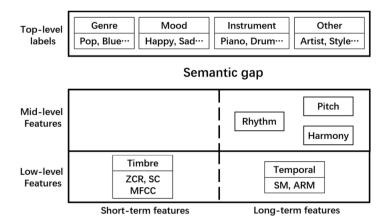
Music Genre Classification with LSTM using Time and Frequency Domain Features

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The Problem Statement

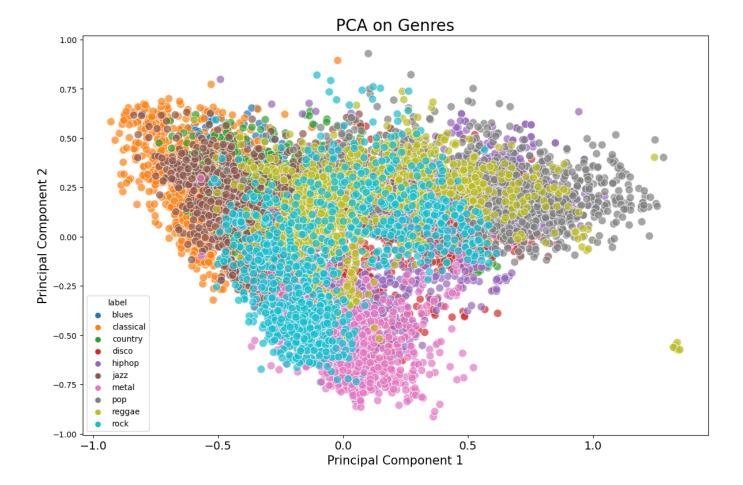
This project aims to classify music genre using deep feature extraction by LSTM. This problem is originally solved by Yinhui et al [8] and through this project I made an attempt to implement this paper.

Motivation



Human's understanding of music is greatly influenced by semantic labels like genre, mood, instruments used and other features like artist, style etc. But they cannot be obtained from low level features or mid level features because of the semantic gap that exists. Hence, there is a need for an algorithm that can bridge this gap.

Short-term features are less useful for classifying music genres than deep features produced by deep learning models. So in order to produce deep features and classify music genres, this project employs an LSTM model and compares it with other models like KNN, SVM, 1-D CNN and 2-D CNN.

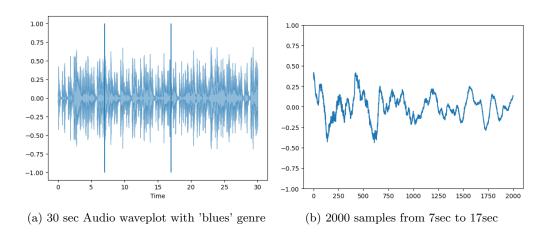


Dataset

The dataset that we are using is available on Kaggle. It is a music collection consisting of 10 music genres with 100 audio files for each genre. The 10 music genres in the collection are blues, classical, country, disco, hiphop, jazz, metal, pop, reggae and rock. The audio files are saved in waveform with a length of 30 seconds.

Visualisation

To visualise the audio dataset in time domain, we used librosa.display.waveshow.



Similarly we can visualise time domain features like zero crossing rate, which tells us the rate at which the signal changes it sign. From figure below we can see that classical and jazz have lower zer values compared

to other genres.

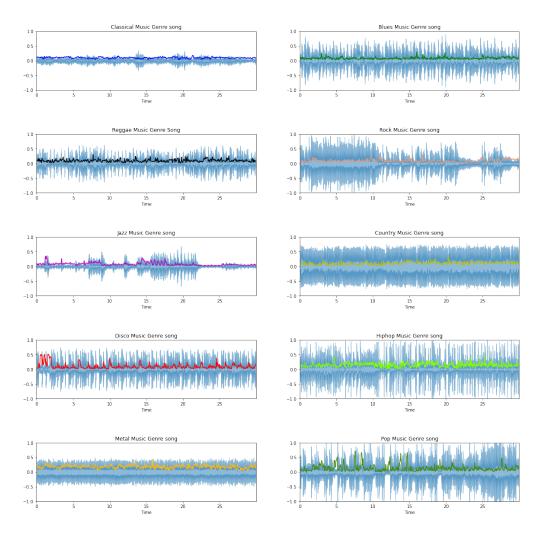
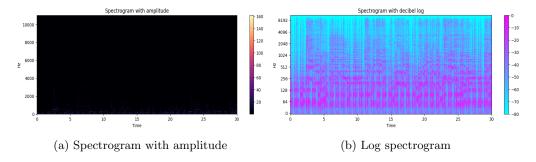
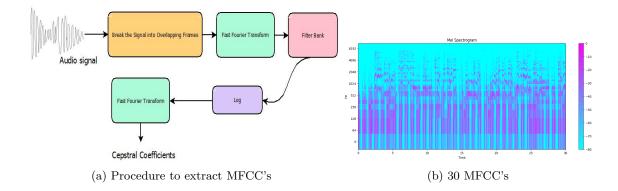


Figure 2: Zero crossing rate for music with different genres

To view frequency domain features we first take short time fourier transform (stft) of the audio files with window size of 2048 and hop length of 512 to get a spectrogram which you can think of as a bunch of FFTs stacked on top of each other. It is a way to visually represent a signal's loudness, or amplitude, as it varies over time at different frequencies. But unfortunately this spectrogram doesn't have much information for us to analyse this is because we humans preceive sounds which is concentrated in narrow range of frequencies and amplitudes so we have to convert the linear scale to log scale to clearly see audible frequencies.



