

A BIN-BASED ONTOLOGICAL FRAMEWORK FOR LOW-RESOURCE N-GRAM SMOOTHING IN LANGUAGE MODELLING

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ABSTRACT

In this paper, we introduce a novel method of smoothing language models (LM) based on the semantic information found in ontologies that is especially adapted for limited-resources language modeling. We exploit the latent knowledge of language that is deeply encoded within ontologies. As such, this work examines the potential of using the semantic and syntactic relations between words from the WordNet ontology to generate new plausible contexts for unseen events to simulate a larger corpus. These unseen events are then mixed-up with a baseline Witten-Bell (WB) LM in order to improve its performance both in terms of language model perplexity and automatic speech recognition word error rates. Results indicate a significant reduction in the perplexity of the language model (up to 9.85% relative) all the while reducing word error rate in a statistically significant manner compared to both the original WB LM and baseline Kneser-Ney smoothed language model on the Wall Street Journal-based Continuous Speech Recognition Phase II corpus.

Keywords: Language modeling, context modeling, ontologies, low-resource speech recognition

1. INTRODUCTION

Language modeling is an integral part of the speech recognition task where the purpose is to try to reliably predict what the speaker is most likely to say next. Today, n -gram language modeling remains the dominant technique in automatic speech recognition applications. Formally, we compute the probability of word $w = w_n$ given its history or context $h : P(w|h)$ where $h = w_{n-1}, w_{n-2}, \dots, w_{n-N}$ is obtained from counts of very large textual corpus. However, this is not always convenient or practically achievable since new sentences are constantly being created [1]. For example it would be impossible to predict a novel sentence, since its count would be zero. In addition, it is easy to see that as the length of the sentence and/or vocabulary grows, the task of obtaining those counts becomes impractical. The goal of n -gram language models then is to simplify the prediction of the n -th word based on the $n-1$ previous words using maximum likelihood estimates (MLE):

$$P_{MLE}(w_n|h) = \frac{C(h, w_n)}{C(h)}. \quad (1)$$

Two well-known drawbacks of language modeling with n -grams are its sensitivity to the corpus and data sparseness, i.e. if we train our model on text from Shakespeare, it is evident that the model will not be adequate for the prediction of astrophysical text, namely the count

of the word "magnetohydrodynamics" in a Shakespearian work $C(\text{magnetohydrodynamics}_{\text{astrophysics}} | h_{\text{Shakespeare}})$ is surely equal to 0. That is why multiple approaches exist to either obtain better language models (LM) or smooth existing ones in order to improve their statistical accuracy and generalizability. Notable techniques include discounting, interpolation and various backoff schemes such as Good-Turing Discounting, Witten-Bell discounting [2], Modified Kneser-Ney smoothing [3] or Hierarchical Pitman-Yor language models [4] used to smooth out low-order counts. Variable length n -grams and class-based n -grams [5] also enabled better performance of language models. Other work such as [6, 7, 8, 9] looks into exploiting the vast amount of text available on the Web and raw n -gram counts such as those provided by Google.

Although we have made significant leaps in the area there is still room for progress. This is particularly true in the case where we don't have access to training data that relates to the testing or real-world data or that such resources are limited and/or too expensive. It is for these reasons that this work mainly focuses on semantically expanding the information found in small corpus for smoothing language models.

In this work, we introduce a novel method of smoothing language models by exploiting the semantic information found in ontologies. It will be used in combination with Witten-Bell smoothing since the intuition behind this smoothing algorithm is to look at the diversity of contexts in which a particular history occurs: how likely is it to see a new n -gram given a particular history. This is especially suited for our n -gram count smoothing algorithm since one of its goals is to generate a new set of probable n -gram events that enrich the contextual diversity of the language model.

The contribution of this work is the demonstration that ontological relations can reliably be used for smoothing existing n -gram counts as well as producing novel n -gram contexts to enrich the language model and increase its performance both in terms of perplexity and in an automatic speech recognition task.

