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| **Music Genre Recognition Using Temporal Features** |

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**Abstract**

This project presents an approach for the music genre classification problem. The proposed approach uses temporal feature vector and weighted voting to improve the prediction accuracy. Classical machine learning algorithms such as Naïve-Bayes, k Nearest-Neighbors, and Support Vector Machines are employed and weighted voting procedures were employed in order to enhance final prediction results. Experiments were carried out on a dataset obtained from Music Analysis, Retrieval and Synthesis for Audio Signals (MARSYAS) which is an open source software framework and data collection of audio files. The dataset contains 1,000 audio files categorized in 10 musical genres. Experimental results show that the proposed ensemble approach produces better results than the ones obtained from individual classifiers.

**1 Introduction**

Duke Ellington who once very wisely said: There are simply two kinds of music, good music and the other kind. The only yardstick by which the result should be judged is simply that of how it sounds. If it sounds good, it's successful; if it doesn't it has failed. Exploring different genres of music is about wanting to know a little bit more about some of the things we humans can do, the feelings we didn't want to leave unsaid; the messages we’ve wanted to get across. Today with the amount of music we have, automatic procedures capable of dealing with large amounts of music in digital formats are imperative, and Music Information Retrieval (MIR) has become an important research area. An important task in MIR is Music Genre Classification problem, music genres are categorical labels created by experts in order to identify the style of the music. The music genre is a descriptor that is largely used to organize collections of digital music. It is not only a crucial metadata in large music databases and electronic music distribution (EMD).

The music can be considered as a high-dimensional digital time-variant signal and considering the amount of music data we have today; it is a good opportunity to automate music genre classification using temporal feature vectors. The approach involves classical machine algorithms such as Naïve-Bayes, k Nearest-Neighbors, and Support Vector Machines and using weighted voting procedures to improve final prediction results. The Naïve Bayes is implemented with 10-fold cross validation using ‘e1071’ package in R. For Support Vector Machines, we have implemented ‘libSVM’ with 10-fold cross validation. The K-Nearest Neighbor is implemented with ‘kkNN’ package of R.

**2. Dataset and Features**

**2.1 Dataset**

The dataset we are using is from Marsyas (Music Analysis, Retrieval and Synthesis for Audio Signals) which is an open source software framework for audio processing with specific emphasis on Music Information Retrieval applications. The dataset consists of 1000 audio tracks each 30 seconds long. It contains 10 genres, each represented by 100 tracks. The tracks are all 22050Hz Mono 16-bit audio files in .wav format.

**2.2 Preprocessing**

The first step in preprocessing is to divide the dataset in training, testing and validation sets. For this we have written a python script, which divides the dataset for each genre into 70 songs for testing, 30 songs for testing and 30 songs for training. So finally we have 150 songs in training set and 150 each in validation and training set. After that we have used ‘mkcollection’ which is a simple utility provided by Marsyas framework for creating collection files. The ‘mkcollection’ utility takes the folder containing all songs as an input and the soundfiles residing in that directory or any subdirectories are added to the collection.

**2.3 Feature Extraction**

For features extraction, we are using MARSYAS which is implemented in C++ and retains the ability to output feature extraction data to ARFF format. With the tool bextract.exe, the following features are extracted: Zero Crossings, Spectral Centroid, Spectral Flux, Spectral Rolloff, Mel-Frequency Cepstral Coefficients (MFCC), and chroma features (). The total number of features extracted are 124.

A zero-crossing is a point where the sign of a mathematical function changes (e.g. from positive to negative), represented by a crossing of the axis (zero value) in the graph of the function.\* The spectral centroid is a measure used in digital signal processing to characterize a spectrum. It indicates where the "center of mass" of the spectrum is. Perceptually, it has a robust connection with the impression of "brightness" of a sound.\* Spectral flux is a measure of how quickly the power spectrum of a signal is changing, calculated by comparing the power spectrum for one frame against the power spectrum from the previous frame.\* Spectral rolloff point is defined as the Nth percentile of the power spectral distribution, where N is usually 85% or 95%. The rolloff point is the frequency below which the N% of the magnitude distribution is concentrated.\* In sound processing, the mel-frequency cepstrum (MFC) is a representation of the short-term power spectrum of a sound, based on a linear cosine transform of a log power spectrum on a nonlinear mel scale of frequency.\* Chroma features are an interesting and powerful representation for music audio in which the entire spectrum is projected onto 12 bins representing the 12 distinct semitones (or chroma) of the musical octave.\*

Figure 1. Preprocessing process

**3 Techniques**

**3.1 K-nearest neighbors**

The first machine learning technique used is the simple K nearest neighbor. We used R library to implement the classifier. The results by the standard R library were not satisfactory, and we tried the new KKNN, a library for weighted k-nearest neighbors classification.