Since human's perception of sound is not linear i.e. we can easily tell the difference between lower frequencies but not between the ones on higher end. So we used a Mel-scale to obtain a set of features called Mel frequency cepstral coefficients or MFCCs for our classification task. Figure 2 shows how 30 MFCC's evolve with time.



Pre-processing

For pre-processing we perform data normalisation, noise-filtering, silence reduction and removing any inconsistent values. Since GTZAN dataset has only 1000 audio files which is not enough to train our model efficiently so we perform data augmentation by splitting our 30 sec audio files into 10 segments of 3 sec each and then we divide this 3 sec file into 30 time steps and for each time step we calculated zer values which finally gave us the feature set with dimension of 30×1 , similarly we can calculate 30 mfcc's for each segment. The output MFCC feature set is a two-dimensional array with 30×1293 . But since we cannot input this to LSTM. Therefore, the mean and variance of each row were calculated as features and the final shape of MFCC feature set is obtained i.e. 60×1 .

Long Short Term Memory (LSTM)

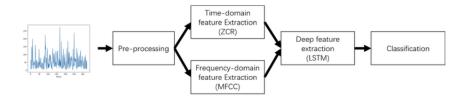


Figure 5: Genre Classification Using LSTM

The two short-term features i.e. zero crossing rate (ZCR) (time domain feature) and mel-frequency spectral coefficients or MFCC's (frequency domain feature) that we calculated in pre-processing stage are then fed to the LSTM with 128 units. Then we train our LSTM network and extracted the output of the LSTM layer which has the dimension of (Number of samples X 128). There are two kinds of outputs depending on the input feature set: (1) deep feature generated from ZCR; (2) deep feature generated from MFCC; finally we used this output as feature set to classify the genre of music using support vector machine (SVM) and k-nearest neighbors (KNN).

The benefit of LSTM is that, on the one hand, music features are time-serial, making them appropriate for LSTM. On the other hand, LSTM was created to address the vanishing gradient issue with recurrent neural networks (RNN). The LSTM's memory cells provide it the ability to retain and access data for extended periods of time and produce better long-term features. LSTM's usually tend to overfit and to overcome this issue we used batch normalization and dropout layers to bring down the complexity of the network.

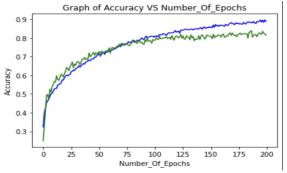
Experimental results and Analysis

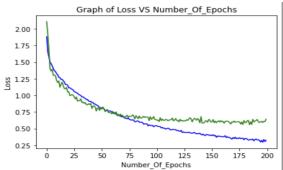
Classification accuracy without LSTM			
Short-term Feature	ZCR	MFCC	
SVM	23.67%	64.33%	
KNN	19%	31.67%	

Table above shows the accuracy of music classification without deep feature extraction layer (LSTM layer). The lowest classification accuracy is 23.67% when music is classified by SVM based on ZCR. While changing the classifier to KNN, the accuracy is still low. However, The classification accuracy significantly increases to 64.33% when music is classified by SVM based on MFCC. Table also shows that compared to KNN, SVM has better performance in all aspect.

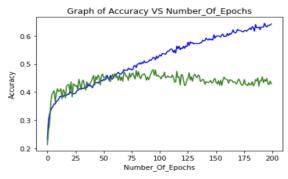
Classification accuracy with LSTM			
Short-term Feature	ZCR	MFCC	
SVM	44.4%	81.8%	
KNN	41.25%	77.8%	

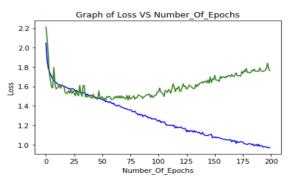
Table above shows the classification accuracy using deep feature extraction layer. Compared to previous table, classification accuracy increases greatly in most cases regardless of SVM classifier based on ZCR. While classifier changes to KNN, the accuracy of the model based on ZCR increases by more than 117%, from 19.0 to 41.25. The highest classification accuracy appears on SVM classifier based on the MFCC feature with 81.8.





(a) Accuracy curve for LSTM training with only (b) Loss curve for LSTM training with only MFCC MFCC features





(a) Accuracy curve for LSTM training with only ZCR (b) Loss curve for LSTM training with only ZCR features tures

Future work

- 1. Test the current model on more robust dataset like FMA [4]
- 2. Use more kinds of low-level features like spectral centroid, RMS, chroma etc.
- 3. Optimising the deep learning model by using bi-directional LSTM or advance LSTM.
- 4. Further explore the relationship between deep features and mid-level features.

The Team

For further queries feel free to email: Naresh Kumar Kaushal

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- [8] [5] [1] [6] [3] [2] [4] [7]