The outline of this paper is as follows. Section 2 introduces ontologies and the WordNet ontology used in this work. Section 3 describes our proposed ontological n -gram count smoothing algorithm. Section 4 proceeds with the description of the experiment set-up and the evaluation of our algorithm. Finally, in section 5, we conclude and discuss our results.

2. ONTOLOGIES

Ontologies are used to relate concepts in a hierarchical and relational fashion that ultimately describe the meaning of terms. Logical and philosophical entities can be grouped contextually with the use of named relations. Figure 1 shows a portion of the en-

N is a lexeme (lemma + forms). Each node can be from multiple classes, e.g. *cat* is both a noun and a verb and each edge E represents a relation between two nodes.

The second step is to tag our corpus's terms by their part-of-speech. For this task, we used the Penn-State TreeTagger [12]. This is to preserve the meaning of the n -gram counts. For example, we potentially wouldn't want to smooth the n -gram entry of a verb with that of a noun. As per [11] we restrict the search space to synonymous terms for this work. Once the corpus is tagged, we proceed with the counting task using the SRILM toolkit [13].

Our algorithm operates in D passes: each pass creates a bin containing the smoothed counts for distance d using (2). Each pass also adds unseen $w_r|h$ events in order to reduce the effects of limited corpus. Since the number of new counts generated can be particularly high given a large distance metric, we parallelized our algorithm. This gives us a speed-up of up to 52% on four threads with $d_{max} = 5$ for the whole smoothing process.

3.3. Ontology search algorithm for R related terms

Since the search space of the ontology is considerable we needed an efficient algorithm to look for the related terms of each n -gram entry. The fact that we use an unweighted graph enabled us to avoid the inherent challenges of searching weighted directed graphs. The algorithm is a depth-limited form of unweighted iterative deepening depth-first search. Basically, as we progress through the graph each relation is marked gray; we then recursively explore each unmarked child node (we do not return to it since it has already been seen at a lower level), and thus the "shortest" path to it is already known. This greatly reduces the cost of exploring the ontology all the while guaranteeing optimal paths for each relation.

4. EXPERIMENTS AND RESULTS

4.1. Limited Resources Experimental Setup

The experiments reported in this paper were performed using the Wall Street Journal-based Continuous Speech Recognition Phase II corpus (WSJ1). We train our language models on the 76,136 sentences provided by the standard WSJ1 training sets for a total of 1,243,340 words. For the evaluation we restricted ourselves to the Hub 1 and Spoke 1 tests (10,347 words in 582 sentences) without filtering out sentences with out-of-vocabulary words (OOV rate of 5.41%).

The SRILM toolkit and HTK toolkit [14] were used for our experiments for language modeling and automatic speech recognition respectively. We obtain raw training counts using the SRILM toolkit. These raw counts are then used to generate ontologically smoothed count bins for each distance group, note that, as per eq. 2, the majority of the counts are fractional counts. Then, Witten-Bell LMs were created for each bins. Finally, a mixture-based language model is created by mixing up each bin-based LM with the original Witten-Bell LM. Optimal mixup weights were obtained through the use of Powell's optimization algorithm [15] using the SciPy library [16] with the objective function defined as the word error rate of the held-out development set consisting of 4,340 sentences for a total of 71,759 words. Unfortunately, the SRILM toolkit does not support fractional counts for Kneser-Ney smoothing and as such, we cannot properly interpolate the baseline KN model with our bins.

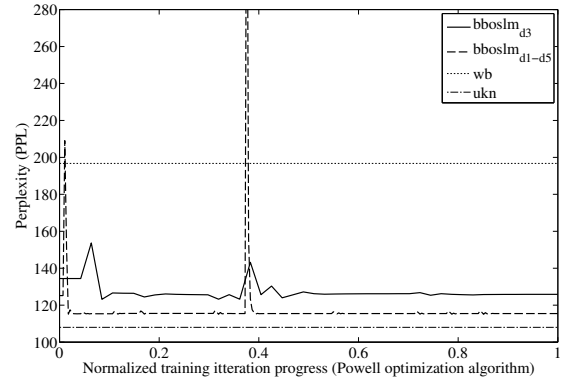


Fig. 2. Perplexity evolution during the Powell WER optimization process to determine the best BBOSLM bin mixup weights for 20K word trigram LMs trained on the WSJ1 training corpus and conditioned the WSJ1 development corpus. The spikes are due to the algorithm trying to see if it is in a local optimum.

