold. However, a weak form of cyclicity, called precyclicity, is admitted, which has the following properties (Yetter, 1990):

\[ pq \rightarrow r \Rightarrow q \rightarrow p^r \; r \] (1)
\[ q \rightarrow rp \Rightarrow qp^r \rightarrow r \] (2)
\[ pq \rightarrow r \Rightarrow q \rightarrow rp^l \] (3)
\[ q \rightarrow pr \Rightarrow p^l q \rightarrow r \] (4)

Due to this, we obtain the following rules for precyclicity with double adjoints (Abrusci, 1991):

\[ 1 \rightarrow ab \ll \Rightarrow 1 \rightarrow ba^{ll} \] (5)
\[ 1 \rightarrow ab \rr \Rightarrow 1 \rightarrow b^{rr} a \] (6)

Here, \( ba^{ll} \) and \( b^{rr} a \) are known as the precyclic permutations of \( ab \). Given these precyclic permutations, for \( A, B, C \in P \), the following precyclic transformations are defined:

\[ (ll) - \text{transformation} \]
\[ A \rightarrow B(ab)C \sim_{ll} A \rightarrow B(ba^{ll})C \] (7)

\[ (rr) - \text{transformation} \]
\[ A \rightarrow B(ab)C \sim_{rr} A \rightarrow B(b^{rr} a)C \] (8)

These precyclic transformations provide the following two equations,

\[ p^r q \leq qp^l \] (9)
\[ qp^l \leq p^r q \] (10)

which can be seen to be empirically derived in clitic movement patterns in other languages, explored in Casadio and Sadrzadeh (2009).
4 A Preliminary Pregroup Grammar for Hindi

This section identifies the basic types and compound types used in Hindi. The basic types are pregroup representations of syntactic features discussed in Section 2.

4.1 Basic Types

The basic types will be similar to the set of basic types chosen for English, \( \{\pi, o, p, n, s\} \), which are personal pronouns, the direct object of a transitive verb, simple predicate, noun phrase, and sentence respectively (Lambek, 2004). The basic types for the lexicalized case markers as well as the tense and aspect markers have to be included, explained below.

The basic type \( \kappa_i \) and \( \rho \) represent \textit{karaka} and \textit{sambandhi} markers (for the genitive case) respectively. The subscript on \( \kappa \) denotes the case of the noun that precedes it (refer Table 2). A simple example of the genitive case interaction, for the noun phrase "Ram’s brother" (\textit{raam kaa bhai}), is as:

\[
(n\kappa_1)(n\pi\rho)(\rho s\kappa l) \rightarrow n
\]

The type assignment seems to imply that in the genitive case marker is the headword of the noun phrase \textit{raam kaa}, which is not the case. Genitives in Hindi are gender marked, and they agree with the gender of the following noun phrase, which is preserved by the current type assignment. The type assignment seems to reflect the Paninian framework of modifier-modified relationship, in which the genitive case marker reflects the modification of the following noun phrase (Bharati and Sangal, 1993).

The VP consists of a verb, an optional auxiliary (denoted by \( \alpha \)) and a tense marker (denoted by \( \tau \)), in that order of occurrence. Verb transitivity does affect the verb typing, but only in the sentence. All statement verbs are given the type \( s \). Verbs in Hindi are unique for their tense marking system, which include a gender-marked past tense marker, a gender-neutral present tense marker, and a suffixed future tense marker.

Given the semantic nature of \textit{karaka} markers in Hindi, verb transitivity raises ambiguous cases. For example, the sentence \textit{Ram ne seb ko khaaya} and \textit{Ram ne seb khaaya} both translate to "Ram ate an apple". In this analysis, there are three distinct cases of the use of \textit{karaka} intransitive verbs: (1) no case markers, (2) ergative case marker on the subject and (3) ergative case marker on the object.

Given the set of basic types \( \{\pi, s, p, o, n, \kappa_1, \rho, \alpha, \tau\} \), simple sentences can be typed in Hindi as follows. The toy examples chosen here are similar in to a few of the simple sentences of the Hindi treebank (Bhatt et al., 2009).

1. I go to school.

\[
(\pi)(\rho r s^{\tau l})(\tau) \rightarrow s
\]

1. Tina sang a song.

\[
(n\kappa_1^l)(\kappa_1)(\rho o o)(\rho o o o) \rightarrow s
\]

3. Ram had hit the ball with a bat.

\[
(n\kappa_1^l)(\kappa_1)(\rho o o o)(\rho o o o o) \rightarrow s
\]

Table 2: Type given to \textit{karaka}
4.3 Consequences of Restricted Word Order

Alternation

In order to apply precyclicity rules, the example chosen is a simple transitive verb from the sentence "Tina sang a song", which has been typed above in section 4.2. As above, the arcs denote a reduction, while the underline shows the arguments of the precyclical transformation.

4. gaanaa tina ne gaayaa
song Tina erg. sang-masc.

\[
\begin{align*}
(o)(nk_1^1)(\kappa_1)(o^r n^r s) \\
\rightarrow (o)(n)(o^r n^r s) \\
\sim_{ll} (o)(n)(n^r o^r d\, s) \\
\rightarrow (o)(o^r s) \\
\sim_{rr} (o^r s)^{rr}(o) \\
\rightarrow (o^r d)^{rr}(o) \\
\sim_{ll} (s^r o^r d)(o) \\
\rightarrow s^r o^r d \\
\sim_{ll} s
\end{align*}
\]

An example of a sentence with two movements is as follows, for the sentence "Ram had hit the ball with a bat".

5. balle se gend ko
bat with ball acc.

raam ne maaraa thaa
Ram erg. hit-masc. was-masc.

\[
\begin{align*}
(pk_3^1)(\kappa_3)(ok_2^1)(\kappa_2)(nk_1^1)(\kappa_1)(p^r o^r n^r s) \\
\rightarrow (p)(o)(n)(p^r o^r n^r s) \\
\sim_{ll} (p)(o)(n)(p^r o^r n^r s) \\
\sim_{ll} (p)(o)(n)(p^r o^r n^r s) \\
\sim_{ll} (p)(o)(n)(p^r o^r n^r s) \\
\rightarrow (p)(o)(p^r o^r d\, s) \\
\sim_{rr} (p)(o)(p^r o^r d\, s) \\
\rightarrow (p)(o)(p^r o^r d\, s) \\
\sim_{ll} (s^r o^r d)(p) \\
\rightarrow s^r o^r d \\
\sim_{ll} s
\end{align*}
\]

Note that in the first example, the movement was gaanaa and tinaa ne, and not just tinaa; similarly, in the second sentence, the constituents being shuffled are balle se and gend ko. Hindi allows only constituent scrambling, the pregroup grammar has to account for this restriction. For example, the following construction should NOT be allowed:

\[
\begin{align*}
tinaa gaanaa ne gaayaa \\
(nk_1^1)(\kappa_1)(p^r o^r n^r s) \\
\sim_{ll} (nk_2^1)\sim_{ll} (nk_1^1)(p^r o^r n^r s) \\
\rightarrow (n)(p^r o^r n^r s) \\
\sim_{ll} (o)(n)(p^r o^r d\, s) \\
\rightarrow (o)(o^r s) \\
\sim_{ll} (s^r o^r d)(o) \\
\rightarrow s^r o^r d \\
\sim_{ll} s
\end{align*}
\]

5 Restricting Word Movement

As seen in section 4, the current pregroup grammar rules allow for the reduction of sentences which are disallowed by the grammar. This section explains the methods taken to restrict word movement, in order to allow only constituent scrambling, keeping the order of the words within a constituent constant.

5.1 Pregroup Grammar Rules

Pregroup grammar rules for restricting word order movement have been briefly discussed in (Casadio, 2004) and (Casadio and Sadrzadeh, 2014). Here, instead of treating restrictive rules as an exception, we treat it as a part of the pregroup framework, by creating a formal representation of these restrictions.

For example, a word order rule in Hindi syntax is "The alternation of ANY phrase with a karaka marker is disallowed" (Bopche et al., 2012), it may be represented in the following way:

\[
\forall x \in \mathcal{B}
\]

\[
(x)(\kappa_i) \sim_{ll} (\kappa_i)(x^{ll})
\]

\[
(x)(\kappa_i) \sim_{rr} (\kappa_i^{rr})(x)
\]

where, as discussed in Section 3, \( \mathcal{B} \) is the set of all basic types in the language. A similar set of
rules can be created for alternation in verb constituents, where no word order alternation is allowed between a verb and its aspect marker, and between the aspect and the tense marker, mirroring the rules of the language.

\[
\begin{align*}
(s)(\tau) \not\sim_{II} (\tau)(s) & \quad (s)(\tau) \not\sim_{rr} (\tau^r)(s) \\
(\tau)(\alpha) \not\sim_{II} (\alpha)(\tau) & \quad (\tau)(\alpha) \not\sim_{rr} (\alpha^r)(\tau) \\
(s)(\alpha) \not\sim_{II} (\alpha)(s) & \quad (s)(\alpha) \not\sim_{rr} (\alpha^r)(s)
\end{align*}
\]

Therefore, in the example \textit{tinaa gaanaa ne gaayaa} (Tina song erg. sang) presented above, we have:

\[
(nk^1_s(o)(\kappa_1))(o^r n^r s) \not\sim_{II} (nk^1_s)(o)(\kappa_1)(o^r n^r s)
\]

Thus the sentence will not reduce to \(s\) as it is deemed ungrammatical.

