

Real-time speech emotion analysis for smart home assistants

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Real-Time Speech Emotion Analysis for Smart Home Assistants

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Abstract

Artificial Intelligence (AI) based Speech Emotion Recognition (SER) has been widely used in the consumer field for control of smart home personal assistants, with many such devices on the market. However, with the increase in computational power, connectivity and the need to enable people to live in the home for longer through the use of technology, then smart home assistants that could detect human emotion will improve the communication between a user and the assistant enabling the assistant to offer more productive feedback. Thus, the aim of this work is to analyze emotional states in speech and propose a suitable method considering performance versus complexity for deployment in Consumer Electronics home products, and to present a practical live demonstration of the research. In this paper, a comprehensive approach has been introduced for the human speech-based emotion analysis. The 1-D convolutional neural network (CNN) has been implemented to learn and classify the emotions associated with human speech. The paper has been implemented on the standard datasets (emotion classification) Ryerson Audio-Visual Database of Emotional Speech and Song (RAVDESS) and Toronto Emotional Speech Set database (TESS) (Young and Old). The proposed approach gives 90.48%, 95.79% and 94.47% classification accuracies in the aforementioned datasets. We conclude that the 1-D CNN classification models used in speaker-independent experiments are highly effective in the automatic prediction of emotion and are ideal for deployment in smart home assistants to detect emotion.

Index Terms

Artificial intelligence, Convolutional neural network, Emotion recognition system, Smart home assistants.

I. INTRODUCTION

Speech Emotion Recognition (SER) was first proposed in 1997 by Picard [1] and has attracted widespread attention. It is well known that language communication is the preferred method when communicating with others in daily life, and human language is first formed through speech. It can be said that speech plays a decisive supporting role in language. Human speech not only contains important semantic information, but also implies rich emotional information [2]. The aim of SER is to obtain the emotional states of a user derived from their speech [3], thereby achieving harmonious communication between humans or between humans and machines, and in this paper we refer to a machine as a smart home assistant.

Emotion is a comprehensive state that occurs when an individual receives an internal or external stimulus, including physiological reaction, subjective experience, and external behavior [4]. When the internal or external stimulus are consistent with a user's needs or requests, they will experience positive emotions, whereas negative emotions can be experienced with unpleasant experiences or distress.

The ability for consumer devices to detect emotion has been a hot research topic since 2006 with the introduction of an early music recommender system [5], and facial expression recognition [6] for personal cameras in 2010. The first emotion recognition systems in the consumer field appeared in 2011 that used a database [7], and then biofeedback [8]. While research into music recommender systems has been buoyant [9], [10], other interesting systems for human emotion include lighting for emotion [11] and emotion aware service robots [12], [13]. Recent research indicates seamless human-device interaction [14]. With the advent of smart consumer home devices [15], consumers can live in their home for longer, safer [16] and to live healthier lifestyles [17].

According to Venkataramanan and Rajamohan [18], the common feature extractions techniques that have been used in speech analysis include the Log-Mel spectrogram, Human Factor Cepstral Coefficients, Mel scale cepstral analysis, Mel Frequency Cepstral Coefficients and Short Term Fourier Transform. This paper is concerned with defining suitable technology and computationally efficient algorithms to enable smart home assistants to predict the emotional state of a user, and in doing so offer more suitable responses to questions and home events.

In this work, an Independent Component Analysis (ICA) algorithm [19] [20] has been used to identify and extract the main audio components. The Mel Frequency Cepstral Coefficients (MFCCs) [21-23] are then used as major prediction coefficients [24] with other feature extraction techniques [25], [26]. This work has culminated in a practical live system to demonstrate the applicability of the research to enable smart home assistants to predict the user's emotional state. For this, two well-known speeches have been used (discussed in section V). Besides this, the model has been examined to detect various emotions based on the age and the gender as well as real time experiments with six different subjects. The work has considered many classifiers and has found that the 1-D CNN is an excellent candidate for SER when considering both recognition performance, and real-time operation.

The paper has been organized as follows. Section II demonstrates the various approaches taken in the literature concerning speech emotion detection. In Section III, the related background concepts have been discussed. The description about the used datasets, the experimental details and the obtained results are discussed in Section V. The obtained empirical observations are also compared with other state-of-the-art techniques in the same section to examine the robustness of our proposed model. Finally, we conclude in Section VI.

