Emotion Invariant Speaker Embeddings for Speaker Identification with Emotional Speech

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Abstract—Emotional state of a speaker is found to have significant effect in speech production, which can deviate speech from that arising from neutral state. This makes identifying speakers with different emotions a challenging task as generally the speaker models are trained using neutral speech. In this work, we propose to overcome this problem by creation of emotion invariant speaker embedding. We learn an extractor network that maps the test embeddings with different emotions obtained using i-vector based system to an emotion invariant space. The resultant test embeddings thus become emotion invariant and thereby compensate the mismatch between various emotional states. The studies are conducted using four different emotion classes from IEMOCAP database. We obtain an absolute improvement of 2.6% in accuracy for speaker identification studies using emotion invariant speaker embedding against average speaker model based framework with different emotions.

1. INTRODUCTION

Identifying speakers with their unique traits such as voice is often affected by both external and internal factors [1]–[4]. The external factors include environmental noise, channel/session effects and sensor mismatch [5]–[8]. On the contrary, the internal factors refer to the health and emotional state of speakers [9], [10]. Most of the studies consider neutral speech to recognize speakers and the performance of such systems deviates in presence of external as well as internal factors [11]–[16].

The internal source mismatch based studies have gained interest of the community over the time as they are mostly related to human computer interactions and criminal investigations, where there is not much control over the mode of speech [15]–[19]. In addition, presence of emotion in speech is very common and natural, which cannot be avoided in practical scenarios [20]. A study [21] shows that there is more than 37% of emotional speech present in the data collected from realistic environment. This showcases the need to investigate speaker identification using emotional speech [22].

The earlier attempts in speaker identification with emotional speech include normalization and transformation of features from emotional speech to that of neutral speech [23], [24]. The mel frequency cepstral coefficient (MFCC) features computed from the emotional speech using auto associative neural network are transformed to be used in a Gaussian mixture model (GMM) based speaker recognition system. Another work studied an emotion dependent score normalization technique for speaker verification [3]. The GMM-universal background model (UBM) based speaker verification system performance was reported to be improved for emotional speech after score normalization. The authors of [25] performed speaker recognition using model space migration through translated learning. In [26], a two-staged approach with hidden Markov model (HMM) was proposed for emotion dependant speaker verification, and later effectiveness of suprasegmental HMMs was shown in [27]. Again, the authors of [28] converted the emotional speech to neutral for speaker recognition studies.

Most of the prior works to recognize speakers with emotional speech rely on waveform or feature conversion and score normalization techniques. Further, they did not consider advanced techniques such as i-vectors and x-vectors for speaker modeling [29], [30]. A few studies investigated speaker recognition with emotional speech using i-vectors [31], [32]. Similarly, the authors of [33] predicted the i-vector based speaker recognition by classifying the speech into broad emotion classes such as arousal and valence. However, these works with latest systems did not focus much on improving the performance of identifying speakers with emotional speech. This shows the requirement of having robust methods for compensating such mismatch with latest systems, which motivated our current work.

The speaker embedding representation such as i-vectors and x-vectors are used widely for speaker recognition studies [29], [30]. Such embeddings capture dominant speaker information that are robust to various external factors like channel/session variation and background noise. We believe it is more preferable to compensate the mismatch due to internal factors such as emotions at the embedding level, rather than at feature or waveform level as that may affect the performance due to external factors. In this regard, we propose an emotion invariant speaker embedding that is independent from emotional effect. These are extracted by using an extractor that is learned using examples obtained through data augmentation at the speaker embedding level. We use i-vector based modeling as a reference system to obtain the speaker embeddings in this study [29]. The extracted i-vectors from different emotions are transformed into an emotion invariant space to normalize the emotion specific information present in them.

The studies in this work are conducted using IEMOCAP database, which is a standard corpus for emotional speech research [34]. Four emotion classes, namely, happiness, anger, sadness and neutral are considered for the study. We note
that the background models are learned using neutral speech of a large number of speakers with standard well known datasets for speaker recognition. The contribution of this work lies in proposal of emotion invariant speaker embedding for identifying speakers with emotional speech.