### 5.2 Selective Transformation

Selective transformation is a procedure that disallows the precyclic transformation of a step in the reduction, provided that some elements of the pregroup representation belong to a set that does not allow precyclic transformation, hence allowing only a selective application of the precyclic conversion rules.

As discussed in section 3, \(B\) is the set of all basic types in the language, and \(T(B)\) is the free pregroup over that set. In order to apply selective transformation, two sets \(\mathcal{B}_T\) and \(\mathcal{B}_{NT}\) are defined, such that the set of all basic types \(B = \mathcal{B}_T \cup \mathcal{B}_{NT}\), where \(T(\mathcal{B}_{NT})\) is a free, proper, non-precyclic pregroup, while \(T(\mathcal{B}_T)\) is a proper, free, precyclic pregroup. The union of a precyclic and a non-precyclic pregroup is a non-precyclic pregroup, and no other properties of the pregroup are affected (Refer to the Appendix for the proof).

Within the examples seen above, the basic types of the karaka marker and sambandh marker in the noun phrase, and the tense and aspect marker in the verb phrase should not \(ll\)- or \(rr\)-transformed, while the personal pronoun, sentence, predicate, noun phrase, and direct object types are allowed to transform. Therefore, \(\{\kappa_i, \rho, \alpha, \tau\} \in \mathcal{B}_{NT}\) and \(\{\pi, s, p, o, n\} \in \mathcal{B}_T\). Therefore, to apply transformation rules, only the types in the sentence should be those belonging in \(\mathcal{B}_T\). Therefore, in the example \textit{tinaa ne gaanaa gaayaa} (Tina erg. song sang), we have:

\[
(o)(n^1_k(o)(\kappa_1))(o^r n^r s) \rightarrow (o)(n)(o^r n^r s)
\]

where all the elements in the second line belong to \(\mathcal{B}_T\), which allows the transformations and reductions:

\[
\begin{align*}
(\sim)_{II} (o)(n^1)(o^r)(s) & \rightarrow (o)(o)(s) \\
(\sim)_{rr} (o^r)(s)(o) & \rightarrow (o)(s) \\
(\sim)_{II} (s^r o)(o) & \rightarrow s^r o \\
& \sim_{II} s
\end{align*}
\]

while in the example of the construction \textit{tinaa gaanaa ne gaayaa} (Tina song erg. sang) presented above, we have:

\[
(nk^1_s(o)(\kappa_1))(o^r n^r s)
\]

which is not reducible as the transformations cannot be applied. Therefore ungrammatical reductions are disallowed.\(^4\)

### 5.3 Two-Step Reduction

As mentioned above, Hindi allows constituent scrambling as opposed to word order scrambling. Therefore the reduction of a sentence can be deconstructed into two steps, the reduction of constituents, followed by reduction of the sentence. This process guarantees that ungrammatical reductions will not take place.

For the two-step reductions, first the "constituent profile" has to be defined. A constituent profile is a general construction to which constituents in Hindi can be mapped. Each constituent profile has a reduction which can be applied to a sentence. A constituent profile is specific to the type of phrase expected. A sequence of words which does not follow any constituent profile cannot be reduced. This will disallow the reduction of ungrammatical sentences.

Table 3 shows the constituent profiles for the examples provided above. The \([x]^+\) represents one or more elements of the basic type \(x\). Note that there is no need for a transformation in any of the constituent profiles, as it reduces to the constituent head form automatically. The constituents can be nested, and the constituent profile reflects this. A sentence may be defined as a specific case of a constituent profile, as it is the only profile which allows transformations before reductions.

\(^4\)Refer to Appendix for mathematical correctness of selective transformation.
Two-step reduction can be achieved as follows:

- **Type the sentence**: This allows the recognition of all possible constituent profiles, as well as conflicts with the constituent profiles.

- **Isolate and reduce constituents**: The constituents are mapped to the constituent profiles, and reduced accordingly.

- **Replace reduced forms in the sentence**: The constituents, once reduced, are placed back into the order in which they occurred in the original sentence. The sentence form should resemble the "sentence" profile.

- **Transform and reduce**: The sentence is then transformed and reduced according to the rules of transformation, as has been done above.

An example of two-step reduction for the sentence *raam ne gend ko balle se maara thaa* (Ram hit the ball with a bat) would be as follows:

**Step 1**: 

\[(nk^1_1)(k_1) → n, (ok^2_2)(k_2) → o, (pk^3_3)(k_3) → p\]

**Step 2**: 

\[(nk^1_1)(k_1) \rightarrow n\], according to the subject constituent profile. Similarly, \((ok^2_2)(k_2)\) → \(o\) and \((pk^3_3)(k_3)\) → \(p\) are also valid reductions according to the object and predicate profiles. There is also the \(([x]^+n^r s^t)(τ) \rightarrow ([x]^+n^r s)\), which is a valid verb form reduction.

**Step 3**: 

\[(n)(o)(p)(p^r o^r n^r s)\] is obtained, which fits the sentence profile.

**Step 4**: 

\[(n)(o)(p)(p^r o^r n^r s) \rightarrow s\], which is the required reduction.

An incorrect reduction can be recognized easily. Given the example that was provided in Section 4.2 tinaa gaanaa ne gaayaa. After step 1, the following is obtained:

\[(nk^1_1)(o)(k_1) \rightarrow (o^r n^r s)\]

which cannot be found in any constituent profile. Therefore, the sentence is ungrammatical, and will appropriately not be represented by the grammar.

### 6 Word Order Alternation in English

Lambek’s work in English type grammar (Lambek, 2004) and the work that has followed (Preller, 2007; Stabler, 2008) have not dealt with word order alternation in English yet. While English is a relatively fixed word order language, note that prepositional phrases and adverbial phrases are relatively free, especially with intransitive verbs. Therefore, while the default order remains "Subject-Adverb-Verb-PP", it can be seen in the following sentences that this word order can also change.

- Default (S-Adv-V-PP): "I quickly ran into the fields."
- PP-S-Adv-V: "Into the fields, I ran quickly."
- Adv-S-PP-PP: "Quickly I ran into the fields."
- S-V-Adv-PP: "I ran quickly into the fields."
- S-V-PP-Adv: "I ran into the fields quickly."
To develop a robust representation of English word order, first, the initial word order constraints must be noted. The standard word order in English is SVO, which reduces to SV in the case of intransitive verbs, which is the focus of this sample. We examine the word order alternation in this fragment of English, using the methods explored in the paper.

First, the set of basic types \{\pi, o, s, n, p\} has to be expanded to include types for adverbial phrase \(A\) and type for prepositional phrase \(\rho\). The subscripts which characterize gender, number, person or tense have been ignored for this example. The determiner is considered a part of the noun phrase. Therefore, the sample sentence, in the default word order, can be typed as follows ("the fields" has been typed \(p\)):

\[
\begin{align*}
\pi & \quad A \quad A' \pi' s \rho' \quad p \\
\end{align*}
\]

This reduction is reasonably straightforward. On changing the word order, precyclic transformations are applicable, so a similar reduction for the statement, "Quickly I ran into the fields", can be handled as follows:

\[
\begin{align*}
\pi & \quad A \quad A' \pi' s \rho' \quad p \\
\end{align*}
\]

Therefore, the methods discussed in section 5 can be used to disallow this reduction.

- Using the method of restriction of pregroup grammar rules, the transformation \((pp') \sim_{rr} (p' \rho')\) is disallowed. So the order of the prepositional phrase and its predicate remains the same, disallowing an ungrammatical reduction.

- Two-step reduction can be used to establish the prepositional phrase profile as \((pp')(p)\), and the sentence does not follow this profile. Therefore, further reduction is disallowed.

- Under selective transformation, the only basic type that can belong to \(B_{NT}\) is \(p\). Therefore, its alternation with the type \(\rho\) is considered ungrammatical and the sentence is not reduced.

Hence, all three methods can be used to disallow the representation and reduction of ungrammatical sentences.