II. RELATED WORKS

The detection of psychological states of mind is of great interest and has two major approaches, (a) categorical, and (b) dimensional. The categorical approach defines the psychological state of mind as discrete classes of basic emotion, such as happy, sad, disgust, fear, anger, surprise, and neutral [27], [28]. The dimensional approach deals with emotion as a combination of several psychological dimensions. In 1897, it was known that Wundt [29] proposed that emotion was a combination of three psychological dimensions, (a) strain versus relaxation, (b) arousing versus subduing, and (c) pleasurable versus unpleasant. Russel [30], and Kuppens *et al.* [31] stated that emotion as a combination of only two dimensions, valence, and arousal.

Facial expression analysis and speech analysis [32], [33] are the major approaches to detect the psychological states of the human mind. However, not all humans are able to convey the information related to their psychological states, using facial expression or speech features. Several physiological signals including heart rate, electroencephalogram, blood pressure, skin conductance, skin temperature, and blood volume pulse are also used in the detection of the human emotion [4]. Nowadays, brainwaves are used in the detection of different psychological states [34], [35]. But, that has limited practical use due to a lack of flexibility in the data (signal) recording. The hierarchical model used by Shaqraet *et al.* [36] stated that age and gender account for the acoustic and psychological features of the human voice. Conventional vocal feature extraction methodology, including calculation of Mel Frequency Cepstral Coefficients, Log-Mel spectrogram have been discussed [18, 21, 25, 37]. The Short Term Fourier Transform has been implemented to extract suitable features by Venkataramanan, and Rajamohan [18]. Finally, Long Short Term Memory Cells (LSTM), Hidden Markov Model (HMM) and Deep Neural Networks (DNN) have been employed for

the detection of emotion from human speech. However, the said approaches are not robust and do not detect speech emotions efficiently. Relevant works along with their feature extraction techniques and the classifiers used are given in Table I.

TABLE I
RELEVANT RESEARCH WORKS FOR THE SPEECH EMOTION DETECTION: A COMPARISON

Works	Feature extraction techniques	Used classifiers
[38]	Spectral features + Source features extraction + Sequential Minimal Optimization	Random Forest
[39]	Mel Frequency Cepstrum Coefficients + PCA	K-Means Clustering
[40]	Pitch features	Binary and GMM classifiers
[36]	Age and gender based features	Hierarchical Classification
[37]	Mel-Frequency Cepstral Coefficients	CNN
[18]	Log-Mel Spectrogram, Mel-Frequency Cepstral Coefficients (MFCCs), Pitch and Energy Features	CNN, LSTM and Hidden Markov Models
[41]	Prosodic and spectral features	SVM classifier

III. CONCEPTUAL BACKGROUND

This section describes some related backgrounds to introduce the proposed technique.

A. Independent Component Analysis

Independent Component Analysis (ICA) [20] helps to distillate different components from an audio. The procedure contains following steps:

1. Each pulse of the signal has been depreciated by its mean value for the centering of a complete signal X . The correlation between different components is removed. Then, it is followed by the calculation of their Eigen values and corresponding Eigenvectors. Whitening of a signal refers to eigen value decomposition of the covariance matrix of it. Matrix E is an orthogonal matrix of the eigenvectors and matrix D is a diagonal matrix of eigen values. E^T is the transpose of matrix E . The

whitened signal Y can be calculated as $Y = E \cdot D^{-\frac{1}{2}} \cdot E^T \cdot X$.

2. All the signals are then separated through several iterations (as per the Equ.(1)). After normalization, the algorithm checks whether it has crossed the tolerance level, or not, by using Equ. 2.

$$Y_p = \frac{1}{n} \left[\sum_i^n X \cdot \tanh(Y^T \cdot X) - \sum_i^n (1 - \tanh^2(Y^T \cdot X) \cdot Y) \right] \quad (1)$$

where n is the number of iterations, Y_p is used for denoting new calculated value for de-mixed matrix for a single iteration as:

$$Y_p - \sum_{j=1}^{p-1} (Y_p^T \cdot Y_j) Y_j = \frac{Y_p}{\|Y_p\|},$$

Where $Y_p^T \cdot Y_p + 1 \approx 1$ (2)

The source signal, matrix S , is calculated as the dot product of matrix Y and input signal X , i.e. $S = Y \cdot X$, where Y is the final value of Y_p . The first row of the end signal is collected as the major component.

B. Mel-Frequency Cepstral Coefficients Extraction

First, the complete signal is framed into shorter ones and their Fourier Transform is taken as mentioned in the section III-B1. The power spectrogram is calculated. Using the Mel Filter Bank (Fig. 1.), the sum of energy of each filter is collected and fit to a logarithm. Finally, the Discrete Cosine Transform (DCT) (section III-B3) is applied on top of log Filter Bank energies.

