Stress Detection Through Speech Analysis

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Abstract: The work presented in this paper uses speech analysis to detect candidates stress during HR (human resources) screening interviews. Machine learning is used to detect stress in speech, using the mean energy, the mean intensity and Mel-Frequency Cepstral Coefficients (MFCCs) as classification features. The datasets used to train and test the classification models are the Berlin Emotional Database (EmoDB), the Keio University Japanese Emotional Speech Database (KeioESD) and the Ryerson Audio-Visual Database of Emotional Speech and Song (RAVDESS). The best results were obtained with Neural Networks with accuracy scores for stress detection of 97.98\% (EmoDB), 95.83\% (KeioESD) and 89.16\% (RAVDESS).

1 INTRODUCTION

Automating some of the HR recruitment processes alleviates the tedious HR tasks of screening candidates and hiring new employees. This paper summarizes the results of applying several speech analysis approaches to determine if the interviewed candidates are stressed.

Automatic video analysis is adopted in an HR screening product, to be able to automatically detect stress during an interview process. The screening process works as follows: 1) the candidate receives an invitation for a screening, 2) he/she connects to the Website (usually an interview platform) and 3) is asked a few questions. For each question, the candidate records his/her answer whilst respecting a predefined time limit. Once validated, the video is sent to the recruiter and can be visualized later on.

In order to help the recruiter in the process of making his decision, the candidate’s stress is assessed using speech analysis techniques.

2 RELATED WORK

Several existing studies target emotion recognition through speech analysis. Speech emotion analysis focuses on the non-verbal aspect of the speech. Studies have shown that emotions show different patterns and characteristics through vocal expressions (Banse and Scherer, 1996). Because of the high number of ethnicities and people in the world, the speech analysis field remain fairly complicated. Moreover, the complex nature of voice production and the different families between emotions (like cold anger and hot anger for example) increase even more the complexity.

The emotions expressed through the voice can be analyzed in three different levels:

- Physiological: Describes nerves impulse or muscle innervation patterns involved in the voice production process.
- Phonatory-articulatory: Describes the position and movements of the major structures like the vocal folds (also called more commonly vocal cords).
- Acoustic level: Characteristics of the audio signal produced by the voice.

The study summarized in this paper focus on analyzing the acoustic level as it is a non-intrusive method and it is widely used across various studies (e.g. (Lanjewar et al., 2015; Seepaopoch and Wongthanavasu, 2013)).

Currently, stress cannot be precisely defined (Johnstone, 2017). However, it is subject to a lot of recent studies because of its importance for people in everyday life. The results are hard to interpret because reaction against stress can be different among people, everyone having a certain behavior towards it. Plus, stress can have different forms, like cognitive or emotional.
In (Johnstone, 2017), stress is defined as a people state in different situations that may cause anxiety or mental challenge. Since it looks like stress has characteristics really close to anxiety, particular attention has been given to anxiety characteristics.

3 METHODOLOGY

3.1 Feature Selection

In order to identify different emotions in a human speech, features like pitch (also referred as fundamental frequency), articulation rate, energy or Mel-Frequency Cepstral Coefficients (MFCCs) are used.

In (Banse and Scherer, 1996), a complete table listing speech characteristics according to 6 emotions is presented. By analyzing this table, it appears that mean energy and mean intensity could be enough to successfully distinguish the 5 following emotions: happiness, disgust, sadness, fear/anxiety (stress) and anger. The MFCCs do not appear in this table but will also be chosen for speech analysis because of their excellent results in such problems. Mean energy, mean intensity and the MFCCs are therefore chosen as features for emotions classification.

3.2 Feature Extraction

3.2.1 Mean Energy

The vocal energy is defined by the following formula (Boersma and Weenink, 2006):

\[ \int_{t_1}^{t_2} x^2(t) \, dt \]  

where \( t_1 \) and \( t_2 \) are the beginning and the end of the audio signal and \( x(t) \) the signal function. If the unit of the amplitude is in [Pa], the obtained energy is then in [Pa^2s].

Since such mathematical formula is not easy to implement in a programmatic way (mostly because \( x(t) \) is unknown), the results obtained with a Python library have been compared with the results returned by the Praat software (which use the above formula).

To compare the results, different audio samples have been chosen and for each one, the mean energy computed using Praat then Python.

