A Passive Similarity based CNN Filter Pruning for Efficient Acoustic Scene Classification

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Abstract

We present a method to develop low-complexity convolutional neural networks (CNNs) for acoustic scene classification (ASC). The large size and high computational complexity of typical CNNs is a bottleneck for their deployment on resource-constrained devices. We propose a passive filter pruning framework, where a few convolutional filters from the CNNs are eliminated to yield compressed CNNs. Our hypothesis is that similar filters produce similar responses and give redundant information allowing such filters to be eliminated from the network. To identify similar filters, a cosine distance based greedy algorithm is proposed. A fine-tuning process is then performed to regain much of the performance lost due to filter elimination. To perform efficient fine-tuning, we analyze how the performance varies as the number of fine-tuning training examples changes. An experimental evaluation of the proposed framework is performed on the publicly available DCASE 2021 Task 1A baseline network trained for ASC. The proposed method is simple, reduces computations per inference by 27%, with 25% fewer parameters, with less than 1% drop in accuracy.

Index Terms: Acoustic scene classification, convolutional neural networks, pruning.

1. Introduction

Acoustic scene classification (ASC) frameworks utilise sounds produced in the underlying environment to classify sounds into pre-defined scene classes [1]. Typically, ASC frameworks can be deployed on resource-constrained devices such as smart phones, particularly when millions of smart devices are connected in the network [2, 3, 4]. Therefore, reducing power consumption and memory storage of ASC frameworks is an important task.

Research in ASC actively started when the detection and classification of acoustic scenes and events (DCASE) community released the acoustic scenes dataset publicly and organised a challenge in 2013 [5]. Initially, DCASE challenges [6, 7, 8] focused on designing ASC frameworks to improve performance under different recording locations and using various recording devices. Recent DCASE challenges [9, 10] for ASC have also focused on designing low-complexity solutions to reduce computational complexity and memory storage.

Typically, learning-based methods have shown promising results compared to hand-crafted methods for ASC. Many learning-based methods use convolutional neural networks (CNNs) [11], which give state-of-the-art performance in several related tasks such as audio classification [12] and image scene classification [13]. However, CNNs are resource hungry due to their large size and high computations [14]. Also, CNNs may have redundancy in their architecture that lie in their redundant parameters that include weights or convolutional filters [15].

An illustration of similar convolutional filters where similarity is measured using cosine distance in filter representative space is shown in Figure 1. Such similar filters mostly contribute to redundancy in CNNs [16] and hence, can be eliminated. Eliminating redundancy in CNNs can reduce the number of parameters or memory storage and speed-up CNNs during their inference as well [17]. In addition, the carbon footprint generated during training of CNNs reduces as training time per epoch decreases in compressed CNNs [18]. A few of the important convolutional filters selected using Algorithm 1 proposed in Section 3 are also shown in Figure 1.

The rest of this paper is organised as follows. Section 2 gives a brief overview on filter pruning methods to remove CNN parameters. The proposed framework to identify similar filters is described in Section 3. Next, Section 4 includes experimental setup and Section 5 presents results and analysis. Finally, the discussion and conclusion is presented in Section 6 and Section 7 respectively.

2. Filter Pruning Methods

Filter pruning methods [19, 20] have been widely employed to eliminate convolutional filters based on their importance and to yield a pruned network. The importance can be measured either in an active manner that involves a dataset [16, 19, 21, 22, 23] or in a passive manner [20, 24] that does not involve a dataset. Mostly, active filter pruning methods measure entropy [25], variance [26] or similarity [16] in the outputs produced by each filter across dataset to quantify their importance. On the other
hand, passive filter pruning methods [20, 27] rely only on convolution filters to quantify their importance without involving any dataset, and hence are relatively simpler than active filter pruning methods.