1) **Fast Fourier Transform** (FFT): Mel-Frequency Cepstrums have widely used features for speech processing. Sounds generated by a human solemnly depend upon the structure of their vocal tract. The envelope of the time-power spectrum of any speech signal represents the structure of a vocal track, and MFCCs are used to represent the envelope of a short time-power spectrum. To estimate the power spectral density, the FFT is used as shown in Fig. 2. The discrete signal in frequency domain

$X(k)$ is calculated as $X(k) = \sum_{n=0}^{N-1} x(n)W_N^{nk}$, where $x(n)$ signifies the continuous time domain signal and twiddle factor

$$W_N^{nk} = e^{-\frac{j \cdot 2\pi \cdot k \cdot n}{N}}$$

The Radix-2-Decimation in Time Fast Fourier Transform (DIT-FFT) is used. In Equ.3, n is replaced with $2m$ to differentiate between odd and even portions of a signal.

2) **Filter Bank**: After poly-phase decomposition of the signal stream, inputs of each phase are transmitted to separate channels. The number of filters in a Mel-FB has a set as per the requirements. The filters are triangular and are placed according to the convention of the Mel-scale $M(f)$. The Mel-scale is defined as:

$$M(f) = 2595 \log_{10} \left(1 + \frac{f}{700} \right) \quad (3)$$

where f is used to denote frequency. Normally, triangular filters are distributed from zero to the Nyquist rate ($\geq 2 \times$ Sampling Frequency) (see Fig. 1). As the input of Discrete Cosine Transform, collected absolute amplitudes of log filter bank are used.

3) **Discrete Cosine Transform (DCT)**: The DCT is used to present a signal in terms of the sum of cosine functions oscillating in at different frequencies. There are eight types of DCTs from which, MFCCs are extracted using DCT-II [42] and DCT-III [43] as:

$$X_k = \sum_{n=0}^{N-1} x(n) \cos \left(\frac{\pi \cdot (n + \frac{1}{2}) \cdot k}{N} \right) \quad (3)$$

where $x(n)$ is signal stream and N is the total number of elements. The complete process of extraction of MFCCs is shown in Fig. 3.

4) **Removal of Useless Features**: After clearing the silence from each vocalization, the length of each recording changed accordingly, this could not be fit in a model. Thus, to make the length same for all MFCCs arrays, a specific number of features are kept, and others are discarded. The size of vectors is 120 for TESS and RAVDESS each.

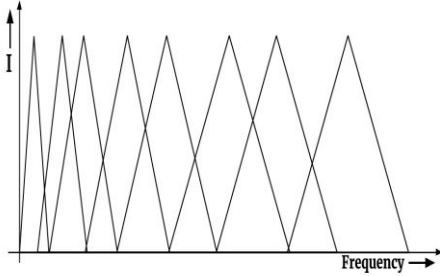


Fig. 1. Mel Filter Bank (Mel-FB).

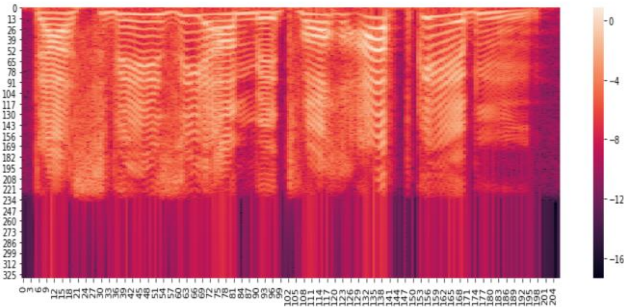


Fig. 2. Calculated Spectrum of an Angry Speech Signal using the FFT.

IV. PROPOSED APPROACH

The approach that involves the channelization of the raw speech (vocal) samples through Silence Filter followed by ICA [19], [20] and mean MFCCs extractor to extract a vector of cepstral coefficients, which are then passed into the 1-D convolution neural network (CNN) to extract the high-level features followed by a fully connected dense layers and a softmax classifier. The proposed emotion detection approach consists of four stages, as shown in Fig. 4.

First is the collection of the voice samples. In the second step, the silences in the samples are cleared. The main component of the sample is extracted, leaving behind the noise. In the next stage, the vectors are processed to calculate the values of the coefficients, and convolutional operation is applied to the output vector. In the final phase, the vectors are passed through a dense neural network followed by an output layer.

The complete system works as shown in Fig. 5. Once the human vocalization is collected, the silent portions are removed from the input raw signal. After that, using ICA, the main speech component has been extracted and fed to a specific classification model which gives a predicted emotion class.

