

BATCH NORMALIZED RECURRENT NEURAL NETWORKS

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ABSTRACT

Recurrent Neural Networks (RNNs) are powerful models for sequential data that have the potential to learn long-term dependencies. However, they are computationally expensive to train and difficult to parallelize. Recent work has shown that normalizing intermediate representations of neural networks can significantly improve convergence rates in feed-forward neural networks [1]. In particular, batch normalization, which uses mini-batch statistics to standardize features, was shown to significantly reduce training time. In this paper, we investigate how batch normalization can be applied to RNNs. We show for both a speech recognition task and language modeling that the way we apply batch normalization leads to a faster convergence of the training criterion but doesn't seem to improve the generalization performance.

Index Terms— batch normalization, RNN, LSTM, optimization

1. INTRODUCTION

Recurrent Neural Networks (RNNs) have received renewed interest due to their recent success in various domains, including speech recognition [2], machine translation [3, 4] and language modelling [5]. The so-called Long Short-Term Memory (LSTM) [6] type RNN has been particularly successful. Often, it seems beneficial to train deep architectures in which multiple RNNs are stacked on top of each other [2]. Unfortunately, the training cost for large datasets and deep architectures of stacked RNNs can be prohibitively high, often times an order of magnitude greater than simpler models like n -grams [7]. Because of this, recent work has explored methods for parallelizing RNNs across multiple graphics cards (GPUs). In [3], an LSTM type RNN was distributed layer-wise across multiple GPUs and in [8] a bidirectional RNN was distributed across time. However, due to the sequential nature of RNNs, it is difficult to achieve linear speed ups relative to the number of GPUs.

Another way to reduce training times is through a better conditioned optimization procedure. Standardizing or

whitening of input data has long been known to improve the convergence of gradient-based optimization methods [9]. Extending this idea to multi-layered networks suggests that normalizing or whitening intermediate representations can similarly improve convergence. However, applying these transforms would be extremely costly. In [1], batch normalization was used to standardize intermediate representations by approximating the population statistics using sample-based approximations obtained from small subsets of the data, often called mini-batches, that are also used to obtain gradient approximations for stochastic gradient descent, the most commonly used optimization method for neural network training. It has also been shown that convergence can be improved even more by whitening intermediate representations instead of simply standardizing them [10]. These methods reduced the training time of Convolutional Neural Networks (CNNs) by an order of magnitude and additionally provided a regularization effect, leading to state-of-the-art results in object recognition on the ImageNet dataset [11]. In this paper, we explore how to leverage normalization in RNNs and show that training time can be reduced.

2. BATCH NORMALIZATION

In optimization, feature standardization or whitening is a common procedure that has been shown to reduce convergence rates [9]. Extending the idea to deep neural networks, one can think of an arbitrary layer as receiving samples from a distribution that is shaped by the layer below. This distribution changes during the course of training, making any layer but the first responsible not only for learning a good representation but also for adapting to a changing input distribution. This distribution variation is termed *Internal Covariate Shift*, and reducing it is hypothesized to help the training procedure [1].

To reduce this internal covariate shift, we could whiten each layer of the network. However, this often turns out to be too computationally demanding. Batch normalization [1] approximates the whitening by standardizing the intermediate representations using the statistics of the current mini-batch. Given a mini-batch \mathbf{x} , we can calculate the sample mean and sample variance of each feature k along the mini-batch axis

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$$\bar{\mathbf{x}}_k = \frac{1}{m} \sum_{i=1}^m \mathbf{x}_{i,k}, \quad (1)$$

$$\sigma_k^2 = \frac{1}{m} \sum_{i=1}^m (\mathbf{x}_{i,k} - \bar{\mathbf{x}}_k)^2, \quad (2)$$

where m is the size of the mini-batch. Using these statistics, we can standardize each feature as follows

$$\hat{\mathbf{x}}_k = \frac{\mathbf{x}_k - \bar{\mathbf{x}}_k}{\sqrt{\sigma_k^2 + \epsilon}}, \quad (3)$$

where ϵ is a small positive constant to improve numerical stability.

However, standardizing the intermediate activations reduces the representational power of the layer. To account for this, batch normalization introduces additional learnable parameters γ and β , which respectively scale and shift the data, leading to a layer of the form

$$BN(\mathbf{x}_k) = \gamma_k \hat{\mathbf{x}}_k + \beta_k. \quad (4)$$

By setting γ_k to σ_k and β_k to $\bar{\mathbf{x}}_k$, the network can recover the original layer representation. So, for a standard feedforward layer in a neural network

$$\mathbf{y} = \phi(\mathbf{W}\mathbf{x} + \mathbf{b}), \quad (5)$$

where \mathbf{W} is the weights matrix, \mathbf{b} is the bias vector, \mathbf{x} is the input of the layer and ϕ is an arbitrary activation function, batch normalization is applied as follows

$$\mathbf{y} = \phi(BN(\mathbf{W}\mathbf{x})). \quad (6)$$

Note that the bias vector has been removed, since its effect is cancelled by the standardization. Since the normalization is now part of the network, the back propagation procedure needs to be adapted to propagate gradients through the mean and variance computations as well.

