### music structural segmentation by hmm clustering

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#### Abstract

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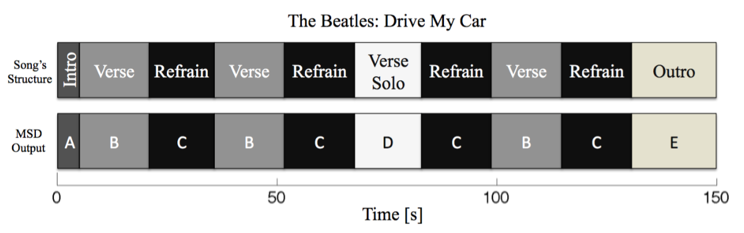
**Index Terms—**Audio, segmentation, structure, HMM, clustering

**1. Introduction**

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**1.1. Problem definition**

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**Figure 1**. Output example of the structural analysis of the song Drive My Car by the Beatles.

**1.2. Application**

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**1.3. Paper organization**

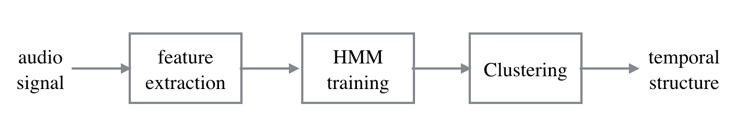
Algorithms for structural segmentation of music take an audio signal as input and give output information about its temporal structure. In our approach, a couple of subtasks can be identified. A rough overview of these subtasks is given in Figure 2.

**2. Related work**

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**3. algorithm overview**

Algorithms for structural segmentation of music take an audio signal as input and give output information about its temporal structure. In our approach, a couple of subtasks can be identified. A rough overview of these subtasks is given in Figure 2. This section will be organized accordingly.



**Figure 2**. System overview: music structural segmentation by clustering based on HMM.

**3.1. Low-level feature extraction**

Our method of low-level feature extraction is based on the audio spectrum envelope, audio spectrum projection and sound model as described in the MPEG-7 standard [3]. This step yields a 21-dimensional audio spectrum projection feature vectors. The first 20 dimensions represent the spectral shape, by reducing dimension of the original normalized power spectrum envelope (64-dimension) using PCA method. And the final dimension is the relative power of each block window, which will be discussed in section 3.1.3.

*3.1.1. Beat tracking*

The dataset audios are mixed from stereo to mono by taking the mean of two channels. The original sampling rate is 44100Hz which is not necessary to be that high, however we downsampled to 11025Hz.

We use a hop size equal to the beat-length of the music (typically 400-600ms), and a window of three times the hop. While beat-lengths are estimated by a beat-tracking algorithm developed by labrosa [4]. For accuracy, we use the highest estimation of tempo instead of the lowest estimation (typically half of the highest tempo).

*3.1.2. Audio spectrum envelope*

The fundamental audio feature is the audio spectrum envelope. It is a standard MPEG-7 descriptor that uses logarithmic frequency scale and decibel scale to describe the power spectrum. Using logarithmic scale imitate the response of the human ear better.

We implement a constant Q transform (CQT) function to extract the audio spectrum envelope. The bin number of CQT is 8, which means there are 8 frequency bands in an octave. The lowest and highest frequency for CQT are 62.5Hz and 16kHz respectively. The output spectrum of the CQT is multiplied by its conjugate which yields a power spectrum.

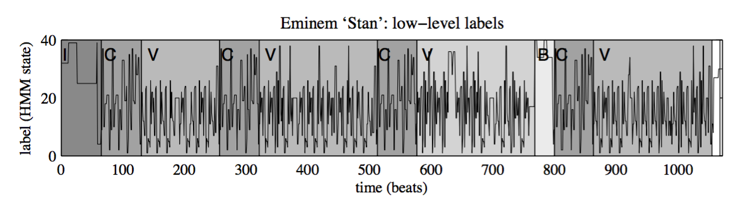
*3.1.3. Audio spectrum projection*

We applied the principal component analysis on the whole feature vector sequence to reduce the dimensionality of feature vector. Only the first 20 principal components are used. We also append the L2-norm, which represents the over all power, of each feature vector, to each feature vector. The norm is normalized by the max value, making all its values between [0, 1]. These 21-dimensional feature vectors, or audio spectrum projection (since PCA is a linear transform that project the feature vector to the principal components), are our low level features.

**3.2. HMM**

We use Baum–Welch algorithm to train an 80-state hidden Markov model on the entire sequence of feature vectors, with a single Gaussian output distribution for each state, and a single covariance matrix tied across all states. We then use Viterbi algorithm to decode the most likely state sequence, which is equivalent to assign a state to each beat.

Fig. 3 shows the resulting state sequence labels for the sample track.

**Figure 3**. Sequence of low-level labels for the sample track.

**3.3. Clustering**

We use a sliding window (16-beat long) to estimate the local state distribution at each beat of the sequence by counting 15 neighbouring states. This yields a 80-bin histogram of local states.

We then use kmeans with K=6 to cluster the histograms (state distributions). The value of K, 6, represents the number of common segmentation type of a song.

**4. Evaluation**

We use Beatle’s data set…

**4.1. Conditional Entropies**

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**4.2. Pair-wise F-measure**

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**5. discussion**

Why our result is not good?

What may be the problem?

**6. conclusion**

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**7. References**

[1] A.B. Smith, C.D. Jones, and E.F. Roberts, “Article Title,” *Journal*, Publisher, Location, pp. 1-10, Date.

[2] Jones, C.D., A.B. Smith, and E.F. Roberts, *Book Title*, Publisher, Location, Date.

[3] Jones, C.D., A.B. Smith, and E.F. Roberts, *Book Title*, Publisher, Location, Date.

[4] D. Ellis., “Beat Tracking by Dynamic Programming,” *J. New Music Research*, Special Issue on Beat and Tempo Extraction, vol. 36, no. 1, pp. 51-60, 2007.