A. Proposed Model Architecture

The first model used here a Convolutional Neural Network. We keep the ADAM optimizer for all the experiments due to a comparatively better rate of convergence. Relu activation function has been used due to overcoming the unwanted loss caused by negative value coverage. The shuffled signals are independent on future inputs for which causal padding is preferred over valid padding. A 1D-CNN has been implemented on 7 classes (happy, sad, disgust, fear, anger, surprise, and neutral). Each layer consists of a window of size 5 (kernel size). After a few convolutional layers, a max pool has been introduced with a window of size 8 (pool size) to reduce the feature set. Later on, the model is tuned by adding some extra convolutional and dropout layers. After flattening, very few Dense layers and outputs are collected at the end (see Fig. 6).

V. EXPERIMENTAL SETUP AND RESULT ANALYSIS

This section discusses the implementation environment and based on the experiments the results are analyzed. Two standard datasets have been used for the validation of our proposed model namely, Ryerson Audio-Visual Database of Emotional Speech and Song (RAVDESS)¹ [44] and Toronto Emotional Speech Set database (TESS)².

The “RAVDESS” dataset is rich, robust, and gender-balanced. It consists of 24 professional actors vocalizing two statements in a neutral North American accent. There are two statements, and each statement is repeated with two different intensities, normal and strong for each emotion except “Neutral” one. The two enacted statements are, “Dogs are sitting by the door” and “Kids are talking by the door”, and each statement is repeated two times by each actor. The “TESS” dataset consists of 200 words spoken as “Say the word a word” by two females of age 26 and 64, respectively. Each sentence is repeated seven times with different emotions portrayed. Both actresses having musical training, this dataset is very well balanced and have threshold value within the normal range.

A. Preprocessing of Raw Data

A voice contains several void spaces which can lead an algorithm to faulty learning at the time of training. Testing with the same data may give an accurate result, but ultimately at the time of real-life implementation of the model, it may overlook critical phases. Spectral subtraction is a widely used technique to remove the background noise. For the first step, the lowest threshold value has been set, intensities below, which are not accepted. A sumptuous data array could be achieved for each vocal utterance. Each recording is of 3 seconds approximately. Duration became near about 2 seconds after cleaning silences, as shown in Fig. 7. The mean of evenly weighted rolling window values with an offset of *sampling_rate*=10 are taken for each vocalization with a minimum period of 2 seconds.

B. Experiments and Result Discussion

Apart from the datasets, the proposed approach is further validated using audio samples from movie clips, orientation functions, and protests. The sample vocals are larger than the normal size of input vocals used for training. The datasets are divided into 10 parts. The first 8 parts are used to train, whereas the remaining 2 parts are used to validate our model. The validation accuracy is considered to be the mean of the validation accuracies around a radius of the 5 epochs (± 5 of the best performing epoch), from the epoch that produces the maximum validation accuracy.

¹<https://zenodo.org/record/1188976>

²<https://doi.org/10.5683/SP2/E8H2MF>

1) Experiment 1a: The Standard Experiment:

The RAVDESS and TESS standard datasets have been used and compared with the latest best performing speech emotion detection models.

• **Result Analysis on RAVDESS dataset:** In Fig. 8, it can be seen that the maximum accuracy of 92.20% has been obtained for the epoch 580. Taking the mean and the standard deviation, of the validation accuracies from epoch 575 to epoch 585 (± 5 epochs), the resultant validation accuracy obtained by our model for the RAVDESS dataset is $90.48 \pm 1.28\%$.

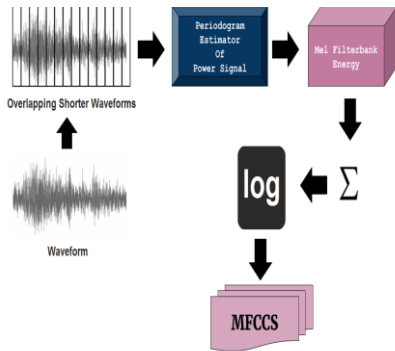


Fig. 3. Mel-Frequency cepstral coefficients extraction process.



Fig. 4. Block diagram of the proposed system.

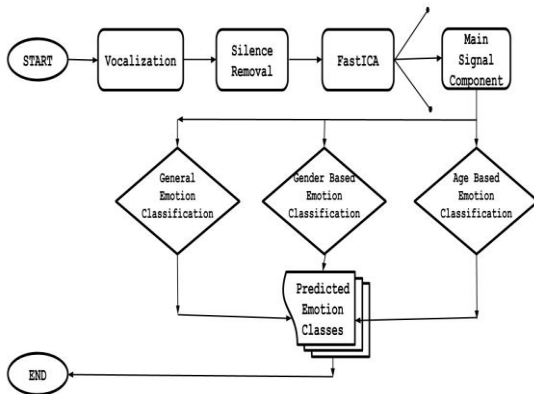


Fig. 5. Flow chart of our proposed system.

