

MIREX AUDIO GENRE CLASSIFICATION

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ABSTRACT

This extended abstract details a submission to the Music Information Retrieval Evaluation eXchange in the Audio Genre classification task, including the feature set used, the feature selection methods and the final classification method.

Index Terms— Audio, Genre, Classification

1. FEATURE SET

We mainly use timbre features to classify the music genre. The features included are Mel-frequency cepstral coefficients (MFCC), the spectral centroid, the MPEG 7's audio spectral flatness (ASF), and some AM modulation features. The MFCC feature is about to characterize the whole spectral envelope and we select the first 13 coefficients as our feature. The spectral centroid is about to characterize the music brightness and darkness. The other two features are more used in the music instrument recognition area (Antti 2001 and Slim 2006), and we believe the genre is much more related with the instrument used in the music. The audio spectral flatness is about to characterize the flatness of each spectral band. We divide the whole frequency band into 20 subbands and the band width is in logarithmic scale, then we calculate the ASF of each band. The last feature is about AM modulation. Strength, frequency and heuristic strength of amplitude modulation is measured at two frequency bands, 4~8Hz and 10~40Hz. In each band, FFT analysis is carried on a windowed envelope and the maxima peak is located. The peak frequency, the difference between the maxima and the band average amplitude and the difference between the maxima and the average amplitude of the whole frequency range under consideration are used as AM features. In summary, we have 13 dimensions for MFCC, 1 dimension for spectral centroid, 20 dimensions for ASF and 3 dimensions for AM modulation for each frame. The frame length is 46.4ms, and the frame shift is 23.2ms. We calculate the mean and variance of each dimension of the feature in 1s and use those as the input for feature selection. At last, for 1s there are 74 real numbers as the output feature.

2. FEATURE SELECTION

We use a simple and intuitive approach called inertia ratio maximization using feature space projection (IRMFSP) as our feature selection method (Geoffroy 2003). This method is about to maximize the ratio between the inter-class distance and the inner-class distance on each dimension of the feature set. In the total 74 dimensions we selected only 40 dimensions which maximize the ratio. We select the feature for each pair of the class, so in some cases one dimension of a certain class is used when compared with some class and is not used when compared with others.

3. CLASSIFICATION

We use the support vector machine as our classifier. We modified the libsvm software package to suit our feature selection method. We use the rbf kernel, and find the best parameters of svm by 5-fold cross-validation. We also use voting method for the last classification stage.

4. References

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