Perplexity and word error rate (WER) are used to measure the performance of our language modeling technique. For the acoustic models, we use the well-known recipe from Vertanen [17]. Our model is trained by using all of the WSJ1 training data using the 40 phones set of the CMU dictionary. Cross-word tied-state triphones with 32 Gaussian mixtures per state and 64 Gaussian mixtures per silence state are eventually generated. The acoustic models are represented by Mel Frequency Cepstral Coefficients with energy, delta and acceleration ($MFCC_E_D_A$) for a total of a 39-dimensional feature vector. The HDecode tool with parameters for the pruning beam width, word insertion penalty, and the language model scale factor of 220.0, -4.0, and 15.0 respectively were used for the automatic speech recognition test.

4.2. Perplexity evaluation

Perplexity evaluation was performed on the Hub 1 (read WSJ baseline) and Spoke 1 (language model adaptation) portion of the WSJ1 corpus which consist of 582 sentences and 10,347 words. The standard 20,000-word WSJ closed-vocabulary was used for the evaluation and backed-off trigram language models were generated using SRILM with the `-float-counts` option enabled.

The average path length of the ontology is 59.24 with a network diameter of 15 measured using the Gephi toolkit [18]. The network diameter is a measure of the maximum distance between any two pairs in the graph. Since it would be too computationally intensive to fully explore each node, we chose to study the effects of d_{max} from 1 to 5. From the $\sim 346,000$ original raw counts we end up with $\sim 15,284,000$ smoothed counts for B_5 . Figure 2 shows the evolution of the language model perplexity throughout the Powell optimization algorithm for the Bin-based ontological smoothing against the baseline models trained using Witten-Bell (WB) smoothing and the original Kneser-Ney (UKN) smoothing. Note that we were not able to include results for Modified Kneser-Ney (MKN) smoothing as the SRILM toolkit kept computing negative discount parameters for the trigrams. Results for both single and multiple bins mixup are provided. Table 1 shows a summary of the best results obtained. As was expected, our technique is able to significantly reduce the perplexity of the baseline language models with the best model showing maximal perplexity reduc-

Table 1. Comparison of perplexity on WSJ1 Hub 1 and Spoke 1 evaluation corpus using 20K word trigrams using best WER performing single and multiple bins mix-up.

Language Model	Perplexity
Witten-Bell	142.61
Kneser-Ney	117.15
BBOSLM	-
$d = 1$	137.46
$d = 2$	137.5
$d = 3$	137.92
$d = 4$	137.54
$d = 5$	137.53
$1 \leq d \leq 2$	137.14
$1 \leq d \leq 3$	128.55
$1 \leq d \leq 4$	129.45
$1 \leq d \leq 5$	122.49

tion of 9.85% for the WB LM. We were able to get very close to the perplexity of Kneser-Ney discounting for Bins $1 \leq d \leq 5$. However this is to be expected as the training process did not explicitly look at minimizing perplexity but rather minimizing Word Error Rates. This is consistent with the literature as it is well known that a reduction in perplexity will not always translate in improvements in automatic speech recognition tasks. The spikes in perplexity are due to the Powell algorithm exploring the bounded space to see if it is not currently in a local optimum. This is encouraging since it demonstrates that our technique can compete with state-of-the-art language modeling techniques.