II. EMOTION IN Variant Speaker Embedding

This section details the proposed emotion invariant speaker embedding for speaker identification. We discuss the i-vector based system, emotion invariant extractor and the framework for speaker identification in the following subsections.

A. i-vector based Speaker Representation

An i-vector is a compact representation for an utterance, which is derived by a factor analysis approach [29]. In this kind of system, the GMM mean supervector of an utterance is projected into a lower dimensional space by a transformation matrix. This transformation matrix, in short T-matrix is learned with a large amount of background speaker population that captures different variabilities present in the speech signal. The low dimensional representation of an utterance thus obtained is referred to as i-vector. However, as it contains channel/session information, some compensation techniques are applied to nullify those unwanted information. In this work, we have chosen linear discriminant analysis (LDA) [35] and within class covariance normalization (WCCN) [36] for such compensation on top of i-vectors as given in [29].

B. Emotion Invariant Extractor

In general, speaker enrollments can be done with neutral speech. However, test speech can contain any emotional state in practical scenarios. Therefore, we plan to learn an extractor that maps the i-vector based speaker embeddings of different emotions to that of neutral state. We refer the speaker embedding thus produced as emotion invariant speaker embedding.

Figure 1 shows the training process of the emotion invariant extractor. A deep neural network with three hidden layers, namely, encoder, bottleneck and decoder are used in the process. The i-vectors of neutral or any emotional speech from a particular speaker is used as input and that of neutral speech belonging to the same speaker is used as the target to train the network. The input dimension equals to the i-vector dimension. A data augmentation process is used to increase the number of training i-vectors of different emotions. The details of this process is described in Section III-D.

We used linear activation function at the final layer as the i-vectors can have any real values. The ReLU activation function is used for all the layers except the final layer. Mean squared error is used as the loss function and Adam optimizer is used for optimizing the network [37]. The model is trained for 20 epochs with batch size of 256. The various parameters used in the network are summarized in Table I. We note that $\beta_1$ and $\beta_2$ are parameters of Adam optimizer used in the network.

C. Speaker Identification with Emotional Speech

For a given speech irrespective of any emotion, its i-vector is obtained, followed by extraction of corresponding emotion invariant speaker embedding. We use this process with two different ways in terms of considering the speaker models. The first one refers to EINV-Test that extracts emotion invariant speaker embeddings only for the test data and uses the raw train i-vectors as speaker models. On the other hand, the second one is referred to as EINV-Pair that extracts emotion invariant speaker embeddings for both train and test data. It is noted that the baseline system follows the standard i-vector pipeline without any emotion invariant extractor.

III. EXPERIMENTS

In this section, we discuss the details of the corpus, feature extraction, experimental setup for baseline and proposed framework in the following subsections.

A. Database

The IEMOCAP database is considered for the studies in this work. It is an acted speech database from 10 different speakers comprising of 5 male and 5 female. The database is segmented and each segment is labeled with one of the emotion classes. We consider four emotion classes, namely, neutral, happiness,
anger and sadness for the study due to availability of sufficient amount of the data from those emotions in the corpus. First two minutes of each emotion class are used for the training and the rest for testing. The speaker models are created using the 2 minutes of training data, whereas the test data is divided into many non-overlapping 30 seconds segments and each of those segments are used for testing. A summary of the training and testing examples is shown in Table II. Further, we note that the Switchboard Corpus-II is used for learning the background models for the i-vector system.

B. Feature Extraction

The speech signals are short-term processed with 20 ms frame with a shift of 10 ms to extract 39-dimensional (13-base+13+13-ΔΔ) MFCC features [38]. An energy based voice activity detection is performed to consider the regions with sufficient voice activity [39]. The features of those regions are then normalized by cepstral mean and variance normalization to fit to zero and unit variance [40].