## Conclusion

This paper is an attempt to formalize the syntactic notion of word order alternation using pregroup grammars by expanding on the developments in the current literature and proposing two novel methods of representing restrictions in word movement. The inspiration for this development stems from the syntactic structure of Hindi, an isolating free word order language which allows only for constituent scrambling, keeping the order of the words in the constituents the same.

In order to express the syntactic properties of Hindi, a preliminary pregroup grammar for Hindi has also been established. This pregroup grammar has been developed in a manner similar to their development in other languages, with the establishment of basic types and examples of sentences represented using the grammar. Further work can be done in the development of the pregroup grammar of Hindi, in modeling agreement rules and other aspects of the Hindi syntax.

The methods proposed include the expansion and formal representation of the method used for Sanskrit, as well as two novel approaches which are selective transformation and two-step reduction. Using examples in the paper, these methods have been applied to prove their effectiveness,
and an example has also been taken from English, which does the same.

Future work in this direction could include development of pregroup representations of common cross-lingual syntactic phenomena, such as agglutination, that would make the adaptation of pregroups to other languages easier. A more thorough grammar, based on other properties such as subject-verb agreement, verb complements and so on are other directions of developing the work.

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References


A Appendix: The Mathematics of Selective Transformation

The operations between the sets $T(B_T)$ and $T(B_{NT})$ that were implemented in the paper in Section 5.2 and the non-precyclic nature of $T(B)$ are explained here.

Precyclic transformation was introduced in Yetter (1990), for relaxing the conditions on the cut theorem on one sided sequent calculus for pure non-commutative classical linear propositional logic. The original rules of precyclic transformation:

$$\Gamma, A \vdash \Gamma, \updownarrow, A$$

$$\Gamma, \Gamma \vdash \Gamma, \Gamma$$

The calculus for which this rule was applicable was called SPNCL’, while the calculus where this rule was not applicable was SPNCL. In order to understand the affect of this theorem, note the cut theorem in SPNCL:

$$\Gamma, A, \Gamma \vdash \Delta_1, A \updownarrow, \Delta_2$$

$$\vdash \Delta_1, \Gamma, \Delta_2, \Gamma$$

$$\vdash \Gamma, A, \Gamma \vdash \Delta_1, A, \Delta_2$$

$$\vdash \Delta_1, \Gamma_1, \Delta_2, \Gamma_2$$

if $\Delta_1 = \emptyset$ or $\Gamma_2 = \emptyset$. And the cut theorem in SPNCL’, due to these rules, was reduced to:

$$\vdash \Gamma, A \vdash \Delta, A \updownarrow$$

$$\vdash \Delta, \Gamma$$

$$\vdash \Gamma, \updownarrow A \vdash \Delta, A$$

Precyclic pregroups are a result of a bilinear mapping of this reduction in SPNCL’ to pregroups, using the interpretation map of compact
bilinear logic (Buszkowski, 2003). However, note that this reduction was made to relax the constraints on cut theorem in SPNCL. Buszkowski (2002) notes that pregroups are cut eliminated, which means that all properties that can be proved using cut theorem can be proved without it. Due to this cut-elimination, in a pregroup mapped from SPNCL and one mapped from SPNCL’, the adjoint behaviour is identical and therefore their concatenation and reduction do not undergo any change.

Now, the non-precyclic nature of the pregroup $T(B)$ is to be proved. $T(B)$ has been defined as $T(B_T) \cup T(B_{NT})$, where $T(B_T)$ is the pregroup that allows for precyclic transformations and $T(B_{NT})$ is the pregroup that does not allow the same. Note that both SPNCL and SPNCL’ are defined over the same set of sequents, but are defined using different formulae. For a formula $A$ of $\mathcal{L}(\text{SPNCL})$ and formula $B$ of $\mathcal{L}(\text{SPNCL'})$, the cut theorem in SPNCL’ will not be applicable for a proof with both $A$ and $B$, as the rule $\bot \bot$ cannot be used for formulae of SPNCL. Hence, given the compact map from bi-linear logic to pregroups, the analogous $ll$- and $rr$-transformations become inapplicable over a pregroup which has any element that does not obey precyclicity. Therefore, the union of a precyclic and a non-precyclic pregroup is a non-precyclic pregroup.
Abstract

The #MeToo movement is an ongoing prevalent phenomenon on social media aiming to demonstrate the frequency and widespread of sexual harassment by providing a platform to speak up and narrate personal experiences of such harassment. The aggregation and analysis of such disclosures pave the way to the development of technology-based prevention of sexual harassment. We contend that the lack of specificity in generic sentence classification models may not be the best way to tackle text subtleties that intrinsically prevail in a classification task as complex as identifying disclosures of sexual harassment. We propose the Disclosure Language Model, a three-part ULMFiT architecture, consisting of a Language model, a Medium-Specific (Twitter) model, and a Task-Specific classifier to tackle this problem and create a manually annotated real-world dataset to test our technique on this, to show that using a Discourse Language Model often yields better classification performance over (i) Generic deep learning based sentence classification models (ii) existing models that rely on handcrafted stylistic features. An extensive comparison with state-of-the-art generic and specific models along with a detailed error analysis presents the case for our proposed methodology.

1 Introduction

Thirty-five percent of women, including people in the LGBTQIA+ community, are globally subjected to sexual or physical assault, according to a study by UN Women. With the advent of the #MeToo movement (Lee, 2018), discussions about sexual abuse have finally seen the light as compared to before, without the fear of shame or retaliation. Abuse in general and sexual harassment, in particular, is one topic that is socially stigmatized and difficult for people to talk about in both non-computer-mediated and computer-mediated contexts. The Disclosure Processes Model (DPM) (Andalibi et al., 2016) examines when and why interpersonal disclosure may be beneficial and focuses on people with concealable stigmatized identities (e.g., abuse, rape) in non-computer-mediated contexts. It has been found that disclosure of abuse has positive psychological impacts (Manikonda et al., 2016); (McClain and Amar, 2013)), and the #MeToo movement has managed to make social media avenues like Twitter a safer place to share personal experiences.

The information gathered from these kinds of online discussions can be leveraged to create better campaigns for social change by analyzing how users react to these stories and obtaining a better insight into the consequences of sexual abuse. Prior studies noted that developing an automated framework for classifying a tweet is quite challenging due to the inherent complexity of the natural language constructs (Badjatiya et al., 2017).

Tweets are entirely different from other text forms like movie reviews and news forums. Tweets are often short and ambiguous because of the limitation of characters. There are more mis-
spelled words, slangs, and acronyms on Twitter because of its casual form (Mahata et al., 2015). This motivates our study to build a medium-specific Language Model for the segregation of tweets containing disclosures of sexual harassment.

While there is a developing body of literature on the topic of identifying patterns in the language used on social media that analyze sexual harassment disclosure (Manikonda et al., 2018); (Andalibi et al., 2016), very few attempts have been made to segregate texts containing discussions about sexual abuse from texts containing personal recollections of sexual harassment experiences. Efforts have been made to segregate domestic abuse stories from Reddit by Schrading et al. (2015) and Karlekar and Bansal. However, these approaches do not take into consideration the model’s domain understanding of the syntactic and semantic attributes of the specific medium in which the text is present.

In that regard, our paper makes two significant contributions.


2. Comparison of the proposed Medium-Specific Disclosure Language Model architecture for segregation of tweets containing disclosure, with various deep learning architectures and machine learning models, in terms of four evaluation metrics.

2 Related Work

Twitter is fast becoming the most widely used source for social media research, both in academia and in industry (Meghawat et al., 2018) (Shah and Zimmermann, 2017). Wekerle et al. (2018) have shown that Twitter is being used for increasing research on sexual violence. Using social media could support at-risk youth, professionals, and academics given the many strengths of employing such a knowledge mobilization tool. Previously, Twitter has been used to tackle mental health issues (Sawhney et al., 2018b) (Sawhney et al., 2018a) and for other social issues like detection of hate speech content online (Mathur et al., 2018). Mahata et al. (2018) have mobilized Twitter to detect information regarding personal intake of medicines. Social media use is free, easy to implement, available to difficult to access populations (e.g., victims of sexual violence), and can reduce the gap between research and practice. Bogen et al. (2018) discusses the social reactions to disclosures of sexual victimization on Twitter. This work suggests that online forums may offer a unique context for disclosing violence and receiving support. Khatua et al. (2018) have explored deep learning techniques to classify tweets of sexual violence, but have not explicitly focused on building a robust system that can detect recollections of personal stories of abuse.

Schrading et al. (2015) created the Reddit Domestic Abuse Dataset, to facilitate classification of domestic abuse stories using a combination of SVM and N-grams. Karlekar and Bansal improved upon this by using CNN-LSTMs, due to the complementary strengths of both these architectures. Reddit allows lengthy submissions, unlike Twitter, and therefore the use of standard English is more common. This allows natural language processing tools trained on standard English to function better. Our method explores the merits of using a Twitter-specific Language Model which can counter the shortcomings of using pre-trained word embeddings derived from other tasks, on a medium like Twitter where the language is informal, and the grammar is often ambiguous.

