SUBMISSION TO MIREX 2011 GENRE CLASSIFICATION AND AUDIO SIMILARITY TASKS

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ABSTRACT

In this submission, audio features that approximate timbre, rhythm and tempo are used for genre classification and music similarity estimation. This abstract describes the feature set, distance computation method, and classifier model used for the submitted algorithms.

1. INTRODUCTION

In our system, audio data are modeled as long-term accumulative distribution of frame-based spectral features. This is also known as the "bag-of-frames" (BOF) approach wherein audio data are treated as a global distribution of frame occurrences. This approach is widely used in MIREX submissions. For MIREX 2010 genre classification and audio similarity estimation task, the BOF approach was used for some of the top performing systems[1][2].

The features that are extracted from audio files are approximations of timbre, rhythm and tempo. The feature extraction, distance computation and classification algorithms are implemented in MATLAB $^{\oplus}$.

2. FEATURE EXTRACTION

This section describes the processes involved in feature extraction. More detailed explanation can be found on the cited references.

2.1 Audio Preprocessing

The input signal is assumed to be sampled at 22050 Hz, as specified in MIREX wiki¹. The signal is then cut into frames with a window size 512 samples (~23 msec.) and hop size 512 samples.

2.2 Timbre Component

The timbre component is represented by the Mel-Frequency Cepstral Coefficients (coefficients 2:20) [3]. We then model the distribution of the MFCCs for the audio file using a Gaussian mixture model (GMM).

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In this work, we use a single Gaussian represented by its mean μ and covariance matrix Σ .

Other features derive derived from the spectral representation are *spectral flux*, *spectral flux for delta spectra*, *spectral flux for Mel-frequency spectra*, and *spectral flux for Mel-frequency delta spectra*.

2.3 Rhythm Component

The rhythm component based on the Fluctuation Patterns [4] (FPs) of the audio signal. Fluctuation patterns describe the amplitude modulation of the loudness per frequency band.

For each frame, the fluctuation pattern is represented by a 12x30 matrix. The rows correspond to reduced Melfrequency bin while the columns correspond to modulating frequency bands. To summarize the FPs, the median of the matrices is computed. Additional features derived are FP mean and FP standard deviation.

2.4 Tempo Component

The tempo component is derived from a technique using onset autocorrelation [5]. The tempo is computed by taking the first-order difference along time of a Melfrequency spectrogram then summing across frequency. A high-pass filter is used to remove slowly-varying offsets. The global tempo is estimated by autocorrelating the onset strength and choosing the period with the highest windowed peak.

3. MUSIC SIMILARITY ESTIMATION

The timbre, rhythm, and tempo distances are calculated separately. Before they are combined, each distance component is normalized by removing the mean and dividing by the standard deviation of all the distances. Symmetry is obtained by summing up the distances in both directions for each pair of tracks [6].

Distances between timbres are computed by comparing the GMM models. We use symmetric Kullback-Leibler (KL) distance between two models [7]. The Euclidean distance is used to compute distance between rhythms. For tempo distances, a simple absolute distance is computed.

1 http://www.music-

ir.org/mirex/wiki/2011:Audio_Music_Similarity_and_Retrieval

4. GENRE CLASSIFICATION

Genre classification involves training a classifier. The timbre, rhythm and tempo features are combined as a single vector to represent each training file. For the classifier, the Directed Acyclic Graph-Support Vector Machine [8] is used to handle the multi-class problem. For the DAG-SVM classifier, the radial basis function kernel is used.

5. ACKNOWLEDGMENTS

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6. REFERENCES

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