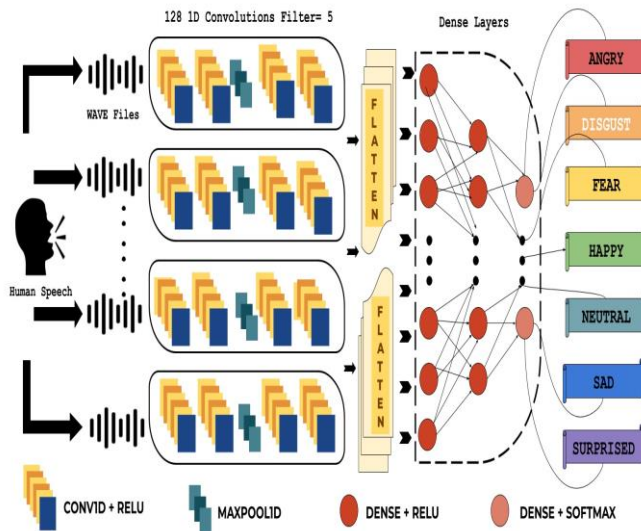


Fig. 6. Model architecture.

- **Result Analysis on TESS dataset:** The TESS dataset is divided into two parts, young and old. For the TESS young dataset, about the Fig. 9, the highest accuracy 96.5% has been found for the 569th epoch. Taking the mean and the standard deviation, of the validation accuracies from epoch 564 to epoch 574 (± 5 epochs), the resultant validation accuracy obtained by our model for this dataset is $95.79 \pm 0.38\%$.

For the TESS old dataset, the highest accuracy 96.62% has been found for the 690th epoch. Taking the mean and the standard deviation, of the validation accuracies from epoch 685 to epoch 695 (± 5 epochs), the resultant validation accuracy obtained by our model for this dataset is $94.47 \pm 1.24\%$ as shown in Fig. 10.

Additionally, the preprocessed data is also passed through an SVM classifier to compare the rest. The accuracy for the RAVDESS, TESS Young and TESS Old datasets are 44.73%, 91.66%, and 90.62% respectively.

- **Comparison of the classification accuracy:** The proposed approach outperforms most of the approaches taken by different researchers all over the globe. Comparative analysis of the validation accuracy for the RAVDESS, TESS Young, and TESS Old datasets, respectively, has been given in Table II.

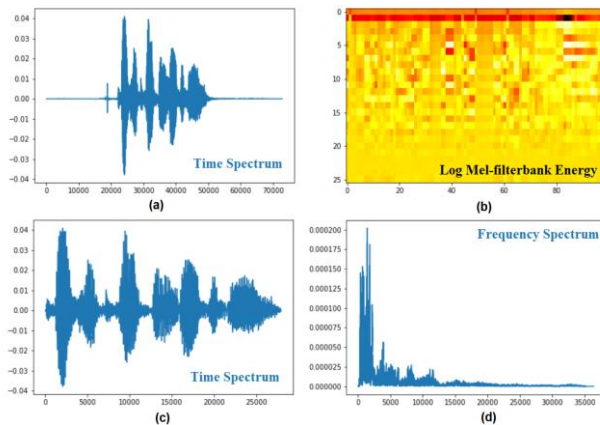


Fig. 7. Waveforms: (a) time domain signal before filtering voids, (b) Log-Mel Spectrogram of signal after filtering voids, (c) time domain signal after filtering voids, and (d) frequency domain signal after filtering voids. Model architecture.

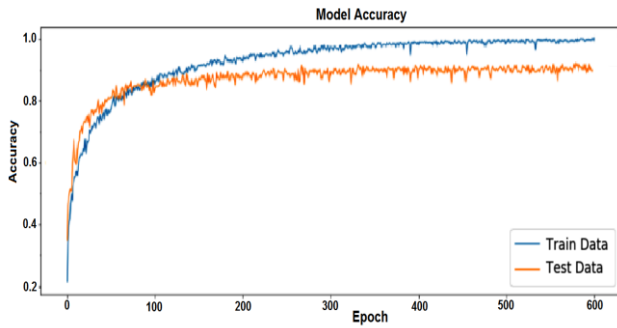


Fig.8. Accuracy graph obtained from RAVDESS dataset.

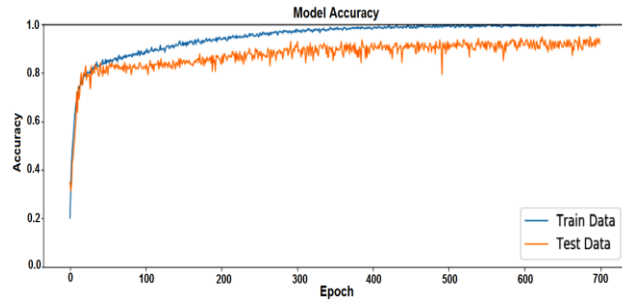


Fig. 9. Accuracy graph obtained from the TESS YoungDataset.