N-gram based Twitter Language Models (Vo et al., 2015) have been previously used to detect events and for analyzing Twitter conversations (Ritter et al., 2010). Atefeh and Khreich (2015) used Emoticon Smoothed Language Models for Twitter Sentiment Analysis. Rother and Rettberg (2018) used the ULMFiT model proposed by Howard and Ruder (2018) to detect offensive tweets in German. Manikonda et al. (2018) try to investigate social media posts discussing sexual abuse by analyzing factors such as linguistic themes, social engagement, and emotional attributes. Their work proves that Twitter is an effective source for human behavior analysis, based on several linguistic markers. Andalibi et al. (2016) attempt to characterize abuse related disclosures into different categories, based on different themes, like gender, support seeking nature, etc. Our study aims to bridge the gap between gathering information and analyzing social media disclosures of sexual abuse. Our approach suggests that the language used on Twitter can be treated as a separate language construct, with its own rules and restrictions that need to be ad-
dressed to capture subtle nuances and understand the context better.

3 Data

3.1 Data Collection

Typically, it has been difficult to extract data related to sexual harassment due to social stigma but now, an increasing number of people are turning to the Internet to vent their frustration, seek help and discuss sexual harassment issues. To maintain the privacy of the individuals in the dataset, we do not present direct quotes from any data, nor any identifying information.

Anonymized data was collected from microblogging website Twitter - specifically, content containing self-disclosures of sexual abuse from November 2016 to December 2018.

The creation of a new dataset mandates specific linguistic markers needed to be identified. Instead of developing a word list to represent this language, a corpus of words and phrases were developed using anonymized data from known Sexual Harassment forums. User posts containing tags of *metoo, sexual violence* and *sexual harassment* were also collected from microblogging sites like Tumblr and Reddit. For e.g., subreddits like r/traumatoolbox, r/rapecounseling, and r/survivorsofabuse. Then the TF-IDF method was applied to these texts to determine words and phrases (1-grams, 2-grams and 3-grams) which frequently appeared in posts related to sexual harassment and violence. Finally, human annotators were asked to remove terms from this which were not based on sexual harassment, as well as duplicate terms. This process generated 70 words/phrases which were used as a basis for extraction of tweets.

The public Streaming API was used for the collection and extraction of recent and historical tweets. These texts were collected without knowing the sentiment or context. For example, when collecting tweets on the hashtag #metoo, it is not known initially whether the tweet has been posted for sexual assault awareness and prevention, or if the person is talking about their own experience of sexual abuse, or if the tweet reports an incident or a news report.

<table>
<thead>
<tr>
<th>was assaulted</th>
<th>molested me</th>
</tr>
</thead>
<tbody>
<tr>
<td>raped me</td>
<td>touched me</td>
</tr>
<tr>
<td>groped</td>
<td>I was stalked</td>
</tr>
<tr>
<td>forced me</td>
<td>#WhyILStayed</td>
</tr>
<tr>
<td>#WhenIwas</td>
<td>#NotOkay</td>
</tr>
<tr>
<td>abusive</td>
<td>relationship</td>
</tr>
<tr>
<td>drugged</td>
<td>underage</td>
</tr>
<tr>
<td>inappropriate</td>
<td>followed</td>
</tr>
<tr>
<td>boyfriend</td>
<td>workplace</td>
</tr>
</tbody>
</table>

Table 1: Words/Phrases linked with Sexual Harassment

3.2 Data Annotation

Then, text posts equaling 5117 in all were collected which were subsequently human annotated. The annotators included Clinical Psychologists and Academia of Gender Studies. All the annotators had to review the entire dataset. The tweets were segregated based on the following criteria.

*Is the user recollecting their personal experience of sexual harassment?*

Every post was scrutinized and carefully analyzed by three independent annotators $H_1$, $H_2$ and $H_3$ due to the subjectivity of text annotation. Ambiguous posts were set to the default level of Non-Disclosure.

The following annotation guidelines were followed.

- The default category for all posts is Non-Disclosure.
- The text is marked as Disclosure if it explicitly mentions a personal abuse experience; e.g., "I was molested by my ex-boyfriend"
- Posts which mentioned other people’s recollections were not marked as Disclosure; e.g. "My friend’s boss harassed her"
- If the tone of the text is flippant. e.g. "I can’t play CS I got raped out there hahaha", then it is marked as Non-Disclosure
- Posts related to sexual harassment related news reports or incidents, e.g., "Woman gang-raped by 12 men in Uttar Pradesh", are marked as Non-Disclosure.

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2 http://www.aftersilence.org/
3 https://pandys.org/
4 http://issurvive.org/
• Posts about sexual harassment awareness e.g. "Sexual assault and harassment are unfortunately issues that continue to plague our college community.", are marked as Non-Disclosure.

Finally, after an agreement between the annotators (Table 4), 1126 tweets in the dataset (22% of the dataset) were annotated as Self-Disclosure with an average value of Cohen Kappas inter-annotator agreement $\kappa = 0.83$, while the rest fell into the category of Non-Disclosure. The imbalance of the dataset is encouraged to represent a realistic picture usually seen on social media websites. Our dataset is made publicly available\footnote{\url{github.com/ramitsawhney27/NAACLWSRW19meToo}}, following the guidelines mentioned in Section 7 to facilitate further research and analysis on this very pertinent issue.

4 Methodology

4.1 Preprocessing

The following preprocessing steps were taken as a part of noise reduction: Extra white spaces, new-lines, and special characters were removed from the sentences. All stopwords were removed. Stopwords corpus was taken from NLTK and was used to eliminate words which provide little to no information about individual tweets. URLs, screen names, hashtags(#), digits (0-9), and all Non-English words were removed from the dataset.

4.2 The Disclosure Language Model (DLM)

Previous studies show that traditional learning methods such as manual feature extraction or using representation learning methods followed by a linear classifier have been inefficient in comparison to recent deep learning methods (Khatua et al., 2018). Bag-of-words approaches tend to have a high recall but lead to high rates of false positives because lexical detection methods classify all messages containing particular terms only. Following this stream of research, our work considers deep learning techniques for the detection of social media disclosures of sexual harassment.

CNNs also have been able to generate state of the art results in text classification because of their ability to extract features from word embeddings (Kim, 2014). Recent approaches that concatenate embeddings derived from other tasks with the input at different layers (Maas et al. (2011)) still train from scratch and treat pre-trained embeddings as fixed parameters, limiting their usefulness.

We propose a three-part Disclosure Classification method, based on the Universal Language Model Fine-tuning (ULMFiT) architecture, introduced by (Howard and Ruder, 2018) that enables robust inductive transfer learning for any NLP task, akin to fine-tuning ImageNet models: We use the 3-layer AWD-LSTM architecture proposed by Merity et al. (2017) using the same hyperparameters and no additions other than tuned dropout hyperparameters. Dropouts have been successful in feed-forward and convolutional neural networks, but applying dropouts similarly to an RNNs hidden state is ineffective as it disrupts the RNNs ability to retain long-term dependencies, and may cause overfitting. Our proposed method makes use of DropConnect (Merity et al., 2017), in which, instead of activations, a randomly selected subset of weights within the network is set to zero. Each unit thus receives input from a random subset of units in the previous layer. By performing dropout on the hidden-to-hidden weight matrices, overfitting can be prevented on the recurrent connections of the LSTM.

4.3 Classification

For every tweet $t_i \in D$, in the dataset, a binary valued value variable $y_i$ is used, which can either be 0 or 1. The value 0 indicates that the text belongs to the Non-Disclosure category while 1 indicates Disclosure.

The training has been split into three parts as shown in Figure 1.

• Language Model (LM) - This model is trained from a large corpus of unlabeled data. In this case, a pre-trained Wikipedia Language Model was used.
Disclosure

# When I Was 15 I was molested by my best friend
I was sexually assaulted by my step brother in 2009.
At 8 years old, an adult family member sexually assaulted me.
I was 7 the first time I was sexually assaulted.
I was sexually assaulted by at least 3 different babysitters by the time I was 6 years old.

Table 2: Human Annotation examples for Self Disclosure.

Non-Disclosure

Sexual assault and harassment are unfortunately issues that continue to plague our community.
Trying to silence sexual assault victims is another one. The list goes on and on
Then call for people that cover up sexual assault like Jim Jordan to resign???
sexual assault on public transport is real
agreed! metoo is not just exclusively for women!

