Focusing on Fast Neural Network Inference

Anonymous ACL submission

Abstract

This paper describe the submissions to the efficiency track for GPUs by members of the University of Edinburgh, Adam Mickiewicz University, Tilde and University of Alicante. We focus on efficient implementation of the recurrent deep-learning model as implemented in Amun, the fast inference engine for neural machine translation. We improve the performance with an efficient mini-batching and maxibatching algorithm. Translation speed was also reduced by fusing the softmax operation with the beam seach algorithm.

1 Introduction

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As neural machine translation (NMT) models have become the new state-of-the-art, the challenge is to make their deployment efficient and economical. This is the challenge that this shared task is shining a spotlight on.

It is tempting to use a general purpose deeplearning toolkit such as Tensorflow or Pytorch to train and do the the inference where faster translation could be obtained by tuning parameters within the toolkit, and choosing fast models.

We take an opposing approach by using and enhancing a custom inference engine, Amun (Junczys-Dowmunt et al., 2016), which we developed on the premise that fast deeplearning inference is an issue that deserves dedicated tools which are not compromised by other objectives such as training or support for multiple models. As well as delivering on the useful goal of fast inference, it can serve as a test-bed for novel ideas on neural network inference, and is useful as a means to explore the upper limit of the possible speed for a particular model and hardware. That is, Amun is an inference-only engine that supports a limited number of NMT models that put fast inference on modern GPU above all other considerations.

We submitted two systems to this year's shared task for the efficient translation on GPU. Our first submission was tailored to be as fast as possible while being above the baseline BLEU score. Our second submission trade some of the speed of first submission to return good quality translation.

We will describe below our experimental setup as well as the enhancements to Amun that we feel enabled it to achieve its outstanding speed.

2 Improvements

We desribe the main enhancements to Amun since the original 2016 publication.

2.1 Batching

The use of mini-batching is critical for fast model inference. The size of the batch is determined by the number of inputs sentences to the encoder in an encoder-decoder model. However, the number of batches during decoding can vary as some sentences have completed translating or the beam search add more hypotheses to the batch.

It is tempting to ignore these considerations, for example, by always decoding with a constant batch and beam size and ignoring hypotheses which are not needed but this comes at a cost of lower translation speed due to wasteful processing.

The Amun engine implements an efficient batching algorithm that takes into account the actual number of hypotheses that needs to be decoded at each decoding step, Figure 1.

Algorithm 1 Mini-batching procedure BATCHING(encoded records i) Create batch b from iwhile b is not empty do $b' \leftarrow \text{DecodeAndBeamSearch}(b)$ $b \leftarrow \emptyset$ for all hypo h in b' do if $h \neq </s>$ then Add h to bend if end for end while end procedure

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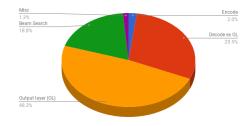
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Figure 1: Proportion of time spent during translation (Europarl test set)

2.2 Softmax and Beam Search Fusion

Most NMT more predict a large number of classes in their output layer, corresponding to the number of words or subword units in their target language. For example, Sennrich et al. (2016) experimented with target vocabulary sizes of 60,000 and 90,000 sub-word units. This makes the output layer of NMT models very computationally expensive. Figure 1 shows the breakdown of amount of time during translation our NMT system; nearly 70% of the time is involved in the output layer. We describe the steps to fuse the operations to improve efficiency.

The output layer of most deep learning models consist of the following steps

- 1. multiplication of the weight matrix with the input vector p = wx
- 2. addition of a bias term to the resulting scores p = p + b
- 3. applying the activation function, most commonly softmax $p_i = \exp(p_i) / \sum \exp(p_i)$
- 4. a beam search for the best (or n-best) output classes $\operatorname{argmax}_i p_i$

In models with a small number of classes such as binary classification, the computational

effort required is trivial and fast but this is not the case for the large number of classes found in typical NMT models.

We leave step 1 for future work and focus on the last three steps, the outline for which are shown in Algorithm 2. For brevity, we show the algorithm for 1-best, a beam search of the n-bests is a simple extension of this.

As can be seen, the matrix p is iterated over five times - once to add the bias, three times to calculate the softmax, and once to search for the best classes. We propose fusing the three functions into one kernel, a popular optimization technique (Guevara et al., 2009), making use of the following observations.

Firstly, softmax and exp are monotonic functions, therefore, we can move the search for the best class from FIND-BEST to SOFTMAX, at the start of the kernel.

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Secondly, we are only interested in the probabilities of the best classes. Since they are now known at the start of the kernel, we compute softmax only for those classes.