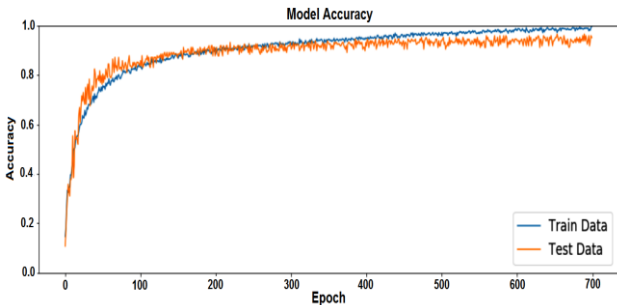


Fig. 10. Accuracy graph obtained from the TESS Old dataset.

2) Experiment 1b: Gender and Age-based Speech Emotion Classification:

The models are used to classify age and gender, respectively, to increase the robustness of the system. In a similar manner, a CNN based model has been prepared. For age classification, the TESS dataset had been used. The model is trained with 100 epochs and having only two classes, that is, young and old, binary cross-entropy is preferred for loss function. Among all activation functions, Sigmoid came out with a comparatively better option for the output layer. For gender classification, the RAVDESS dataset had been used. The overall model has been kept the same for gender classification. For extraction of MFCCs, DCT Type-III is preferred over DCT Type-II.

- **Analysis on Age classification:** The TESS dataset is further used to classify the age (young or old). The learning pattern is shown in Fig. 11. The obtained highest classification accuracy equals to 97.50% at the epoch 87. Taking the mean and the standard deviation, of the validation accuracies from epoch 77 to epoch 97 (± 5 epochs), the resultant validation accuracy obtained by our model for the TESS dataset is $95.79 \pm 1.15\%$. The comparison in accuracy using CNN and SVM has been shown in Table III.

- **Analysis on Gender classification:** The RAVDESS dataset is further used to classify the gender of the samples.

- The learning trend is given in Fig. 12. The obtained highest classification accuracy equals to 100.00% at epoch 45. Taking the mean and the standard deviation, of the validation accuracies from epoch 35 to epoch 55 (± 5 epochs), the resultant validation accuracy obtained by our model for the RAVDESS dataset is $100.00 \pm 0\%$. The comparison in accuracy using CNN and SVM has been shown in Table IV.

3) Experiment 2: Emotion Detection from two Famous Speeches:

As a part of experimental analysis, the model has been validated using the vocals of the famous speeches of the great Indian sage Swami Vivekananda delivered at Chicago³ [45] [46] in 11th September 1893 and an inspiring speech during the promotion of #MeToo movement⁴ [47] in 8th January 2018 at the Golden Globe Awards. The detailed speech emotions have been shown in Fig. 13 and Fig. 14 respectively for the Experiment 2. Both the audio-visual clippings are longer than 9 minutes. Among which a random 2 minute and 45 seconds, that is, 165 seconds are taken for the experimental purpose. For each experiment, the signal sampling rate is taken as 44100 and FFT window is considered of 2048 length and hop length is taken as 512. So, the number of MFCCs can be obtained using $n\text{-Coeffs} = (len \times sr) / hop$, where, n-Coeffs is number of coefficients, len is the audio length, sampling rate is denoted as sr and hop is used for number samples between successive frames. According to which, the calculated n-Coeffs will be near about 87 (approximately) for 1 second (that is, $1 \times 44100 / 512$). Similarly, for 165 seconds, 14355 coefficients will be generated. We are fitting 120 coefficients for each iteration of emotion detection. Therefore, zero padding is done in the MFCCs array to make it completely divisible. In the result, a coefficient array of length 14400 is generated. Finally, the complete array has been divided into 120 coefficient subarrays (i.e., $14400 / 120$).

For each iteration, the model is giving a percentage of each seven emotions present in that vocalization. In the provided charts we can visualize the distributions for each emotion according to their percentage of intensities. The pie chart indicates the proportional analysis of individual emotional states for 165 seconds. The intensity represented in the Y-axis is mapped according to the model confidence representation for each emotion in percentage. The X-axis suggests the number of batches obtained from 165 seconds audio input.