It turned out that the values are not exactly the same but they follow the same tendency for every audio sample.

3.2.2 Mean Intensity

The vocal intensity is the amplitude of the signal, in [dB]. In order to have a mean intensity, the following formula is applied to the signal (Boersma and Weenink, 2006):

\[ 10 \log \left[ \frac{1}{(t_2 - t_1)} \int_{t_1}^{t_2} 10 x(t) \, dt \right] \]  

with \( t_1 \) being the beginning of the frame and \( t_2 \) the end of it. As for the mean energy, comparison have been done between Praat and a Python library allowing to extract the mean intensity.

3.2.3 Mel Frequency Cepstral Coefficients

The speech produced by a human being involves a sequence of complicated articulator movements but also the airflow from the respiratory system. The vocal cords, the tongue, the teeth, these are all elements that filter the sound and make it unique for every speaker.

The sound is therefore determined by the shape of all these elements. This is where MFCCs come in to play, this shape manifests in the envelope of the short time power spectrum, and the MFCCs represent this envelope (Lyons, 2015). Several steps are done in order to extract these coefficients from an audio file: the framing, the windowing, FFT (Fast Fourier Transform), Filter Banks and the MFCCs step.

The output of such computations is a NxM matrix with N being the number of frames obtained after the framing step (a frame is usually 25-40ms long) and M the number of coefficients, which is 13. It is possible to extract 26 (deltas) or 39 (deltas-deltas) coefficients according to the needs. The deltas give the trajectories of the coefficients and thus give information about the dynamics of the speech.

The deltas-deltas are computed from the deltas and inform on the acceleration. In this work, only the first 13 coefficients are used because the deltas and deltas-deltas represents only small details and the 13 first coefficient contribute to most of the Automatic Speech Recognition.

All these features are obtained with a couple of Python programming language libraries which use a somewhat different approach that the mathematical formulas cited above. However, as seen in the previous chapters, the results are the same and it does not affect the classification performance.

3.3 Datasets

Three different datasets were used, the Berlin Emotional Database (EmoDB) (Burkhardt et al., 2005), the Keio University Japanese Emotional Speech...
Database (KeioESD) (kei, Mori et al., 2006; Moriyama et al., 2009) and the Ryerson Audio-Visual Database of Emotional Speech and Song (RAVDESS) (Livingstone et al., 2012). The first one, EmoDB, contains about 500 audio files where 10 different actors spoke in German in 7 different emotions. The length of an audio file varies from about 2 to 4 seconds. The second one, in Japanese, is a set of words pronounced by a male speaker. There is a total of 19 words spoken in 47 different emotions. Since it is only words and not sentences, the average length of a file is 0.5 second. Finally, the third dataset contains about 1000 samples with different sentences spoken by different speakers (24), male and female. As for the EmoDB dataset, the sentences last for about 3 seconds.

4 EXPERIMENTS

Four kinds of feature sets have been built from the data obtained in the datasets. There are several ways to handle the MFCCs and these feature sets will allow to determine which one is the best.

For the MFCCs, the framing window size is set to 25ms and the number of coefficients returned is 13 (deltas and deltas-deltas are omitted). Note that the frames overlap themselves and that the step length between 2 frames is 10ms.

The Table 1 summarizes the created feature sets and the features they contain.

<table>
<thead>
<tr>
<th>No</th>
<th>Features</th>
<th>Nb. of features</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>Mean energy</td>
<td>2</td>
</tr>
<tr>
<td></td>
<td>Mean intensity</td>
<td></td>
</tr>
<tr>
<td>2</td>
<td>Mean energy</td>
<td>2 x (n x 13)</td>
</tr>
<tr>
<td></td>
<td>Mean intensity</td>
<td></td>
</tr>
<tr>
<td></td>
<td>Every MFC coefficient</td>
<td></td>
</tr>
<tr>
<td>3</td>
<td>Every MFC coefficient</td>
<td>n x 13</td>
</tr>
<tr>
<td>4</td>
<td>Mean energy</td>
<td>2 + (13 x 2)</td>
</tr>
<tr>
<td></td>
<td>Mean intensity</td>
<td></td>
</tr>
<tr>
<td></td>
<td>Mean of MFCCs</td>
<td></td>
</tr>
<tr>
<td></td>
<td>Std. of MFCCs</td>
<td></td>
</tr>
</tbody>
</table>

When every MFC coefficient is mentioned, this means that each value in the MFCCs matrix is used as a feature. This concept leads to big feature sets with more than one thousand features. In the number of features column, \( n \) represents the number of line of the matrix (therefore the number of frames obtained after the framing step).