At test time, we can't use the statistics of the mini-batch. Instead, we can estimate them by either forwarding several training mini-batches through the network and averaging their statistics, or by maintaining a running average calculated over each mini-batch seen during training.

3. RECURRENT NEURAL NETWORKS

Recurrent Neural Networks (RNNs) extend Neural Networks to sequential data. Given an input sequence of vectors $(\mathbf{x}_1, \dots, \mathbf{x}_T)$, they produce a sequence of hidden states $(\mathbf{h}_1, \dots, \mathbf{h}_T)$, which are computed at time step t as follows

$$\mathbf{h}_t = \phi(\mathbf{W}_h \mathbf{h}_{t-1} + \mathbf{W}_x \mathbf{x}_t), \quad (7)$$

where \mathbf{W}_h is the recurrent weight matrix, \mathbf{W}_x is the input-to-hidden weight matrix, and ϕ is an arbitrary activation function.

If we have access to the whole input sequence, we can use information not only from the past time steps, but also from the future ones, allowing for bidirectional RNNs [12]

$$\vec{\mathbf{h}}_t = \phi(\vec{\mathbf{W}}_h \vec{\mathbf{h}}_{t-1} + \vec{\mathbf{W}}_x \mathbf{x}_t), \quad (8)$$

$$\overleftarrow{\mathbf{h}}_t = \phi(\overleftarrow{\mathbf{W}}_h \overleftarrow{\mathbf{h}}_{t+1} + \overleftarrow{\mathbf{W}}_x \mathbf{x}_t), \quad (9)$$

$$\mathbf{h}_t = [\vec{\mathbf{h}}_t : \overleftarrow{\mathbf{h}}_t], \quad (10)$$

where $[\mathbf{x} : \mathbf{y}]$ denotes the concatenation of \mathbf{x} and \mathbf{y} . Finally, we can stack RNNs by using \mathbf{h} as the input to another RNN, creating deeper architectures [13]

$$\mathbf{h}_t^l = \phi(\mathbf{W}_h \mathbf{h}_{t-1}^l + \mathbf{W}_x \mathbf{h}_t^{l-1}). \quad (11)$$

In vanilla RNNs, the activation function ϕ is usually a sigmoid function, such as the hyperbolic tangent. Training such networks is known to be particularly difficult, because of vanishing and exploding gradients [14].

3.1. Long Short-Term Memory

A commonly used recurrent structure is the Long Short-Term Memory (LSTM). It addresses the vanishing gradient problem commonly found in vanilla RNNs by incorporating gating functions into its state dynamics [6]. At each time step, an LSTM maintains a hidden vector \mathbf{h} and a cell vector \mathbf{c} responsible for controlling state updates and outputs. More concretely, we define the computation at time step t as follows [15]:

$$\mathbf{i}_t = \text{sigmoid}(\mathbf{W}_{hi} \mathbf{h}_{t-1} + \mathbf{W}_{xi} \mathbf{x}_t) \quad (12)$$

$$\mathbf{f}_t = \text{sigmoid}(\mathbf{W}_{hf} \mathbf{h}_{t-1} + \mathbf{W}_{xf} \mathbf{x}_t) \quad (13)$$

$$\mathbf{c}_t = \mathbf{f}_t \odot \mathbf{c}_{t-1} + \mathbf{i}_t \odot \tanh(\mathbf{W}_{hc} \mathbf{h}_{t-1} + \mathbf{W}_{xc} \mathbf{x}_t) \quad (14)$$

$$\mathbf{o}_t = \text{sigmoid}(\mathbf{W}_{ho} \mathbf{h}_{t-1} + \mathbf{W}_{hx} \mathbf{x}_t + \mathbf{W}_{co} \mathbf{c}_t) \quad (15)$$

$$\mathbf{h}_t = \mathbf{o}_t \odot \tanh(\mathbf{c}_t) \quad (16)$$

where $\text{sigmoid}(\cdot)$ is the logistic sigmoid function, \tanh is the hyperbolic tangent function, \mathbf{W}_h are the recurrent weight matrices and \mathbf{W}_x are the input-to-hidden weight matrices. \mathbf{i}_t , \mathbf{f}_t and \mathbf{o}_t are respectively the input, forget and output gates, and \mathbf{c}_t is the cell.