4.3. Automatic speech recognition experiments

This experiment investigates the performance of the BBOSLM on word error rate in an automatic speech recognition task. Much like the perplexity evaluation, we provide the WER results for single bin mixup and multiple bins mixup in relation to the LM perplexity optimization process. The optimization process was run three times with evenly distributed starting λ values, figure 3 shows the optimization process for the best performing final models and starting with $\lambda_0 = 0.9$. Final λ_1 from eq. 3 values for the two best performing LMs, $BBOSLM_{d3}$ for single bin mix-up and $BBOSLM_{d1-d5}$ for multiple bin mix-up, are 0.9962 and 0.8913 respectively. Globally, results show that the best performing LM are the ones mixing-up multiple smoothed counts bins. These results are confirmed by every statistical significance test available in the NIST Scoring Toolkit: a statistically significant reduction (mean $p = 0.001$) in word error rate of 6.8% relative for $BBOSLM_{d1-d5}$ and statistically significant reduction (mean $p = 0.01$) for $BBOSLM_{d4}$ with a considerable reduction in insertion and substitution rates compared to the baseline Witten-Bell discounted LM.

5. CONCLUSION AND DISCUSSION

In this paper, we introduced an original method of preprocessing and smoothing n-gram counts based on the semantic information found in ontologies for use in limited resources automatic speech recognition. It exploits the relations found in ontologies to create bins of new n-gram events related to existing n-gram

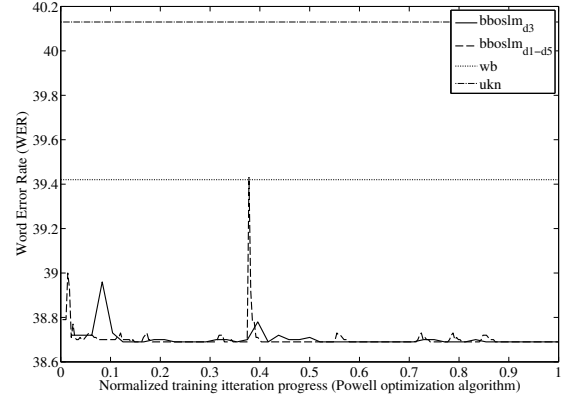


Fig. 3. WER (%) evolution of the BBOSLM during the WER optimization process on WSJ1 development corpus using 20K word trigrams. The “large” spike in the training process is due to the Powell algorithm jumping in λ values to see if it is in a local optimum.

Table 2. ASR performance comparison on WSJ1 Hub 1 and Spoke 1 evaluation corpus using 20K word trigrams using single and multiple bins mix-up.

LMS	PPL	SUB	DEL	INS	WER
WBLM	142.61	22.43	5.89	2.64	30.96
KNLM	117.15	21.87	7.71	1.41	30.98
BBOSLM	-	-	-	-	-
$d = 1$	137.46	21.20	6.65	1.72	29.56
$d = 2$	137.5	21.13	6.67	1.77	29.57
$d = 3$	137.92	20.99	6.67	1.78	29.44
$d = 4$	137.54	21.08	6.73	1.74	29.55
$d = 5$	137.53	21.05	6.79	1.73	29.57
$1 \leq d \leq 2$	137.14	21.18	6.65	1.73	29.56
$1 \leq d \leq 3$	128.55	20.69	6.76	1.58	29.03
$1 \leq d \leq 4$	129.44	20.72	6.74	1.6	29.06
$1 \leq d \leq 5$	122.49	20.23	7.03	1.59	28.85

events. These bins are transformed into language models (LM) and then mixed-up with the original LM. Experiments demonstrate that our smoothing technique reduced the perplexity of a given language model by up to 9.85% when considering bins of related words with a maximal ontological distance of up to 5. In addition, we evaluated our language models on the Hub 1 and Spoke 1 tasks of the Wall Street Journal-based Continuous Speech Recognition Phase II corpus. We showed that it reduced word error rate (WER) in a statistically significant manner compared to Kneser-Ney discounting ($p = 0.001$). The fact that we were able to obtain statistically significant performance increase indicate that it shows promise as a tool for language model smoothing, especially when textual resources are sparse. Future work will look to incorporate different ontological distance metrics such as those proposed by Jing et al.[19] as well as exploring information theory-based schemes to weight the different relations between words described by the ontology.

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