Thirdly, the calculation of max and sum can be accomplished in one loop by adjusting sum whenever a higher max is found:

$$sum = e^{a-max_b} + e^{b-max_b}$$

$$= e^{a-max_a+\Delta} + e^{b-max_b}$$

$$= e^{\Delta}e^{a-max_a} + e^{b-max_b}$$

where max_a is the original maximum value, max_b is the new, higher maximum value, ie. $max_b > max_a$, and $\Delta = max_a - max_b$. The outline of our function is shown in Algorithm 3.

In fact, a well known optimization is to skip softmax altogether and calculate the argmax over the input vector, Algorithm 4. This is only possible for beam size 1 and when we are not interested in returning the softmax probabilities.

```
Algorithm 2 Original softmax and beam
Search Algorithm
  procedure ADDBIAS(vector p, bias vector
      for all p_i in p do
           p_i \leftarrow p_i + b_i
       end for
  end procedure
  procedure SOFTMAX(vector p)

    ▷ calculate max for softmax stability

       max \leftarrow -\infty
      for all p_i in p do
           if p_i > max then
               max \leftarrow p_i
           end if
       end for
                          > calculate denominator
       sum \leftarrow 0
       for all p_i in p do
           sum \leftarrow sum + \exp(p_i - max)
      end for

    ▷ calculate softmax

       for all p_i in p do
          p_i \leftarrow \frac{\exp(p_i - max)}{sum}
       end for
  end procedure
  procedure FIND-BEST(vector p)
       max \leftarrow -\infty
      for all p_i in p do
           if p_i > max then
               max \leftarrow p_i
               best \leftarrow i
           end if
      end for
        return max, best
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end procedure

Algorithm 3 Fused softmax and beam search **procedure** FUSED-KERNEL(vector p, bias vector b) ⊳ add bias, calculate max & argmax $max \leftarrow -\infty$ $sum \leftarrow 0$ for all p_i in p do $p_i' \leftarrow p_i + b_i$ if $p_i' > max$ then $\Delta \leftarrow max - p_i'$ $sum \leftarrow \Delta \times sum + 1$ $max \leftarrow p'_i$ $best \leftarrow i$ else $sum \leftarrow sum + \exp(p_i' - max)$ end if end for **return** $\frac{1}{sum}$, best end procedure

```
Algorithm 4 Find 1-best only procedure FUSED-KERNEL-1-BEST(vector p, bias vector b)