4. BATCH NORMALIZATION FOR RNNs

From equation 6, an analogous way to apply batch normalization to an RNN would be as follows:

$$\mathbf{h}_t = \phi(BN(\mathbf{W}_h \mathbf{h}_{t-1} + \mathbf{W}_x \mathbf{x}_t)). \quad (17)$$

However, in our experiments, when batch normalization was applied in this fashion, the model failed to learn. In unnormalized RNNs, the tied nature of the recurrent weight matrix \mathbf{W}_h makes optimization difficult since small changes

drastically change the solution due to the repeated application across time steps. Similarly, when normalizing the hidden state, the standardization procedure not only affects the current time step but all future time steps, which, in turn, are also standardized. Thus, the sequential nature of the hidden state makes normalization much more difficult than in CNNs. So we propose to apply batch normalization only to the input-to-hidden transition ($\mathbf{W}_x \mathbf{x}_t$), i.e. as follows:

$$\mathbf{h}_t = \phi(\mathbf{W}_h \mathbf{h}_{t-1} + \text{BN}(\mathbf{W}_x \mathbf{x}_t)). \quad (18)$$

This idea is similar to the way dropout [16] can be applied to RNNs [17]: batch normalization is applied only on the vertical connections (i.e. from one layer to another) and not on the horizontal connections (i.e. within the recurrent layer). We use the same principle for LSTMs: batch normalization is only applied after multiplication with the input-to-hidden weight matrices \mathbf{W}_x .

4.1. Frame-wise and Sequence-wise Normalization

In experiments where we don't have access to the future frames, like in language modelling where the goal is to predict the next character, we are forced to compute the normalization at each time step

$$\hat{\mathbf{x}}_{k,t} = \frac{\mathbf{x}_{k,t} - \bar{\mathbf{x}}_{k,t}}{\sqrt{\sigma_{k,t}^2 + \epsilon}}. \quad (19)$$

We'll refer to this as *frame-wise normalization*.

In applications like speech recognition, we usually have access to the entire sequences. However, those sequences may have variable length. Usually, when using mini-batches, the smaller sequences are padded with zeroes to match the size of the longest sequence of the mini-batch. In such setups we can't use frame-wise normalization, because the number of unpadded frames decreases along the time axis, leading to increasingly poorer statistics estimates. To solve this problem, we apply a sequence-wise normalization, where we compute the mean and variance of each feature along both the time and batch axis using

$$\bar{\mathbf{x}}_k = \frac{1}{n} \sum_{i=1}^m \sum_{t=1}^T \mathbf{x}_{i,t,k}, \quad (20)$$

$$\sigma_k^2 = \frac{1}{n} \sum_{i=1}^m \sum_{t=1}^T (\mathbf{x}_{i,t,k} - \bar{\mathbf{x}}_k)^2, \quad (21)$$

where T is the length of each sequence and n is the total number of unpadded frames in the mini-batch. We'll refer to this type of normalization as *sequence-wise normalization*.

5. EXPERIMENTS

We ran experiments on a speech recognition task and a language modelling task. The models were implemented using Theano [18] and Blocks [19].

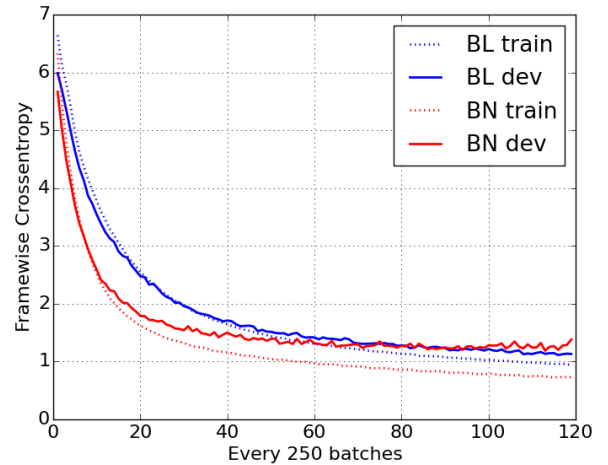


Fig. 1. Frame-wise cross entropy on WSJ for the baseline (blue) and batch normalized (red) networks. The dotted lines are the training curves and the solid lines are the validation curves.

Model	Train		Valid	
	FCE	FER	FCE	FER
BiRNN	0.95	0.28	1.11	0.33
BiRNN (BN)	0.73	0.22	1.19	0.34

Table 1. Best framewise cross entropy (FCE) and frame error rate (FER) on the training and development sets for both networks.

5.1. Speech Alignment Prediction

For the speech task, we used the Wall Street Journal (WSJ) [21] speech corpus. We used the si284 split as training set and evaluated our models on the dev93 development set. The raw audio was transformed into 40 dimensional log mel filter-banks (plus energy), with deltas and delta-deltas. As in [22], the forced alignments were generated from the Kaldi recipe tri4b, leading to 3546 clustered triphone states. Because of memory issues, we removed from the training set the sequences that were longer than 1300 frames (4.6% of the set), leading to a training set of 35746 sequences.