Majority of the existing passive filter pruning methods are norm-based that uses e.g. $l_1$-norm [20] of the filters to quantify their importance. A filter with relatively low norm is considered to be less important. After ranking filters based on their importance, few filters are eliminated based on a user-defined pruning ratio\(^1\). Eliminating filters reduce performance of the pruned network, which is regained by performing a fine-tuning step.

The efficacy of passive norm-based filter pruning methods in defining importance of each filter depend on two conditions: (a) the minimum norm of the filter should be close to zero, and (b) the difference between the minimum and maximum norm of the filters should be as large as possible. However, such conditions are not always satisfied [28].

This paper proposes a passive filter pruning approach to compress any pre-trained convolutional neural networks designed for ASC. The proposed approach eliminates convolutional filters based on their similarity rather than considering filter norm, and does not require any user-defined pruning ratio.

Our hypothesis is that few of the convolutional filters in CNNs are similar and such filters give mostly redundant information. Therefore, one of the similar filters can be eliminated from the network without much loss of performance. In the proposed approach, we consider the relationship among convolutional filters by measuring their cosine-similarity to quantify their importance and proposes an Algorithm 1 to select few important convolutional filters.

We also analyse how many training examples are sufficient in the fine-tuning process to achieve similar performance to that obtained using all training examples. The major contributions of this paper are summarised as follows:

• We propose a passive filter pruning framework which removes similar filters from a pre-trained CNN without any requirement of user-defined pruning ratio.

• We show that a pruned network obtained using the proposed pruning outperforms an existing $l_1$-norm based pruning method.

• We show that a reduction of training examples by 25% during fine-tuning process gives a similar performance with reduced training time.

3. Proposed Method

Consider a convolutional layer in a CNN which has $n$ filters denoted by $F_i, 1 \leq i \leq n$. Each filter has the same size with width $w$, height $h$ and number of channels $c$. Given the $n$ filters from the convolutional layer, our aim is to identify similar filters and select few important filters.

First, each filter is transformed into a 2-D matrix of size $(wh \times c)$, where each column is obtained by stacking the width and height parameters. Next, we find Rank-1 approximation of each filter by performing singular value decomposition (SVD) that gives best Rank-1 approximation with respect to the Frobenius norm.

A filter $F_i$ is approximated as $F_i \approx \sigma_1 u_1 v_1^T$, where $\sigma_1$ denotes the significant singular value, $u_1$ denotes first left singular vector and $v_1$ denotes first right singular vector. Next, $F_i$ is normalised column-wise and any column from the normalised $F_i$ is chosen that acts as a “filter representative”. $f_i$.

It is important to note that each column in the normalised $F_i$ has unit rank. After obtaining filter representatives corresponding to each filter, we compute the pairwise cosine distance between filter representatives. This gives a similarity matrix, $W \in \mathbb{R}^{n \times n}$. Now, given the similarity matrix, a greedy Algorithm 1 is proposed to identify important filters.

To identify important filters, we select the closest pair for each filter representative using $W$, and sorted each closest pair according to their distance in the order of low to high. Here, the cosine distance between the closest pair defines their importance.

Next, we define one of the filter representative as important or redundant from each closest pair starting from the least important closest filter pair. A filter representative (the first index) from the least important closest pair is chosen as an important and other as a redundant. This procedure is repeated for each closest pair in the order of their importance except for the closest filter pair containing already identified redundant filter representative. Finally, a set of important filter representatives are obtained after ignoring all redundant filter representatives. An illustration of the important filters selected using Algorithm 1 is shown in Figure 1.

The redundant filters obtained using Algorithm 1 are explicitly eliminated form the unpruned network to yield a pruned network. The pruned network thus obtained has reduced number of parameters, less computational complexity and loss in performance due to elimination of few filters. To regain lost performance, a fine-tuning step is performed which involves a training dataset to re-train the pruned network. The proposed pruning algorithm can be found at: https://github.com/Arshdeep-Singh-Boparai/passive-filter-pruning-Interspeech22.