max \leftarrow -\infty

for all p_i in p do

if p_i + b_i > max then

max \leftarrow p_i + b_i

best \leftarrow i

end if

end for

return best

end procedure
```

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These changes do not reduce the asymptotic runtime of our algorithm, which remains at $\mathcal{O}(|p|)$, but it does bring the number of iterations over p down from 5 to 1. In practise, this has a significant affect on inference speed.

2.3 Half-Precision

Reducing the number of bits needed to store floating point values from 32-bits to 16-bits promises to increase translation speed through faster calculations and reduced bandwidth usage. 16-bit floating point operations are supported by the GPU hardware and software available in the shared task.

In practise, however, efficiently using half-precision value requires a comprehensive red-velopment of the GPU code. We therefore make do with using the GPU's Tensor Core fast matrix multiplication routines which transparently converts 32-bit float point input matrices to 16-bit values and output a 32-bit float point product of the inputs.

3 Experimental Setup

Both of our submitted systems uses the sequence-to-sequence model similar to that described in Sennrich et al. (2016), containing a bidirectional RNN in the encoder and a two-layer RNN in the decoder. We use BPE to adjust the vocabulary size.

We used a variety of GPUs to train the models but all testing was done on an Nvidia V100. Translation quality was measured using BLEU, specifically multi-bleu as found in the Moses toolkit. The validation and test sets provided by the shared task organisers were used to measure translation quality, but a 50,000 sentence subset of the training data was used to measure translation speed to obtain longer, more accurate measurements.

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	State dim		
Vocab size	256	512	1024
1000	12.23		12.77
5000	16.79		17.16
10,000	18.00		18.19
20,000	_		19.52
30,000	18.51	19.17	19.64

Table 1: Validation set BLEU (newstest2014) for GRU-based model

3.1 GRU-based system

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Our first system uses gated recurrent units (GRU) throughout, trained with Marian, but submitted both systems using the Amun inference engine.

We experimented with varying the vocabulary size and the RNN state size before settling for a vocabulary size of 30,000 (for both source and target language) and 256 for the state size, Table 1.

After further experimentation, we decided to use sentence length normalization and NVidia's Tensor Core matrix multiplication which increased translation quality as well as translation speed. The beam was kept at 1 throughout for the fastest possible inference.

3.2 mLSTM-based system

Our second system uses multiplicative-LSTM in the encoder and the first layer of a decder, and a GRU in the second layer, using an extension of the Nematus toolkit which supports such model. We trained 2 systems with differing vocabulary sizes and varied the beam sizes before choosing the the settings that produces good translation quality on the validation set, Table 2.

	Vocab size		
Beam size	40,000	50,000	
1	23.45	23.32	
2	24.15	24.04	
3	24.48		
4	24.42		
5	24.48		

Table 2: Validation set BLEU for mLSTM-based model

4 Result

We have identified two areas where typical NMT models differ significantly from other deep-learning models that impact on their inference speed. Firstly, NMT models usually contains a large number of output classes, corresponding to the output vocabulary. Secondly, rather than a single label, the output from machine translation models is a sequence of class labels that makes up the words or sub-words in the target sentence. Target sentence lengths are unpredictable, leading to inefficiencies during batch processing. We provide solutions for each of these issues which together increase batched inference speed by up to 57% on modern GPUs without affecting model quality.

We examine two areas that are critical to fast NMT inference where the models differ significantly from other deep-learning models. We believe the optimizations of these models have been overlooked by the general deep-learning community.

Even with sequence-to-sequence NMT models, target sentence lengths will still differ even for similar length inputs, compromising decoding performance. Figure 2 shows the actual batch size during decoding with a maximum batch size of 128; the batch was full in only

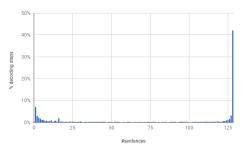


Figure 2: Actual batch size during decoding (Europarl test set)

42% of decoding iterations.

We will propose an alternative batching algorithm to increase the actual batch size during decoding without the problems associated with mini-batching and maxi-batching.

5 Prior Work

Deep-learning models has been successfully used on many machine learning task in recent years in areas as diverse as computer vision and natural language processing. This success have been followed by the rush to put these model into production. The computational resources required in order to run these models has been challenging but, fortunately, there has been a lot work to meet this challenge.

Hardware accelerators such as GPUs are very popular but other specialized hardware such as custom processors (Jouppi et al., 2017) and FPGA (Lacey et al., 2016) have been used. Hardware-supported reduced precision is also an ideal way to speed up model inference (Micikevicius et al., 2017).

Simpler, faster models have been created that are as good, or almost as good, as more slower, more complex models (Bahdanau et al., 2014). Research have also gone into smaller, faster models that can approximate slower, big-

ger models (Kim and Rush, 2016). Some models have been justified as more suited for the parallel architecture of GPUs (Vaswani et al., 2017) (Gehring et al., 2017).

The speed of the softmax layer when used with large vocabularies have been looked at in Grave et al. (2016) but there have been many other attempts at faster softmax, for example Mikolov et al. (2013), Zoph et al. (2016).

There has been surprisingly little work on batching algorithms, considering its critical importance for efficient deep-learning training and inference. Neubig et al. (2017) describe an novel batching algorithm, however, its aim is to alleviate the burden on developers of batching models rather than faster batching.

Our paper follows most closely on from Devlin (2017) which achieved faster NMT inference mainly by novel implementation of an existing model. However, we differ by focusing on GPU implementation, specifically the output layer, and the batching algorithm which was not touched on by the previous work. Our model is based on that of Cho et al. (2014) and Sennrich et al. (2016) but our work is applicable to other models and applications.

6 Proposal

6.1 Top-up Batching

The standard mini-batching algorithm is outlined in Algorithm 1.

This algorithm encode the sentences for a batch, followed by decoding the batch. The decoding stop once all sentences in the batch are completed. This is a potential inefficiency as the number of remaining sentences may not be optimal.

We will focus on decoding as this is the more compute-intensive step, and issues with differing sentence sizes in encoding can partly be ameliorated by maxi-batching.

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Our proposed top-up batching algorithm encode and decode asynchronously. The encoding step, Algorithm 5, is similar to the main loop of the standard algorithm but the results are added to a queue to be consumed by the decoding step.

Algorithm 5 Encoding for top-up batching procedure ENCODE while more input do Create encoding batch b Encode(b) Add b to queue q end while end procedure

Rather than decoding the same batch until all sentences in the batch are completed, the decoding step processing the same batch continuously. New sentences are added to the batch as old sentences completes, Algorithm 6.

```
Algorithm 6 Decoding for top-up batching

procedure DECODE

create batch b from queue q

while b is not empty do

Decode(b)