Even if stress is the targeted emotion for this research, tests have also been done on features sets containing five emotions (labels) as well as on feature sets having only two emotions (Anxiety/Stress and No Stress). The five chosen emotions are Happiness, Disgust, Sadness, Anxiety/Stress and Anger. Stress is an emotion that cannot be clearly defined but is however rather close to anxiety, especially when it comes to interviews.

Various algorithms are used for supervised learning in speech analysis problems. Among them, Artificial Neural Networks (ANNs), Hidden Markov Model (HMM), Gaussian Mixture Model (GMM), Support Vector Machines (SVMs) and even Nearest Neighbors (kNN) are the ones frequently mentioned. Each of them has proved significant results when it comes to classification problems. In this paper, SVMs and ANNs have been chosen. The choice is motivated by the fact that both algorithms are implemented in the library used for the machine learning experiments and also because a comparison can be done with another study dealing with SVMs and the same datasets (Seehapoch and Wongthanavasup, 2013).

The datasets have been divided with a 1/4 ratio, 25% of the data has been used as test set and 75% as training set. Grid search was applied to exhaustively search the best hyper-parameters, as well as the best feature set construction, for both ANNs and SVMs. The accuracy scores obtained after this fine-tuning step are presented in the following section.

5 RESULTS

The results obtained highlight the fact that the MFCCs are obviously very good speech analysis features and that mean energy and mean intensity are not enough to successfully classify emotions. Also, the best way to use the MFCCs is when their mean and standard deviation are computed. This construction has shown the best results for almost every classification with either ANNs or SVMs. Both algorithms have shown excellent results, with ANNs having slightly better scores than SVMs.

The table 2 displays the accuracy scores obtained on the KeioESD dataset for a multiclass classification using SVMs and ANNs respectively.

The table 2 shows that the third feature set format is the best with SVMs. However, the fourth format would be a privileged choice for ANNs. The overall best accuracy score obtained for both algorithms is the same but using a different feature set format.

In addition to accuracy scores, confusion matrices have also been computed to see if there are emotions that are easier to classify than others. Fig. 1 shows the
Table 2: Accuracy scores obtained for multiclass classification on KeioESD dataset with SVM and ANN.

<table>
<thead>
<tr>
<th>Classification algorithm</th>
<th>Without MFCCs</th>
<th>With MFCCs</th>
<th>Only with MFCCs</th>
<th>Mean and Std of MFCCs</th>
</tr>
</thead>
<tbody>
<tr>
<td>SVM</td>
<td>20.83%</td>
<td>75%</td>
<td>83.3%</td>
<td>50%</td>
</tr>
<tr>
<td>ANN</td>
<td>20.83%</td>
<td>62.5%</td>
<td>70.83%</td>
<td>83.33%</td>
</tr>
</tbody>
</table>

confusion matrix obtained for a multiclassification on the KeioESD dataset using ANNs. Happiness (which is the only positive emotion in this paper), is the most difficult to successfully classify whereas anger, sadness and stress have a perfect score of 100.

Results for EmoDB and RAVDESS datasets show that the fourth feature set format performs better with multiclass and two-class classification. Accuracy scores are better with this latter format and confusion matrices show overall higher scores for every emotion.

With the fourth feature set format, the length of the audio file does not matter (since the MFCCs are averaged) which is a really good thing knowing that the number of features must be the same for an unclassified sample and for the feature set the classifier has been trained with.

Figure 1: Confusion matrix for multiclassification on KeioESD dataset using ANNs.

Because of the complexity of such algorithms, a set of parameters have been chosen with a range of value for each parameter.

The Table 3 shows the best accuracy scores obtained for each dataset and the optimal algorithm associated along its parameters value, including two-class feature sets. For ANNs, the parameters are, in order, activation method, solver, alpha value and the maximum number of iterations. For SVMs, parameters are kernel, C coefficient and gamma value.