The KKNN uses the Minkowski distance to calculate the nearest neighbors of the given data point. The library provides various options for kernels implementation like "triweight", "cos", "inv", "gaussian", our results were best for the Gaussian kernel. We created a script to determine the best kernel and the optimum number of neighbors for the implementation. The bet results were generated by using a Gaussian kernel with five neighbors.

**3.2 SVM**

Support Vector machines (SVM) are considered one of the best machine learning algorithm for multiclass classifications along with the neural networks. Our dataset being small, SVM classifier was expected to perform best among all the techniques used.

The libsvm library is used to implement the SVM classifier for the dataset. The libsvm uses the one vs one classification for all the pairs possible to generate the multiclass classifier. Octave is used to implement the libsvm classifier. We used python script grid.py, provided with the library to find the optimal parameters for the radial kernel to train the data. The parameters for which the validation set was lowest were used to create the final model.

**3.3 Naïve Bayes**

The e1071 library of R is used to implement the classifier for Naïve Bayes. It computes the conditional posterior probabilities of a categorical class variable given independent predictor variables using the Bayes rule. It distributes each class variable as the Gaussian distribution and the parameter tuning is done by the library itself.

**3.3 Neural Networks**

Deep learning is one of the forefront runners in the field of machine learning. The neural network implementation was done using both library and self-written code. But opposite to the initial expectations both the library and ours implementation performed poorly for the data set. We ran various validations and parameter tuning runs but results were not comparable to those of the SVM and K-nearest neighbors. We came to the conclusion that the data was not enough to be classified by the neural network classifier.

**3.4 Weighted Majority Algorithm**

Weighted Majority Algorithm (WMA) is a meta-learning algorithm used to build a joint prediction model from a list of prediction algorithms, which could be any type of learning algorithms, classifiers, or even real human experts. The algorithm assumes that we have no prior knowledge about the accuracy of the algorithms in the list, but there are sufficient reasons to believe that one or more will perform well\*.

We use all but neural network model to build new classifier. We created a script to find the different optimum weights for K-near neighbors, SVM and Naïve Bayes. It runs for all possible weight combinations with step size of 0.01 with range 0 to 100 for each model. As expected the contribution of Naïve Bayes was the lowest among the three classifier as it is the weakest classifier among the three and provides less accuracy as compared to the other algorithms used.

Figure2. WMA Process

**4 Experiments**

**5 Conclusion**

**6 Future Work**

The work done by us could be extended on various dimensions. The dataset used by us was very small to be able to work properly for the neural network model. The training data could be increased by two methods, (a) collect more data or (b) split the single audio files into multiple files. In our analysis collecting more audio files would be a better approach since reducing the length of audio for training data could reduce the accuracy. Since there might be parts of the songs which may be of different genre than the song as a whole. The feature set used by us is of length 124, this could also be increased. One can calculate these features for fixed intervals of the audio and collate then as new feature set.

Other classification algorithms such as decision trees and random forrest could also be created and added to the WMA algorithm.

**7. Contributions**

Table. Contributions

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| --- | --- |
| **Task** | **Member** |
| Appropriate Data Set search | Jivjot, Mangesh, Anuj, Bikram |
| Feature Set Extraction Library Comparisons | Jivjot, Mangesh, Anuj, Bikram |
| Naïve Bayes implementation | Mangesh |
| SVM implementation | Jivjot |
| Neural Networks implementation | Anuj |
| K- Nearest Neighbor implementation | Bikram |
| Weighted Majority Algorithm | Jivjot |
| Poster | Jivjot, Mangesh, Anuj, Bikram |
| Report | Jivjot, Mangesh, Anuj, Bikram |

**References**

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[3]Spectral Centroid: Grey, J. M., Gordon, J. W., 1978. Perceptual effects of spectral modifications on musical timbres. Journal of the Acoustical Society of America 63 (5), 1493–1500, doi:10.1121/1.381843

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