for all sentence s in b do

if trans(s) is complete then

Replace s with s' from q

end if

end for

end while

end procedure
```

7 Results

7.1 Softmax and Beam Search Fusion

Fusing the last 3 steps led to a substantial reduction in the time taken, especially for beam size

	Baseline	Fused		
Beam size I				
Multiplication	25.62	26.47 (+3.3%)		
Add bias	4.98			
Softmax	12.81	7.94 (-76.9%)		
Beam search	16.64			
Beam size 5				
Multiplication	112.9	115.04 (+1.9%)		
Add bias	23.66			
Softmax	56.61	67.58 (-56.5%)		
Beam search	75.12			

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Table 3: Time taken in sec (Europarl)

	Baseline	Fused			
Beam size I					
Multiplication	21.47	21.94 (+2.2%)			
Add bias	3.29				
Softmax	8.83	5.70 (-78.1%)			
Beam search	13.91				
Beam size 5					
Multiplication	79.66	80.01 (+4.4%)			
Add bias	14.95				
Softmax	36.86	42.91 (-64.4%)			
Beam search	68.80				

Table 4: Time taken in sec (OpenSubtitles)

of 1 where the softmax probability does not actually have to calculated, Table 3 and 4. This led to an overall increase in translation speed of up to 23% for the Europarl test set, Figure 3, and up to 41% for the OpenSubtitles test set, Figure 4.

The amount of time taken by the output layer dominates translation time for the OpenSubtitles test set, Figure 5, explaining the bigger overall translation speed with the fused kernel.

7.2 Top-up Batching

After some experimentation, we decided to topup the decoding batch only when it is at least half empty, rather than whenever a sentence has completed.

The top-up batching and maxi-batching have similar goals of maximizing efficiency when

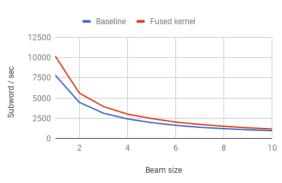


Figure 3: Speed using the fused kernel (Europarl)

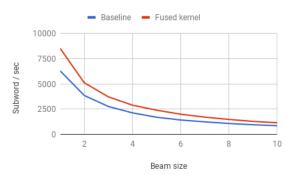


Figure 4: Speed using the fused kernel (Open-Subtitles)

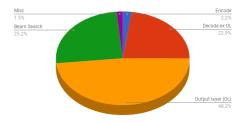


Figure 5: Proportion of time spent during translation (OpenSubtitles)

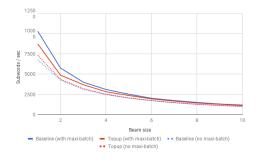


Figure 6: Speed using top-up batching (Europarl)

translating batches of different lengths. Therefore, using both methods together gives limited gains, in fact, using top-up batching with maxibatch slows of between 2% to 18% when translating the Europarl test set due to the overhead of using the algorithm, Figure 6. When maxibatching is inappropriate, using top-up batching alone matches the performance of maxibatching, even being slightly faster when a small beam is used.

The results are better when translating the OpenSubtitles test set, Figure 7. The top-up batching does not harm performance when used with maxi-batching, even helping a little for small beams. However, top-up batching increases translation speed by up to 12% when used alone.

The gain from top-up batching is partially dependent on how much time the standard mini-batching algorithm spend decoding with a very small number of sentences as this is does not fully utilize the GPU cores. From Figure 8, 20% of the decoding iterations have less than 8 sentences remaining in the batch when translating the Europarl test test with maxi-batching. This figure is 50% for the OpenSubtitles test set with no maxi-batching. The OpenSubtitles test set has a higher variance of sentence lengths

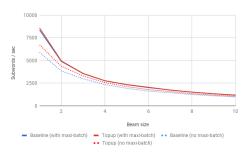


Figure 7: Speed using top-up batching (Open-Subtitles)

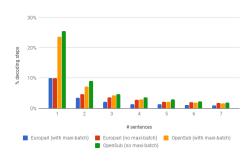


Figure 8: Actual batch size during decoding

relative to its average sentence length, which forces the mini-batch algorithm to continue decoding with a small number of sentences while the other sentences have already completed.

7.3 Cummulative Results

Using both the fused kernel and top-up batching to translate led to a cummalative speed improvement of up to 57% and 34% for the Open-Subtitles and Europarl test set, respectively, when no maxi-batching is used, Figure 9 and Figure 10. With maxi-batching, the speed was up to 41% and 21%.

8 Conclusion and Future Work

We have presented two methods for faster deeplearning inference, targeted at neural machine

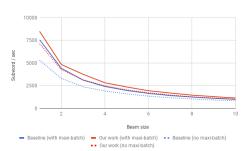


Figure 9: Cummulative results (Europarl

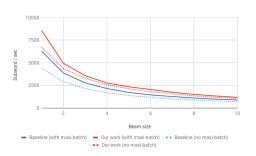


Figure 10: Cummulative results (OpenSubtitles

translation.

The first method focused on output layer of the neural network which accounts for a large part of the running time of the NMT model. By fusing the output layer with the beam search, we are able to increase translation speed by up to 41%.

The second method replaces the minibatching algorithm with one that avoids decoding with a small number of sentences, maximizing the parallel processing potential of GPUs. For certain scenarios, this increases translating speed by up to 12%.

For future work, we would like to apply our optimization for other NLP and deep-learning tasks. We are also interested in further optimization of the output layer in NMT, specifically the matrix multiplication which still takes up a significant proportion of the translation time.

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