\footnote{The percentage of number of filters to be eliminated.}

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**Algorithm 1: Identification of important filters**

**Data:** Pair-wise cosine distance of $n$ filters ($W \in \mathbb{R}^{n \times n}$).

**Result:** Indices of important filters ($\text{Imp} \text{list}$)

\[
\begin{align*}
\text{A} & = [], \quad \text{Imp} \text{list} = [], \quad \text{Red} \text{list} = [] \\
\text{for} \quad i \leq n \text{ do} & \\
\quad [m, d] = \arg\min \{ W[i,j,i\in \{1\}] \}; & /* \text{Identify the closest filter} m \\
\quad \text{to} \ i^{\text{th}} \ \text{filter} \ \text{with their} \ \text{distance} \ d. */ & \\
\quad \text{A}.\text{append}(i, m, d); & \\
\quad \end{align*}
\]

\[
\begin{align*}
\text{A}_\text{sort} & = \text{Sort} (\text{A}); & /* \text{Sort} \ A \ \text{based on the} \\
\text{distance} \ d. */ & \\
\text{for} \quad i \leq \text{len}\left(\text{A}\right) \text{ do} & \\
\quad \text{index}\_\text{imp} = \text{A}_\text{sort}[i][0]; & /* \text{important} \\
\quad \text{index}\_\text{red} = \text{A}_\text{sort}[i][1]; & /* \text{redundant} \\
\quad \text{filter index} */ & \\
\quad \text{if} \ \text{index}\_\text{imp} \notin \text{Red} \text{list} \text{ then} & \\
\quad \quad \text{Imp} \text{list} \text{.append}\left(\text{index}\_\text{imp}\right); & \\
\quad \quad \text{Red} \text{list} \text{.append}\left(\text{index}\_\text{red}\right) & \\
\quad \quad \end{align*}
\]

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We evaluate the proposed method on the publicly available DCASE 2021 Task1A baseline [10] network to this we refer as the unpruned network that accepts log-mel energies of size $40 \times 500$ extracted from 10-second audio signals as input. The unpruned network has three convolutional layers ($C_1$, $C_2$, $C_3$) with 16, 16 and 32 number of filters respectively. These are followed by a dense layer of 100 units and a classification layer of 10 units.

The unpruned network is trained on TAU Urban Acoustic Scenes 2020 Mobile [9] development training dataset to classify 10 acoustic scenes recorded using multiple devices for 200 epochs using Adam optimizer with learning rate 0.001. The unpruned network gives 48.58% accuracy on validation dataset, has 46246 parameters and requires 286M multiply-accumulate operations (MACs) per inference to produce an output corresponding to 10-second audio signal. For fine-tuning the pruned network, we opt similar conditions as used in training the unpruned network except 30 fine-tuning epochs, which are reduced by approximately 7x than that used in training the unpruned network.

The effectiveness of the proposed pruning method is measured in terms of accuracy obtained with or without fine-tuning, reduced MACs per inference and reduced number of parameters in the pruned network. To report performance after fine-tuning, we fine-tune the pruned network independently for 5 times and reported average accuracy.

The proposed pruning method is compared to an $l_1$-norm based pruning method [20]. The $l_1$-norm based method ranks convolutional filters according to their norm and requires user-defined pruning ratio. Therefore, we choose the same pruning ratio for $l_1$-norm method as obtained using our proposed method. The experiments are performed on GeForce RTX 2070 with Max-Q Design computing device.

Comparison with $l_1$-norm based pruning: The $l_1$-norm of the filters in $C_2$ layer is shown in Figure 4. It can be observed that the top-5 filters with relatively low norm may be still significant as the ratio of the lowest norm to highest norm is greater than 84%. This verifies that the norm of the filter alone may not be a good metric to define filter importance.