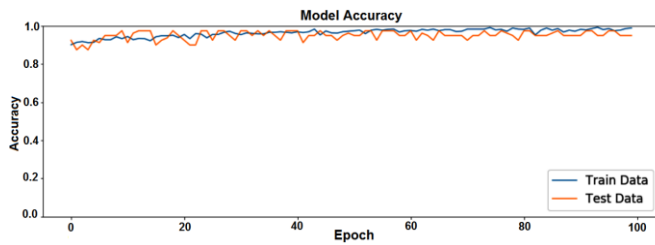


Fig. 11. Classification accuracy plot obtained from the proposed age-based speech emotion detection model (Experiment-1b).

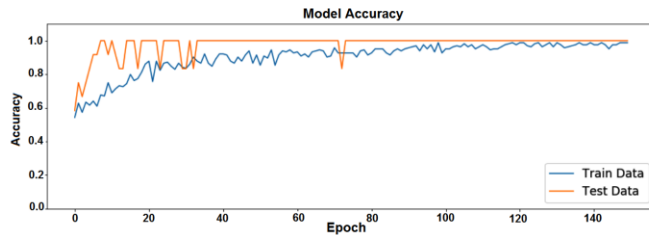


Fig. 12. Classification accuracy plot obtained from the proposed gender-based speech emotion detection model (Experiment-1b).

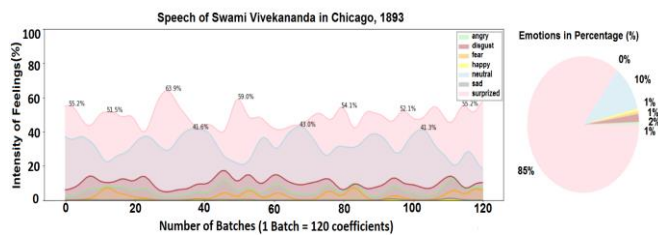


Fig. 13. Results from the speech of Swami Vivekananda at Chicago (1893).

³<https://youtu.be/WIOtnn6meWA>

⁴<https://youtu.be/LyBims8OkSY>

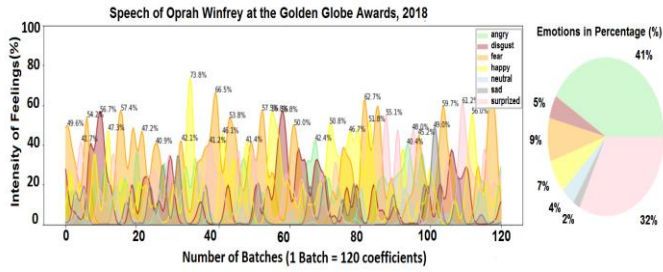


Fig. 14. Results from inspiring speech during the promotion of #MeToo movement.

4) Experiment 3: Real-time Experiment:

The proposed model is tested with live sample speech audio clips to check the effectiveness of our proposed model in the real-world in real-time. The speech audio inputs are taken from 6 subjects (three males and three females), in the age group 19 years to 24 years. The subjects are asked to speak about random topics in English with different emotions⁵ and later the ground emotions are also verified by them. The segment duration of recorded sample audio is below 60 seconds. A sample emotion detection of subjects 2 and 3 has been shown in Figs. 15 and 16 with their respective detection times (in seconds).

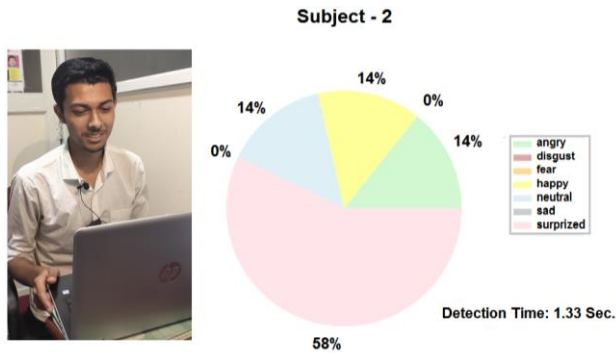


Fig. 15. Real-time speech emotion detection by subject 2 (Experiment-3).

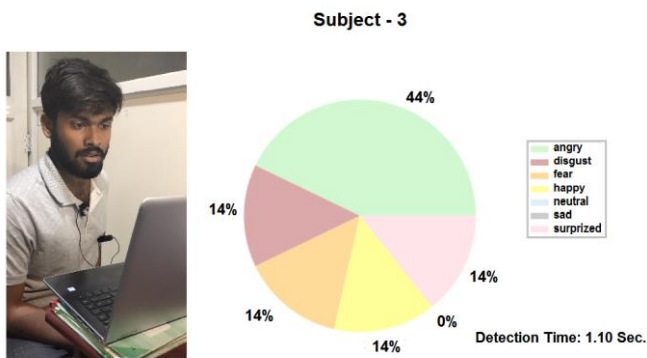


Fig. 16. Real-time speech emotion detection by subject 3 (Experiment-3).