C. Experimental Setup: Baseline i-vector

The background data features from Switchboard-II corpus are used to train the background models for i-vector system. Equal amount of male and female data is chosen to train a 1024 component gender-independent UBM. The zeroth and the first order statistics of the background data are then extracted to learn the T-matrix of 400 columns. The features belonging to IEMOCAP corpus are then used to obtain the sufficient statistics followed by transformation using the T-matrix to derive the 400-dimensional i-vectors. Further, we have learned 150-dimensional LDA and full dimensional WCCN using the background data i-vectors. The LDA and WCCN are then applied on top of the i-vectors for channel/session compensation, which reduces their dimension to 150. For the studies with baseline system, we used the channel/session compensated i-vectors of train and test data to perform a cosine similarity scoring for identifying a speaker with the highest similarity.

D. Experimental Setup: Emotion Invariant Extractor

In order to create the emotion invariant extractor, sufficient amount of training data with emotional speech is required. The data should contain i-vectors of neutral or emotional speech at the input layer and i-vectors of neutral speech of the same speaker at the output layer of the network. Since we have only 2 minutes of training data per speaker for each emotion, data augmentation is performed to enhance the training.

First, the 2 minutes data is split into many overlapping 30 seconds segments at every 10 second interval. This resulted in 10 small segments per speaker for each emotion class. Total number of such training segments becomes 400 with 10 speakers and 4 emotion classes. The i-vectors for all such segments are extracted as described in the previous subsection. Next, data augmentation is performed by averaging i-vectors of two to five segments from the same speaker with different combinations. This augmentation is performed for producing both input and target i-vectors. In this way, 20,000 input and target i-vectors are generated for training. The details of the data augmentation process is shown in Figure 2. We note that the augmented i-vectors are divided into training and validation sets for learning the emotion invariant extractor; 80% of the data is used for training and rest 20% is used for validation.

The emotion invariant extractor thus learned, considers 150-dimensional input i-vectors from either neutral or any other emotional speech to generate corresponding emotion invariant speaker embedding. Once the emotion invariant speaker embedding is obtained, the remaining stages for speaker identification studies follows the pipeline described earlier. In this work, identification accuracy (in %) is used as metric to report the results of various frameworks. Next, we discuss the results and analysis for the studies.

IV. RESULTS AND ANALYSIS

We first evaluate the performance of baseline i-vector system for identifying speakers with emotional speech. Table III reports its performance for different train and test conditions using the four emotion classes. We observe that the average

Table II

Summary of the corpus with different emotion classes namely, neutral (N), happiness (H), anger (A) and sadness (S).

| Speaker Label | # Train Utterances | # Test Utterances |
|---------------|--------------------|-------------------|
| N | H | A | S | N | H | A | S |
| 01F | 1 | 1 | 1 | 1 | 20 | 10 | 25 | 16 |
| 02F | 1 | 1 | 1 | 1 | 29 | 10 | 15 | 22 |
| 02M | 1 | 1 | 1 | 1 | 20 | 10 | 10 | 18 |
| 03F | 1 | 1 | 1 | 1 | 35 | 7 | 10 | 22 |
| 03M | 1 | 1 | 1 | 1 | 16 | 13 | 16 | 25 |
| 04F | 1 | 1 | 1 | 1 | 26 | 9 | 22 | 23 |
| 04M | 1 | 1 | 1 | 1 | 1 | 6 | 30 | 13 |
| 05F | 1 | 1 | 1 | 1 | 23 | 8 | 16 | 19 |
| 05M | 1 | 1 | 1 | 1 | 25 | 10 | 11 | 22 |

Fig. 2. Block diagram showing the data augmentation procedure used in training of emotion invariant extractor.
accuracy is high when neutral emotion is used for speaker modeling. In addition, the accuracy is maximum when neutral speech considered in both training as well as testing and a comparatively a higher performance is obtained when the train and test emotions have a match. However, for sadness emotion, the performance is maximum when the training emotion is neutral. As a summary, we find that the overall performance is comparatively better when the training data contains neutral emotion speech from the speakers. This clearly supports the motivation behind the work, i.e., to map the i-vector based speaker representation of different emotions to that of the neutral speech to derive emotion invariant speaker embeddings.