On the other hand, our proposed method does not use norm of the filters and rely on similarity among filters to select filters.
It can be seen in Figure 4 that our method selects filters with indices \{4,5,1,7\} which are ignored by \(l_1\)-norm method.

Choosing filters based on similarity improves performance of the pruned network as compared to norm-based method. It can be observed from Figure 5 that the accuracy obtained using the proposed similarity-based pruning is better than that of norm-based pruning across various layers. In particular, the proposed method improves accuracy by \((1-1.5)\%\), when the number of reduced parameters are more.

Analysis of fine-tuning at different training examples: Next, we analyse the computational complexity during fine-tuning and the accuracy obtained after fine-tuning the pruned network by varying training examples from 10% to 100% that is shown in Figure 6. The pruned network is obtained by pruning (C1+C2+C3) layers using the proposed method or \(l_1\)-norm method. Also, we fine-tune a pruned network where the parameters of the pruned network are initialized randomly. Without performing any fine-tuning, the accuracy of the pruned network obtained using the proposed pruning method is approximately 6 percentage points better than that of \(l_1\)-norm method or randomly initialized network. Also, the proposed pruning framework gives similar or better performance than \(l_1\)-norm method and randomly initialized network after performing fine-tuning. It is also interesting to note that the randomly initialized network performs similar to the pruned network obtained using the proposed pruning method when more training dataset \((\geq 75\%)\) is used in fine-tuning.

The pruned network significantly improves performance from 16% to 40% at 12x reduction in training time when fine-tuned with only 10% training dataset in contrast to that of using all training dataset. Utilizing 75% training dataset in fine-tuning the pruned network gives similar performance as that of all training dataset with 1.33x (25%) reduction in training time.

6. Discussion

The proposed pruning method produces a low-complexity pruned network that achieves similar performance as that of the unpruned network after performing fine-tuning for only 30 epochs.

Pruning filters using the proposed method gives a set of important filters with improved performance in contrast to that of \(l_1\)-norm method. The proposed method is particularly advantageous when more parameters are eliminated. Moreover, the proposed method automatically selects number of important filters and does not require any user-defined pruning ratio.

We compare other methods such as Chebyshev distance and average filters to compute important set of filters in the pruned network (C1+C2+C3) for analysing the effectiveness of the proposed pruning method. In Chebyshev distance method, we use the Chebyshev distance which is defined as the maximum of differences along any filter representative dimension to generate similarity matrix in Algorithm 1. In average filters method, we compute closest filter pairs based on cosine distance, and generate new filters by averaging the closest filter pairs.

The accuracy of the pruned network without performing any fine-tuning using the proposed method outperforms the other methods as given in Table 1. The performance obtained after averaging the similar filters is more or less random. In contrast to Chebyshev distance, the cosine distance shows promising result.

7. Conclusions and Future work

We propose a cosine distance based passive filter pruning framework to compress a pre-trained convolutional neural network. The proposed framework identifies pruning ratio automatically. We find that considering pair-wise relationships among filters yield a better set of important filters and improves performance compared to norm-based filters selection. We also observe that fine-tuning the pruned network for few epochs and using few training examples improves the performance of the pruned network significantly. Our future goal is to design better distance metrics to measure pair-wise similarity among filters and to reduce the performance loss in the pruned network with minimal requirement of fine-tuning.

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Table 1: Accuracy of the pruned network obtained after pruning (C1+C2+C3) layers without performing any fine-tuning when different methods are used to select important filters.

| Method to obtain Pruned network (C1+C2+C3) | Accuracy |
|------------------------------------------|----------|
| Proposed cosine distance based method    | 16.23%   |
| Chebyshev distance based method          | 12.23%   |
| Average filters                          | 10%      |
| \(l_1\)-norm method                      | 9.84%    |
| Randomly initialized network             | 10%      |

Figure 6: Accuracy and training time required to fine-tune by 30 epochs after pruning (C1+C2+C3) layers, when the number of training examples are varied from 10% to 100%.
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