⁵<https://soundcloud.com/user-766220714/sets/speech-emotion>

TABLE II
RESULTS OBTAINED FROM DIFFERENT SPEECH EMOTION
DETECTION MODELS (EXPERIMENT-1A)

Method	RAVDES	TESS Young	TESS Old
SVM	44.73%	91.66%	90.62%
1D-CNN (proposed)	90.48 ± 1.28%	95.79 ± 0.38%	94.47 ± 1.24%

TABLE III
RESULTS OBTAINED FROM DIFFERENT AGE-BASED
SPEECH EMOTION DETECTION MODELS (EXPERIMENT-1B)

Method	Accuracy
SVM	95.00%
1D-CNN (proposed)	95.79 ± 1.15%

TABLE IV
RESULTS OBTAINED FROM DIFFERENT GENDER-BASED
SPEECH EMOTION DETECTION MODELS (EXPERIMENT-1B)

Method	Accuracy
SVM	97.6%
1D-CNN (proposed)	100.00 ± 0.00%

The detailed results of the Experiment 3 (done by the authors to detect aggression and anger from human speech in real-time) is shown in Table V.

TABLE V
RESULTS OBTAINED FROM DIFFERENT SUBJECTS EXPERIMENTED BY THE AUTHORS (EXPERIMENT-3)

Test subject	Gender	Duration	Top two predicted emotions	Ground truth	Detection time
Subject 1	male	15s	Surprised, Neutral	Surprised	1.28s
Subject 2	male	15s	Surprised, Happy	Surprised	1.33s
Subject 3	male	15s	Angry, Fear	Angry	1.10s
Subject 4	female	36s	Happy, Neutral	Happy	1.21s
Subject 5	female	53s	Surprised, Happy	Happy	1.39s
Subject 6	female	36s	Surprised, Fear	Surprised	1.35s

TABLE VI
COMPARISON OF THE BEST SPEECH EMOTION DETECTION MODELS WITH THE PROPOSED APPROACH

Works	Feature extraction technique	Used classifiers	Emotion accuracy (TESS combined)	Emotion accuracy (RAVDESS)	Age accuracy (TESS)	Gender accuracy (RAVDESS)
[18]	Log Mel Spectrogram	2D CNN with global avg. pool	62%	70%	-	90%
[36]	OpenSmlr Tool & eGeMAPS	SVM	65.1%	70.59%	87.95%	98.67%
[38]	Linear Prediction Residuals	SVM with SMO	75.5%	-	-	-
[48]	eGeMAPS	CNN-LSTM & CNN	49.48%	65.67%	-	-
Proposed Approach	Mel FB Cepstral Coeffs with DCT-III	1D CNN	95.79%	90.48%	95.79%	100%

C. Results Comparison

The proposed model has been compared with the best performing existing detection models in Table VI. The probable reason for achieving the best accuracy is using DCT Type-III instead of DCT Type-II and using causal padding at the time of training a model. Features like MFCCs have always helped to acquire better results previously. Moreover, this work has prepared different classification models for different age groups and introduced 1D CNN classifier which can lead to a better result.

VI. CONCLUSION

In this article, the Mel-Frequency Cepstral Coefficient (MFCC) features were applied to conduct speaker-independent experiments for a Speech Emotion Recognition (SER) element of an AI based smart home assistant. The proposed approach i.e., CNN variant achieved a state-of-the-art classification accuracy for the commonly used benchmark speech datasets in the world. The proposed approach provides 90.48%, 95.79% and 94.47% classification accuracies for the standard datasets (emotion classification) Ryerson Audio-Visual Database of Emotional Speech and Song (RAVDESS) and Toronto Emotional Speech Set database (TESS) (Young and Old) respectively. We have also examined its robustness with some well-known recorded speeches and movie scenes. It must be noted that these test cases are unknown to the model as they are not only part of the training sets but also uncorrelated with the given validation sets. It indicates that our proposed model learns in a true sense to detect speech-based emotions. The training time and testing time consumed were relatively short indicating an ability to operate in real-time and as an excellent candidate for enabling AI smart home assistants to predict the emotion of the user with the aim to improve assistance and feedback.

In future, this work will be extended to address the more number of subjects, different nationalities, multiple accents, and various age and gender categories; and will try to develop an application that could be deployed in the real world to enhance real-time identification with consumers. Furthermore, we will attempt to extract multiple types of acoustic features to fuse features of advanced classification models for SER. Other indicators will also be added to evaluate the performance of a classification model synthetically.

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