Table 3: Human Annotation examples for Non Disclosure

<table>
<thead>
<tr>
<th></th>
<th>H1</th>
<th>H2</th>
<th>H3</th>
</tr>
</thead>
<tbody>
<tr>
<td>H1</td>
<td>–</td>
<td>0.74</td>
<td>0.88</td>
</tr>
<tr>
<td>H2</td>
<td>0.74</td>
<td>–</td>
<td>0.86</td>
</tr>
<tr>
<td>H3</td>
<td>0.88</td>
<td>0.86</td>
<td>–</td>
</tr>
</tbody>
</table>

Table 4: Cohen’s Kappa for Annotators $H_1$, $H_2$, and $H_3$

<table>
<thead>
<tr>
<th>Layer</th>
<th>Howard Dropout</th>
</tr>
</thead>
<tbody>
<tr>
<td>Input</td>
<td>0.25</td>
</tr>
<tr>
<td>General</td>
<td>0.1</td>
</tr>
<tr>
<td>LSTM Internal</td>
<td>0.2</td>
</tr>
<tr>
<td>Embedding</td>
<td>0.02</td>
</tr>
<tr>
<td>Between LSTM Layers</td>
<td>0.15</td>
</tr>
</tbody>
</table>

Table 5: Dropout used by Howard and Ruder (2018)

- **Medium Model (MM)** - The Language Model is used as the basis to train a Medium Model (MM) from unlabeled data that matches the desired medium of the task (e.g., forum posts, newspaper articles or tweets). In our study the weights of the pre-trained Language Model are slowly re-trained on a subset of the Twitter Sentiment140 dataset. This augmented vocabulary improves the model’s domain understanding of Tweet syntax and semantics.

- **Disclosure Model (DM)** - Finally, a binary classifier is trained on top of the Medium Model from a labeled dataset. This approach facilitates the reuse of pre-trained models for the lower layers.

5 Experiment Setup

5.1 Baselines

To make a fair comparison between all the models mentioned above, the experiments are conducted with respect to specific baselines.

Schrading et al. (2015) proposed the Domestic Abuse Disclosure (DAD) Model using the 1, 2, and 3-grams in the text, the predicates, and the semantic role labels as features, including TF-IDF and Bag of Words.

Andalibi et al. (2016) used a Self-Disclosure Analysis (SDA) Logistic Regression model with added features like TF-IDF and Char-N-grams, to characterize abuse-related disclosures by analyzing word occurrences in the texts.

In the experiments, we also evaluate and compare our model with several widely used baseline methods, mentioned in Table 6.

A small subset (10%) of the dataset is held back for testing on unseen data.

5.2 DLM Architectures and Parameters

Our method uses the Weight Dropped AWD-LSTM architecture used by, using the same hyperparameters and no additions other than tuned dropout hyperparameters. Embedding size is 400, the number of hidden activations per layer is 1150, and the number of layers used is 3. Two linear blocks with batch normalization and dropout have

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6https://www.kaggle.com/lazzanova/sentiment140
Architecture Specification

**RNN (Liu et al., 2016)**
Can efficiently represent more complex patterns than the shallow neural networks.

**LSTM (Wang et al., 2018)**
LSTMs are able to capture the long-term dependency among words in short texts.

**Bi-LSTM (Tai et al., 2015)**
At each time step, the hidden state of the Bidirectional LSTM is the concatenation of the forward and backward hidden states.

**GRU (Rana, 2016):**
Simplified variation of the LSTM. Combines the forget and input gates into a single update gate.

**CNN (Kim, 2014)**
Utilize layers with convolving filters that are applied to local features.

**Very Deep-CNN (Conneau et al., 2016)**
Operate directly at the character level and use only small convolutions and pooling operations.

**Char-CNN (Zhang et al., 2015)**
The model can understand abnormal character combinations and new languages.

**fastText Bag of Tricks (Joulin et al., 2017)**
Word features are averaged together to form sentence representations.

**HATT (Yang et al., 2016):**
Two levels of attention mechanisms applied at the word-and sentence-level.

**DP CNN (Johnson and Zhang, 2017)**
Low-complexity word-level CNN that can detect long associations.

**R-CNN (Lai et al., 2015)**
Uses a recurrent structure to capture contextual information when learning word representations.

**CNN-LSTM (Zhou et al., 2015)**
Utilizes CNN to extract higher-level phrase representations, which are fed into an LSTM.

**A-CNN-LSTM (Yuan et al., 2018)**
Combined C-LSTM model with additional attention mechanisms.

**openAI-Transformer (Vaswani et al., 2017)**
Generative pre-training and discriminative fine-tuning of a language model for a specific task.

---

### Table 6: Baseline Specifications

<table>
<thead>
<tr>
<th>Model</th>
<th>Medium</th>
<th>Type</th>
<th>Type</th>
</tr>
</thead>
<tbody>
<tr>
<td>Language Model</td>
<td>Wikipedia</td>
<td>1,000,000,000</td>
<td>Unlabeled</td>
</tr>
<tr>
<td>Medium Model</td>
<td>Twitter</td>
<td>100,000</td>
<td>Unlabeled</td>
</tr>
<tr>
<td>Disclosure Model</td>
<td>Twitter</td>
<td>5117</td>
<td>Labeled</td>
</tr>
</tbody>
</table>

---

been added to the model, with rectified linear unit activations for the intermediate layer and a soft-max activation at the last layer.

The models use different configurations for back-propagation through time (BPTT), learning rate (LR), weight decay (WD), dropouts, cyclical learning rates (CLR) (Smith (2017)) and slanted

triangular learning rates (STLR) (Howard and Ruder (2018)). Additionally, gradient clipping (Pascanu et al. (2013) has been applied to some of the models. The RNN hidden-to-hidden matrix uses a weight dropout for all the models. We train the models for 15 epochs.

For the CLR the four parameters are maximum
<table>
<thead>
<tr>
<th>Architecture</th>
<th>Accuracy</th>
<th>Precision</th>
<th>Recall</th>
<th>F1</th>
</tr>
</thead>
<tbody>
<tr>
<td>DAD Model</td>
<td>0.91</td>
<td>0.90</td>
<td>0.91</td>
<td>0.90</td>
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<tr>
<td>SDA Model</td>
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<tr>
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<td>0.95</td>
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<tr>
<td>openAI-Transformer</td>
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<td>0.96</td>
<td>0.94</td>
</tr>
<tr>
<td>DLM</td>
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<td><strong>0.95</strong></td>
<td>0.97</td>
<td><strong>0.96</strong></td>
</tr>
</tbody>
</table>

Table 8: Comparison with baselines in terms of four evaluation metrics

6 Results and Analysis

6.1 Performance

Table 8 describes the performance of the baseline classifiers as well as the deep learning models based on four evaluation metrics.

The Disclosure Language Model outperforms all baseline models, including RNNs, LSTMs, CNNs, and the linear DAD and SDA models. The A-CNN-LSTM and the Hierarchical Attention Model has a high recall due to its ability to capture long term dependencies better. The attention mechanism allows the model to retain some crucial hidden information when the sentences are quite long. GRUs perform poorly as they are unable to learn some latent features of the sequence that are not directly tied to the elements of the sequence. VD-CNN models typically require a large...
dataset for effective feature extraction. Nine layers were used, as going deeper decreased the accuracy. Short-cut connections tend to help reduce degradation. CL-CNNs may generate unusual words as they would suffer from a higher perplexity due to the nature of prediction (character-by-character). Also, longer training time can lead to vanishing gradients. The fastText model can generate embeddings quicker but performs similarly to the Char-CNN model.

The AWD-LSTM architecture used in the Disclosure Language Model can avoid catastrophic forgetting. The main benefit, however, of the ULMFiT based Disclosure Language Model is that it can perform classifier re-training with a minimal amount of data. The openAI-Transformer model comes a close second in terms of performance. The results show that augmenting the training data with additional domain-specific data (i.e., Tweets) helps to obtain better F1-scores for the segregation of tweets containing instances of personal experiences of sexual harassment.

6.2 Error Analysis

An analysis has been done to show which texts lead to erroneous and a possible explanation of why that might have been the case.

- **Non-Serious** - "I got raped at FIFA the last time I played lol" has a flippant tone. However, the model predicted this as Disclosure because of lack of more contextual information.

- **Third person quote** - "I was followed and harassed by two guys on my way back home last night." This is what my friend had to say after spending one day in Baja. Here someone is referring to another person’s recollection. However, this text contains all the linguistic markers associated with assault disclosure.

- **Uncertainty** - 1. "He was my teacher and I was 12. #metoo"
  2. "I too am a metoo survivor"

Here, the system cannot pick up the context as there is no explicit mention of assault or harassment.

- **Unfamiliarity** - "I was walking home, and I saw in broad daylight a man walking towards me furiously rubbing his privates looking at me". The current dataset lacks in terms of a broad range of phrases that can imply sexual harassment.