Both SVMs and ANNs parameters have been optimized with the help of the scikit-learn library method GridSearchCV. This method is in charge of finding the best combination of values where the algorithm gives the best result for a given set of features.

Experiments on emotion detection through speech have been conducted in (Seehapoch and Wongthanavanu, 2013) focusing on SVMs. Since SVMs are also used in this work, it is interesting to compare the results. MFCCs are also used as features in their work, however in a different way. It is stated that there is a number of 105 features, but it is however not clearly explained how this number is obtained from the MFCCs and if this number is the same for each dataset.

Moreover, two of the three datasets they used are similar to some used in the present work, the KeioESD and the EmoDB datasets. The Table 4 compares the accuracy results they obtained with the results obtained in this paper with the use of MFCCs only (third feature set format) on two different datasets.

The results obtained in the present paper are a little bit worse, which is understandable since in (Seehapoch and Wongthanavanu, 2013) the MFCCs seemed to have been used in a better way than just used as single feature (each coefficient becoming a feature). It is hard to evaluate this comparison because the way the MFCCs they used is not clearly explained. The parameters used to get the MFCCs, like the window length or the step length are also not

Table 3: Best accuracy scores obtained KeioESD dataset with SVM and ANN.

<table>
<thead>
<tr>
<th>Feature set</th>
<th>EmoDB</th>
<th>KeioESD</th>
<th>RAVDESS</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>adam 0.00001</td>
<td>adam 0.001</td>
<td>adam 0.00001</td>
</tr>
<tr>
<td></td>
<td>500</td>
<td>500</td>
<td>1000</td>
</tr>
<tr>
<td></td>
<td>87.23%</td>
<td>83.33%</td>
<td>78.75%</td>
</tr>
<tr>
<td>Two-class</td>
<td>[ANNS] relu</td>
<td>[SVMs] linear</td>
<td>[ANNS] relu</td>
</tr>
<tr>
<td></td>
<td>adam 0.00001</td>
<td>1 0.01</td>
<td>adam 0.001</td>
</tr>
<tr>
<td></td>
<td>500</td>
<td>97.87%</td>
<td>500</td>
</tr>
<tr>
<td></td>
<td>97.87%</td>
<td>95.83%</td>
<td>89.16%</td>
</tr>
</tbody>
</table>
Table 4: Comparison of present paper with (Seehapoch and Wongthanavasu, 2013).

<table>
<thead>
<tr>
<th>Features</th>
<th>Accuracy EmoDB</th>
<th>Accuracy KeioESD</th>
</tr>
</thead>
<tbody>
<tr>
<td>Seehapoch et al., 2013</td>
<td>78.04%</td>
<td>89.23%</td>
</tr>
<tr>
<td>Present paper method</td>
<td>71.23%</td>
<td>83.33%</td>
</tr>
</tbody>
</table>

explicitly stated.

The accuracy scores obtained are really close to those observed in (Seehapoch and Wongthanavasu, 2013). The chosen features allow to perform a good classification, especially when it comes to determine if there is stress or not.

6 CONCLUSION

The main goal of this work, which was to detect stress through speech analysis, has been completed on three different datasets: i) EmoDB (German), ii) KeioESD (Japanese) and iii) RAVDESS (English). The use of the mean energy, the mean intensity and MFCCs proved to be good features for speech analysis, especially the MFCCs. The best way to use these MFC coefficients is the computation of the mean and the standard deviation of each of them, instead of using them as a single feature which can lead to very large feature sets. Neural Networks show the best results even if Support Vector Machines are really close. Both algorithms perform really well for such classification problem.

To conclude, it is interesting to note that the length of audio files does not have a big impact. The results for the EmoDB and KeioESD datasets are really close even if the audio length is not the same (about 3 seconds for the first one and about half a second for the latter).

The results obtained were satisfying but there is however room for improvement. To increase the accuracy scores, features such as formants, MFCCs deltas or speech rate could be added to the feature set and thus used for classification. More time could also be spent on algorithms optimization. A set of parameters with a range of value have been chosen but this range could be increased and more parameters could be used for a better tuning. To finish, acquiring better datasets with much more data would be ideal. The fact that the data is spoken by actors does not exactly reflect what he is feeling. Because of the complexity of emotions and the effects behind, having data from a lot of different people in real life situations would probably give interesting results.

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REFERENCES