We now pay our attention to the studies with proposed emotion invariant speaker embeddings, where the i-vectors of emotional speech are converted to that of the neutral speech. At this stage, we also build a contrast system for comparing to our proposed systems EINV-Test and EINV-Pair. The contrast system considers speaker models that contain information of multiple emotions, thereby showing scope for having a better match to different emotions during testing. In our case, we consider i-vector averaging of different emotions to derive the speaker models for the contrast system. It is to be noted that we also perform averaging of train i-vectors belonging to each speaker with our proposed EINV-Test and EINV-Pair frameworks.

Table IV shows the performance comparison of the contrast system and proposed emotion invariant speaker embedding system with two different frameworks EINV-Test and EINV-Pair. The contrast system considers speaker models that contain information of multiple emotions, thereby showing scope for having a better match to different emotions during testing. We note the improvements with EINV-Pair are slightly due to the unnecessary transformation made on the i-vectors for improvements. It is also to be noted that there is a decrease in performance for the case of neutral emotion, which may be due to the unnecessary transformation made on the i-vectors when they are already obtained from neutral speech. Further, on comparing the average accuracy, we find both EINV-Test and EINV-Pair outperform the contrast i-vector averaging of multiple emotion based system, showing the effectiveness of proposed emotion invariant speaker embeddings. An average accuracy of 90.5% is obtained with EINV-Pair framework, which has an absolute improvement of 2.6% over the average i-vector based approach with different emotions. We also note the improvements with EINV-Pair are slightly higher than that of EINV-Test as both train and test i-vectors are transformed to emotion invariant space, thereby further reduces the emotion mismatch.

We now focus on the performance comparison among the contrast system with average i-vectors of emotional speech and the proposed frameworks with emotion invariant speaker embeddings in Table IV. It is observed that the performance of speaker identification for happiness, anger and sadness emotions consistently improves with both EINV-Test and EINV-Pair frameworks. Further, we find that speaker identification with happiness emotion is benefited maximum, whereas the sadness and anger emotions are more or less equally impacted for improvements. It is also to be noted that there is a decrease in performance for the case of neutral emotion, which may be due to the unnecessary transformation made on the i-vectors when they are already obtained from neutral speech. Further, on comparing the average accuracy, we find both EINV-Test and EINV-Pair outperform the contrast i-vector averaging of multiple emotion based system, showing the effectiveness of proposed emotion invariant speaker embeddings. An average accuracy of 90.5% is obtained with EINV-Pair framework, which has an absolute improvement of 2.6% over the average i-vector based approach with different emotions. We also note the improvements with EINV-Pair are slightly higher than that of EINV-Test as both train and test i-vectors are transformed to emotion invariant space, thereby further reduces the emotion mismatch.

We now plot the t-SNE visualizations to observe the effect of proposed emotion invariant speaker embeddings on i-vectors from different speakers with emotional speech [41]. Figure 3 shows the t-SNE plots for randomly chosen 5 speakers from the database for this analysis. The five different colors represent i-vectors with four different emotions from those different 5 speakers. We find that the emotion invariant speaker embeddings are more distinguishable as speaker clusters and separable compared to the original i-vector based speaker embeddings.
representation. This further strengthens the effectiveness of the proposed emotion invariant speaker embedding for identifying speakers with emotional speech.

In this work, we have studied speaker identification with emotional speech using a relatively smaller database due to unavailability of a larger corpus for such studies. The future work will focus on extending the work on a relatively larger database with more number of emotion classes as well as speakers to demonstrate its significance for real-world scenario.

V. CONCLUSIONS

This work attempts to improve speaker identification with emotional speech from the view of practical systems. Four different emotions, namely, neutral, anger, happiness and sadness from IEMOCAP database are considered for the study. A baseline and another contrast system are built using training data from the four emotions and their average models with different emotions using i-vector modeling. A novel method is then proposed to transform the i-vectors containing speaker-specific information into an emotion invariant space in terms of emotion invariant speaker embedding. This proposed representation gives an absolute improvement of 2.6% in accuracy over the average speaker model with different emotions based system. In addition, we observe significant improvements for happiness, anger and sadness emotion classes, which is maximum for the happiness emotion.

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