- **First person account** - "senatorcollins i beg you for my 12 year old daughter who was sexually assaulted by her teacher please do not vote yes on kavanaugh". The sentence, although in the first person, refers to someone else’s experience.

- **Tweets based on a specific current event** - "I believe Dr. Ford because the same thing happened to me". The user assumes that a majority of the readers will be able to gather context from the amount of information provided. However, the system is unable to pick up this nuance because of lack of information about current events.

7 Ethical Considerations

Human language processing, and human language touches many parts of life, these areas also have an ethical dimension. For example, languages define linguistic communities, so inclusion and bias become relevant topics. Based on the issues highlighted in (Schmaltz (2018)), we address these as:

- **Privacy**: Individual consent from users was not sought as the data was publicly available and attempts to contact the author for research participation could be deemed coercive and may change user behavior.

- **Fairness, Bias & Discrimination**: The exhaustive nature of training data introduces bias in terms of how representative the dataset and hence the trained model is of an underlying community. While it’s not possible to capture all demographics, we try to maximize our coverage by building our dataset in two phases by first developing a lexicon from various microblogging sites.

- **Interpretation**: Although our work attempts to analyze aspects of users’ nuanced and complex experiences, we acknowledge the limitations and potential misrepresentations that can occur when researchers analyze social media data, particularly data from a vulnerable population or group to which the researchers do not explicitly belong. The main
aim of this study was to determine whether it was possible to categorize tweets in this way, rather than to assume the coding was accurate immediately.

8 Conclusion and Future Work

In this work, we proposed a Disclosure Language Model, a three-part ULMFiT architecture, for the task of analyzing disclosures of sexual harassment on social media. On a manually annotated real-world dataset, created in two steps to capture a broad demographic, our systems could often achieve significant performance improvements over (i) systems that rely on handcrafted textual features and (ii) Generic deep learning based systems. An extensive comparison shows the merit of using Medium-Specific Language Models based on an AWD-LSTM architecture, along with an augmented vocabulary which is capable of representing deep linguistic subtleties in the text that pose challenges to the complex task of sexual harassment disclosure. Our future agenda includes: (i) developing a medium-agnostic model robust to the changes in linguistic styles over various forms of social media, (ii) exploring the applicability of our analysis and system to identifying patterns and potential prevention and (iii) applying social network analysis to leverage community interaction and get an overall better understanding.

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SNAP-BATNET: Cascading Author Profiling and Social Network Graphs for Suicide Ideation Detection on Social Media

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Abstract

Suicide is a leading cause of death among youth, and the use of social media to detect suicidal ideation is an active line of research. While it has been established that these users share a common set of properties, the current state-of-the-art approaches utilize only text-based (stylistic and semantic) cues. We contend that the use of information from networks in the form of condensed social graph embeddings and author profiling using features from historical data can be combined with an existing set of features to improve the performance. To that end, we experiment on a manually annotated dataset of tweets created using a three-phase strategy and propose SNAP-BATNET, a deep learning based model to extract text-based features and a novel Feature Stacking approach to combine other community-based information such as historical author profiling and graph embeddings that outperform the current state-of-the-art. We conduct a comprehensive quantitative analysis with baselines, both generic and specific, that presents the case for SNAP-BATNET, along with an error analysis that highlights the limitations and challenges faced paving the way to the future of AI-based suicide ideation detection.

1 Introduction

Suicide is among the top three causes of death among youth worldwide. According to a WHO report1, almost one million people die from suicide annually and 20 times more people attempt suicide. Therefore, suicide causes a global mortality rate of 16 per 100,000, and there is one attempt every 3 seconds on average (Radhakrishnan and Andrade, 2012). Moreover, the effect of it on friends and family members are often devastating (E. Clark and D. Goldney, 2000). What compounds the issue is that while it is preventable and, early detection is crucial in effective treatment, there is a lot of social stigma related to it which prevents people from disclosing their thoughts and seeking professional help. It has been found that people suffering from suicidal ideation make use of social media networks to share information about their mental health online (Park et al., 2012) with many having disclosed their suicidal thoughts and plans (Prieto et al., 2014). Therefore it is a pressing issue to be able to utilize the signals available on social media in order to identify individuals who suffer from suicide ideation in an automated manner and offer them the required help and treatment.

There exists an active field of research in the field of suicidal ideation detection (O’Dea et al., 2015; Sawhney et al., 2018a) that are able to extract meaningful patterns of behavior from users of social media in order to predict suicidal behavior. These have utilized the information presented in the text of the posts that were shared and utilized both traditional as well as deep learning methods. A rich body of literature exists to show the influence of social interactions of at-risk individuals for their effective detection and treatment. However, to the best of our knowledge, no advances have been made to include information from social engagement, ego networks and other user attributes

1https://www.who.int/mental_health/prevention/suicide/suicideprevent/
which we hypothesize would help us in being able to detect suicidal behavior better. Since the interaction of a person with their social surrounding in the form of author profiling from historical tweets and social graph based information can give us a plethora of information about their mental health (Luxtont et al., 2012), we explore the usage of author profiling and other features to detect the presence of suicide ideation in tweets better.

Our contributions to the field are as follows:

1. Creation of a significantly large manually annotated dataset for detection of patterns in suicidal behavior in social media along with historical tweet data and social network graphs which will be made publicly available after anonymization keeping all ethical considerations in mind.

2. Proposing SNAP-BATNET (Social Network Author Profiling - BiLSTM Attention Network), a feature stacking based architecture that uses novel handcrafted features: author profiling, historical stylistic features, social network graph embeddings and tweet metadata with an ablation study for validation.

3. Conducted an extensive quantitative comparison with several traditional and state-of-the-art baselines along with an in-depth error analysis to highlight the challenges faced.

2 Related Work

2.1 Suicidal Ideation Detection

There have been certain advances in the usage of social media to automatically detect cases of suicidal ideation in the past (Sawhney et al., 2018a; De Choudhury et al., 2013; Benton et al., 2017). Cavazos-Rehg et al. (2016) performed a content-based analysis on a small number of depression related tweets to derive certain qualitative insights into the behavior of users displaying suicidal behavior but did not propose any automated solution for the task of detection. Sawhney et al. (2018a) prepared a manually annotated dataset of tweets and proposed a set of features to be used to improve classifier performance but included only text-based features which limits the performance of the classifiers. De Choudhury et al. (2013) developed a crowd-sourced set of patients diagnosed with Major Depressive Disorder(MDD) and used their social media posting through the course of a year to establish a set of signals to help predict depression before its onset. Benton et al. (2017) utilized a novel multitask learning framework to predict atypical mental health conditions with a scope of predicting suicidal behavior but included only text-based features for their multi-task framework.

Furthermore, there have been several forays into tweet classification that utilize a similar set of signals for other applications such as detection of abuse, cyberbullying and hate speech (Mathur et al., 2018b), (Mathur et al., 2018a). Waseem and Hovy (2016) used a public dataset and used a collection of features to show the usefulness of gender-based and location-based information in improving the effectiveness of classifiers. Gambäck and Sikdar (2017) developed a CNN model that used both character n-grams and word2vec features in order to improve the classifier performance greatly. Badjatiya et al. (2017) made use of the same benchmarking dataset, provided a set of baselines and used a combination of randomly initialized embeddings along with LSTM and Gradient Boosting Decision Trees to achieve state of the art performance.

2.2 Author Profiling

The inclusion of author based information has been explored in some tasks related to natural language processing. Waseem and Hovy (2016) utilized gender and location-based information along with text-based features to achieve superior performance. Johannsen et al. (2015) used similar features for syntactic parsing. While it is accepted that such demographic features may improve performance, it is often not possible to extract such features from social media websites like Twitter since this information is often unavailable and unreliable. This has spawned an exciting line of research that makes use of a social graph of interaction between users to derive information about the user. Applications extract information about each user by representing each user as a node in a social graph and creating low dimensional representations usually induced by neural architecture (Grover and Leskovec, 2016; Qiu et al., 2018).

The application of such graph-based features overcomes the limitation caused by unavailability. Mishra et al. (2018); Qian et al. (2018) use such social graph based features to gain considerable improvement in the task of abuse detection. Tasks like sarcasm detection also gain improvement by
using such features (Amir et al., 2016).

3 Data

The unavailability of a public dataset for performing benchmark tests, motivated us to develop our own dataset of considerable size in order to validate our hypothesis. We would like to make our dataset, lexicon, and embeddings public to the research community after making it anonymous and keeping all ethical considerations in mind to improve AI-based suicide prevention and analysis.

The dataset generation was a two-phase process: (i) A lexicon of suicidal phrases was generated (ii) Tweets were scraped using the lexicon and, historic tweets and social engagement data was gathered for each of the users.