The baseline model (BL) is a stack of 5 bidirectional LSTM layers with 250 hidden units each, followed by a size 3546 softmax output layer. All the weights were initialized using the Glorot [23] scheme and all the biases were set to zero. For the batch normalized model (BN) we applied sequence-wise normalization to each LSTM of the baseline model. Both networks were trained using standard SGD with momentum, with a fixed learning rate of 1e-4 and a fixed momentum factor of 0.9. The mini-batch size is 24.

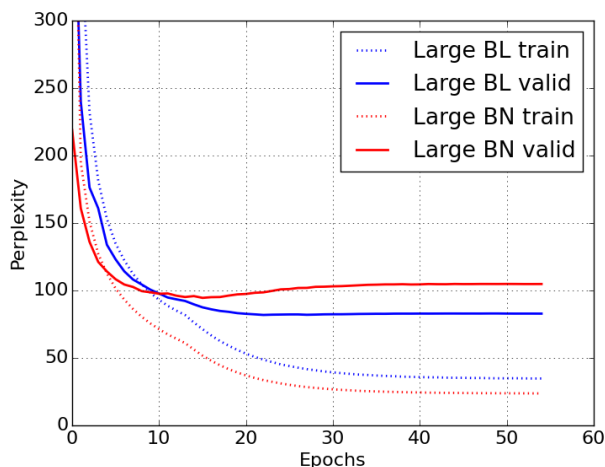


Fig. 2. Large LSTM on Penn Treebank for the baseline (blue) and the batch normalized (red) networks. The dotted lines are the training curves and the solid lines are the validation curves.

5.2. Language Modeling

We used the Penn Treebank (PTB) [20] corpus for our language modeling experiments. We use the standard split (929k training words, 73k validation words, and 82k test words) and vocabulary of 10k words. We train a small, medium and large LSTM as described in [17].

All models consist of two stacked LSTM layers and are trained with stochastic gradient descent (SGD) with a learning rate of 1 and a mini-batch size of 32.

The small LSTM has two layers of 200 memory cells, with parameters being initialized from a uniform distribution with range $[-0.1, 0.1]$. We back propagate across 20 time steps and the gradients are scaled according to the maximum norm of the gradients whenever the norm is greater than 10. We train for 15 epochs and halve the learning rate every epoch after the 6th.

The medium LSTM has a hidden size of 650 for both layers, with parameters being initialized from a uniform distribution with range $[-0.05, 0.05]$. We apply dropout with probability of 50% between all layers. We back propagate across 35 time steps and gradients are scaled according to the maximum norm of the gradients whenever the norm is greater than 5. We train for 40 epochs and divide the learning rate by 1.2 every epoch after the 6th.

The Large LSTM has two layers of 1500 memory cells, with parameters being initialized from a uniform distribution with range $[-0.04, 0.04]$. We apply dropout between all layers. We back propagate across 35 time steps and gradients are scaled according to the maximum norm of the gradients whenever the norm is greater than 5. We train for 55 epochs and divide the learning rate by 1.15 every epoch after the 15th.

Model	Train	Valid
Small LSTM	78.5	119.2
Small LSTM (BN)	62.5	120.9
Medium LSTM	49.1	89.0
Medium LSTM (BN)	41.0	90.6
Large LSTM	49.3	81.8
Large LSTM (BN)	35.0	97.4

Table 2. Best perplexity on training and development sets for LSTMs on Penn Treebank.

6. RESULTS AND DISCUSSION

Figure 1 shows the training and development framewise cross entropy curves for both networks of the speech experiments. As we can see, the batch normalized networks trains faster (at some points about twice as fast as the baseline), but overfits more. The best results, reported in table 1, are comparable to the ones obtained in [22].

Figure 2 shows the training and validation perplexity for the large LSTM network of the language experiment. We can also observe that the training is faster when we apply batch normalization to the network. However, it also overfits more than the baseline version. The best results are reported in table 2.

For both experiments we observed a faster training and a greater overfitting when using our version of batch normalization. This last effect is less prevalent in the speech experiment, perhaps because the training set is way bigger, or perhaps because the frame-wise normalization is less effective than the sequence-wise one. It can also be caused by the experimental setup: in the language modeling task we predict one character at a time, whereas we predict the whole sequence in the speech experiment.

Batch normalization also allows for higher learning rates in feed-forward networks, however since we only applied batch normalization to parts of the network, higher learning rates didn't work well because they affected un-normalized parts as well.

Our experiments suggest that applying batch normalization to the input-to-hidden connections in RNNs can improve the conditioning of the optimization problem. Future directions include whitening input-to-hidden connections [10] and normalizing the hidden state instead of just a portion of the network.

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