3.1 Developing a Lexicon of Suicidal Phrases

In order to scrape tweets to create the dataset, a lexicon of phrases which could indicate suicidal ideation was created. The top posts, most of which are much larger than tweets, were scraped from three different forums which have an abundance of posts with suicidal ideation. These are r/suicidalthoughts (top 100), r/suicidewatch (top 100) and takethislife.com (top 200). Pytextrank is a python module which implements a ranking model for text processing (Mihalcea and Tarau, 2004). This was used to rank and gather the list of the most prominent phrases from these posts. A manual filtering pass was also done to remove posts with little or no suicidal ideation information. The resulting list had 143 phrases such as hit life, think suicide, wanting to die, suicide times, last day, feel pain point, alter- nate life, time to go, beautiful suicide, hate life.

Furthermore, the lexicon was extended by using the lexicon shared in (Sawhney et al., 2018a).

3.2 Data Collection

Collecting tweets: For each phrase in the curated lexicon, tweets were scraped using the Twitter REST API. A total of 48,887 tweets were obtained. Furthermore, retweets and non-English language tweets were removed. A manual check was done to remove the tweets (around 3000) which were trivially non-suicidal. The final dataset has 34,306 tweets. Each tweet in the dataset is described by the fields given in Table 1.

Data for Author Profiling: For the 34,306 tweets in the dataset, there are 32,558 unique users. For each of these users, the tweet timeline (previous 100 tweets or as many available) was scraped. Texts from historical tweets were combined for each of the users to generate the historical corpus for author profiling.

Social Graphs: The engagement between the users from the dataset was captured in the form of social graphs where the users were represented as vertices and edges denoted the relationships. Table 2 shows the different graphs constructed corresponding to four different relationships and also the statistical comparisons between them.

For the Follower Graph, follower lists were scraped for each of the users while for the other three graphs, tweets from the dataset and the historical collection were crawled through.

3.3 Data Annotation

Two annotators, who are students in clinical psychology adept in using social media on a daily basis, were provided with the guidelines to label the tweets as used in (Sawhney et al., 2018b). The guidelines were based on the following classification system -

1. Suicidal intent present
   - Posts where suicide plan and/or previous attempts are discussed.
• Text conveys a serious display of suicidal ideation.
• Posts where suicide risk is not conditional unless some event is a clear risk factor eg: depression, bullying, etc.
• Tone of text is sombre and not flippant.

2. Suicidal intent absent

• Posts emphasizing on suicide related news or information.
• Posts containing no reasonable evidence that the risk of suicide is present; includes posts containing song lyrics, etc.
• Condolences and awareness posts.

An acceptable Cohen’s Kappa score was found between the two annotations (0.72). In cases of ambiguity in labeling or conflicts in merging, the default class 0 (non-suicidal) was assigned. The resulting dataset had 3984 suicidal tweets (12% of the entire dataset).

4 Methodology

The overall methodology is split into three phases: preprocessing of data, extraction of features and finally evaluation of models and feature sets.

4.1 Preprocessing

Due to the unstructured format of the text used in social media, a set of filters were employed to reduce the noise while not losing useful information.

1. A tweet-tokenizer was used to parse the tweet and replace every username mentions, hashtags, and urls with <mention>, <hashtag> and <url> respectively.

2. The tokenized text then underwent stopword removal and was used as an input to WordNet Lemmatizer provided by nltk(Bird and Loper, 2004).

3. Using Lancaster Stemmer, provided by nltk(Bird and Loper, 2004) stemmed text was also generated to be used as inputs for some feature extraction methods.

4.2 Feature Extraction

The features extracted from the data set can be broadly classified into four types: Text-based features, tweet metadata features, User Historical tweets features and Social Graph-based features.

Text Based Features

• TF-IDF: Term Frequency-Inverse Document Frequency was used with the unigrams and bigrams from the stemmed text, using a total of 2000 features chosen by the tf-idf scores across the training dataset. The tf-idf scores were l2 normalized.

• POS: Parts of Speech counts for each lemmatized text using The Penn Tree Bank(Marcus et al., 1993) from the Averaged Perceptron Tagger in nltk is used to extract 34 features.

• GloVe Embeddings: The word embeddings for each word present in the pre-trained GloVe embeddings trained on Twitter (Pennington et al., 2014) were extracted, and for each tweet, the average of these is taken.

• NRC Emotion: The NRC Emotion Lexicon (Mohammad and Turney, 2013) is a publicly available lexicon that contains commonly occurring words along with their affect category (anger, fear, anticipation, trust, surprise, sadness, joy, or disgust) and two polarities (negative or positive). The score along these 10 features was computed for each tweet.

• LDA: Topic Modelling using the probability distribution over the most commonly occurring 100 topics was used as a feature for each tweet. LDA features were extracted by using scikit-learn’s Latent Dirichlet Allocation module (Pedregosa et al., 2011). Only those tokens were considered which occurred at least 10 times in the entire corpus.

Tweet Metadata Features: The count of hashtags, mentions, URLs, and emojis along with the retweet count and favorite count of every tweet was extracted and used as a feature to gain information about the tweets response by the authors environment.

User Historical tweets: To gain information about the behavior of the author and their stylistic choices, a collection of their tweets were preprocessed, and stylistic and semantic features such as the averaged GloVe embeddings, NRC sentiment scores and Parts of Speech counts were extracted.

Social Graph Features: Grover and Leskovec (2016) describe an algorithm node2vec for converting nodes in a graph (weighted or unweighted) into feature representations. This method has been
employed by Mishra et al. (2018) in the task of abuse detection in tweets. node2vec vectors were generated for each of the graphs as introduced in Section 3.2.

5 Baselines

A set of baselines that reflect the current state-of-the-art approaches in short text classification were established. These include methods that use traditional learning algorithms as well as deep learning based models.

- **Character n-gram + Logistic Regression**: Character n-gram with Logistic Regression in the range (1,4) has often been used effectively for classification and works as a strong baseline (Waseem and Hovy, 2016; Badjatiya et al., 2017; Mishra et al., 2018).

- **Bag of Words + GloVe + GBDT**: A Bag of Words (BoW) corpus was generated with unigram and bigram features, the averaged pre-trained GloVe embeddings were then used on a Gradient Boosting Decision Tree which incrementally builds in stage-wise fashion. It is used as a baseline in (Badjatiya et al., 2017).

- **GloVe + CNN**: A CNN architecture inspired from (Kim, 2014; Badjatiya et al., 2017) was used with filter sizes (3,4,5).

- **GloVe + LSTM**: An LSTM with 50 cells was used along with dropout layers \((p = 0.25, 0.5, \text{ preceding and following, respectively})\).

- **ELMo**: Tensorflow Hub \(^9\) was used to get ELMo(Peters et al., 2018) embeddings which are known to have an excellent performance in several fields including sentiment analysis and text classification.

- **USE**: The Universal Sentence Encoder (Cer et al., 2018) encodes text into high dimensional vectors that can be used for tasks like text classification, semantic similarity, and clustering. Tensorflow Hub was used to get sentence encoding. Each tweet was converted encoded onto a dense 512 feature space.

- **Sawhney C-LSTM**: We replicated the C-LSTM architecture used in (Sawhney et al., 2018a) which uses CNN to capture local features of phrases and RNN to capture global and temporal sentence semantics.

- **R-CNN**: Recurrent Convolutional Neural Networks as proposed by (Lai et al., 2015) make use of a recurrent structure to capture contextual information as far as possible when learning word representations.

6 Methodology: SNAP-BATNET

6.1 Graph Embeddings

As discussed in the previous sections, social graphs were constructed for author profiling which could capture demographic features and improve the performance of the classifier. Four such weighted and undirected graphs were constructed: Follower Graph, Mentions Graph, RepliedTo Graph and Quotes Graph. All the self-loops were removed from the graphs, as they do not contribute to suicide-related communication features.

To obtain the author profiles, the nodes in the graphs were converted into feature representation using node2vec (Grover and Leskovec, 2016). node2vec works on the lines of word2vec and determines the context of the nodes by looking into their neighborhoods in the graph. It constructs a fixed number of random walks of constant length for each of the nodes to define the neighborhood of the nodes. The random walks are governed by the parameters \(p\) (return parameter) and \(q\) (input parameter) which have the ability to fluctuate the sampling between a depth-first strategy and a breadth-first strategy.

node2vec by itself does not generate embeddings for solitary nodes which comprised about \(2/3\)rd of the total nodes. As per the empirical rule of normal distribution, 99.73% of the values lie within three standard deviations of the mean. To isolate the solitary nodes from the remaining ones, a random vector was generated three standard deviations away from the mean and was assigned to them.

Embeddings were generated for both weighted and unweighted graphs and were individually studied for the classification task. The number of dimensions and the number of epochs was set to 200 and 10 respectively. A stratified 5-fold grid search was carried out on the hyperparameters \(p, q, \text{walk-length, window-size}\). It was found that the default values for \(p\) and \(q\) (1 and 1) along with

\(^9\)https://tfhub.dev/
Combination F1 AP
Follower+Mentions (CG) 0.808 0.203
Follower+RepliedTo (CG) 0.806 0.196
Mentions+RepliedTo (CG) 0.803 0.197
Follower+Mentions + RepliedTo (CG) 0.807 0.201
Follower+Mentions + RepliedTo (CE) 0.849 0.268

Table 3: Graph combination results (CG-Combining graphs, CE-Combining embeddings) with weighted F1 and area under precision recall curve.

the combination of walk length 10 and window-size 5 performed best. This performance of short walks can be attributed to the sparse nature of the graphs. It was determined that unweighted graphs performed better and were used for generating combined social graph embeddings.

Combining Graph Embeddings: Quotes Graph was discarded from any further study owing to its individual performance in contrast with the other graphs. Its poor performance can be attributed to its statistics as given in the table 2. The rest of the graphs were combined followed two methods: by combining graphs or by combining embeddings using a deep learning approach. The resulting embeddings were trained using a Balanced Random Forest classifier. These results are shown in Table 3.

For generating these combined embeddings, a deep learning model as shown in Figure 1 was designed to be trained on the dataset. After the training, the concatenation layer was picked up as the embedding for the combination, and this was generated for all the users. These embeddings were then used in an LR classifier and a balanced random forest classifier. The results from the balanced random forest classifier were superior and were further used for feature stacking as mentioned in Section 6.2. SNAP-BATNET uses Follower, Mention and RepliedTo embeddings combined using the deep learning approach to generate social graph based features.

6.2 Feature Stacking

The competing systems make use of the text based features for classification. To leverage the availability of different kinds of information in form of tweet metadata, historical author profiling and social graph based embeddings so as to overcome the unavailability of a predefined lexico-semantic pattern in the text, methods of combining information were explored. While tweet metadata is sparse, social graph based embeddings are dense in nature.

Initially, concatenation was used, and several models were tried by performing ablation studies. It was observed that the performance of the classifiers did not change significantly and in some cases deteriorated as features were concatenated. Therefore, it was reasoned that the feature sets should be combined in a way that would have the ability to join them related to their relative importance and also allow learning of non-linear relationships between them. Instead of using concatenation which proved to be ineffective or relative weighing, which is cumbersome, we used a meta learning approach inspired from (Lui, 2012).

One major difference between (Lui, 2012) and our approach is that while it uses Logistic Regression as weak learner for each feature set, different weak learners depending on the feature set or
baseline models were employed in our approach. The weak learners were chosen by using grid search over \{ Logistic Regression, Balanced Random Forest Classifier, SVM \}. For each of the baselines, features from tweet metadata, historical author profiling, and social graph embeddings were combined using Feature Stacking. Logistic Regression was used as L1 learner since stacked LR is theoretically closer to a neural network and can help introduce non-linearity between the features (Dreiseitl and Ohno-Machado, 2002).

Our model SNAP-BATNET uses feature stacking with different L0 learners to combine the feature sets pertaining to text-based information(BiLSTM+Attention), tweet metadata information(Logistic Regression), historical author profiling(Logistic Regression) and social graph embeddings(Balanced Random Forest Classifier). Furthermore, a simple architecture(FeatStackLR) is proposed that uses Logistic Regression as both L0 and L1 learners. An ablation study of the hand-crafted feature sets was carried out using FeatStackLR, which is shown in Table 5. The addition of GloVe based embedding leads to an improvement in results as these embeddings encode semantic information that is missing from statistical features.

### 7 Experiments and Results

#### 7.1 Experimental Setup

All the experiments were conducted with a train-test split of 0.2. The hyperparameters for each learner were calculated by using a 5-fold stratified cross-validation grid search. The CNN and LSTM architectures used 200-dimensional GloVe embeddings pre-trained on Twitter corpus using the Adam optimizer and were run for 10 epochs. The models were implemented in Keras with a Tensorflow Backend. In CNN and LSTM models, 0.1 of the training data was held out as validation data to prevent the model from overfitting. Each baseline model uses Feature Stacking and is used as a L0 learner to extract text-based features to be combined with other feature sets such as tweet metadata, historical author profiling and finally social graph embeddings.

#### 7.2 Results

Zhang and Luo (2018) describe the lacunae of reporting metrics such as micro F1, Precision or Recall provided in cases of highly imbalanced datasets such as Abuse Detection. In order to properly gauge the ability of a system to detect suicidal ideation from tweets, we report the F1, Precision and Recall scores on a per class basis in Table 4. The results in Table 6 include the weighted F1 Score along with the area under the Precision-Recall curve.

<table>
<thead>
<tr>
<th>Model</th>
<th>F1</th>
<th>AP</th>
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<tbody>
<tr>
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<td>.641</td>
</tr>
<tr>
<td>Char ngram+LR</td>
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<td>.646</td>
</tr>
<tr>
<td>BoWV+GloVe+GBDT</td>
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<td>.567</td>
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<td>.612</td>
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<tr>
<td>USE</td>
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<td>.669</td>
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<tr>
<td>ElMo</td>
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<td>.650</td>
</tr>
<tr>
<td>Sawhney-C-LSTM</td>
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<td>.662</td>
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<tr>
<td>RCNN</td>
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<tr>
<td>SNAP-BATNET</td>
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<td>.709</td>
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</table>

Table 5: Ablation study (measured using weighted F1 score and area under Precision-Recall curve).

### Table 4: Results with weighted F1 and area under precision recall curve.
<table>
<thead>
<tr>
<th>Models and Classes</th>
<th>Text Based</th>
<th>+ Metadata</th>
<th>+ Author Profiling</th>
<th>+ Graph</th>
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<td>R</td>
<td>F</td>
<td>P</td>
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<td>.96</td>
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<td></td>
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<td>.62</td>
<td>.63</td>
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<td>.97</td>
<td>.96</td>
</tr>
<tr>
<td></td>
<td>1</td>
<td>.71</td>
<td>.55</td>
<td>.62</td>
</tr>
</tbody>
</table>

Table 6: Results with precision, recall and F1 score on a per class basis.

**Precision-Recall Curve.**

It was observed that adding features such as social graph embeddings, historical author profiling, and tweet metadata led to a considerable improvement in the performance of the classifiers since each feature set encodes a different kind of information that gives the resulting models more adept at the task of classification. As per our hypothesis, the addition of social graph embeddings led to significant improvement in performance across all baseline models. There was an increase in the recall value which is desirable because the reduction of false negatives was more important than the reduction of false positives. Among the traditional classifiers, character n-gram with Logistic Regression performed the best. Moreover, the use of LSTMs in the model such as Sawhney-C-LSTM, RCNN, and SNAP-BATNET improved the classifier performance. This can be reasoned by the effectiveness of LSTM in capturing long term dependencies. Among all the deep learning models, SNAP-BATNET, when combined with all other feature sets using Feature Stacking, performed the best outperforming the current state-of-the-art, i.e., Sawhney-C-LSTM.

### 7.3 Error Analysis

Here we go through some examples posed challenges to highlight limitations and future scope.

1. **Subtle indication:** "Death gives meaning to life" contains subtle indications of suicidal behavior but caused ambiguity between annotators and was not detected by the model.
2. **Sarcasm:** "I want to f**king kill myself lol xD" is one of the several examples where the frivolity of the tweet couldn’t be determined.
3. **Quotes and Lyrics:** "Better off Dead Sleeping With Sirens; I’m as mad, and I’m not going to take this anymore!" are song lyrics and movie dialogues which the annotators were able to identify but the model could not as it lacked real-world knowledge.

### 8 Conclusion

This paper explores the use of information from the behavior of users on social media by using features such as text-based stylistic features in combination with historical tweets based profiling and social graph based embeddings. We develop a
manually annotated dataset on detection of suicidal ideation in tweets, a set of handcrafted features were extracted which were utilized by a set of traditional and state of the art deep learning based models and a quantitative comparison was carried out which validated the hypothesis of the effectiveness of social graph based features and author profiling in suicidal behavior detection with our proposed SNAP-BATNET model, particularly in improving recall. An extensive error analysis and comparison with baselines presents the case for our methodology.

In the future, this work can be extended by exploiting multi-modalities in the data in the form of images, videos, and hyperlinks. Multi-modal approaches have extensively been used for various tasks like predicting social media popularity (Meghawat et al., 2018; Shah and Zimmermann, 2017). Another interesting aspect would be to adapt the pipeline described in this paper to different problems like identifying mentions of personal intake of medicine in social media (Mahata et al., 2018b,a).

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