

AM-FM Modulation Features for Music Instrument Signal Analysis and Recognition

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1. Outline - Contributions

- **Motivation:** Small fluctuations or **micro-modulations** occur naturally in both human voice and musical instruments.
- ➔ Exploration of micro-modulations, based on nonlinear **AM-FM modeling** using the **multiband Gabor Energy Separation Algorithm** (Gabor-ESA) [7].
- AM-FM modulations can capture the fine-structure and the rapid fluctuations of musical signals.
- Application of the **Iterative-ESA** [5] for the estimation of the center frequencies f_c of the Gabor filterbank.
- **Goal:** Alternative feature configurations that offer a modest improvement to the performance of a recognition task.

2. Amplitude and Frequency Modulation (AMFM)

- Musical signals represented as a combination of different "resonances", approximately corresponding to oscillation systems formed by the instruments' characteristics and the sound production procedure.

- We model each resonance component of music signals as an amplitude and frequency modulated sinusoid (AM-FM signal), and the whole signal as a sum of such AM-FM components.

AM-FM model

$$s(t) = \sum_{i=1}^K \alpha_i(t) \cos(\varphi_i(t))$$

instantaneous amplitude
phase

Energy Separation Algorithm (ESA)

estimates the instantaneous amplitude and frequency

$$|\alpha(t)| \approx \frac{\Psi[x(t)]}{\sqrt{\Psi[\dot{x}(t)]}} \quad f(t) \approx \frac{1}{2\pi} \sqrt{\frac{\Psi[\ddot{x}(t)]}{\Psi[\dot{x}(t)]}}$$

Teager Energy Operator $\Psi[x] = \dot{x}^2 - x\ddot{x}$ ($\dot{x} = dx/dt$)

Gabor-ESA

combination of the continuous time ESA and Gabor filtering of the signal (smoother instantaneous estimates)

$$\Psi[s(t) * g(t)] = \left[s(t) * \frac{dg(t)}{dt} \right]^2 - (s(t) * g(t)) \left[s(t) * \frac{d^2g(t)}{dt^2} \right]$$

Modulation Features

Mean Instantaneous Amplitude:

- short-time mean of the $|\alpha_i(t)|$ for each resonance i .
- Parameterizes the resonance amplitudes and captures part of the nonlinear behavior of the signal.

Mean Instantaneous Frequency:

- short-time weighted mean of the $f_i(t)$.
- Models the time-varying frequency of the resonance.
- Provides information about the signal's fine structure taking advantage of ESA's excellent time resolution.

3. Experimental Configuration and Results

Experimental configuration

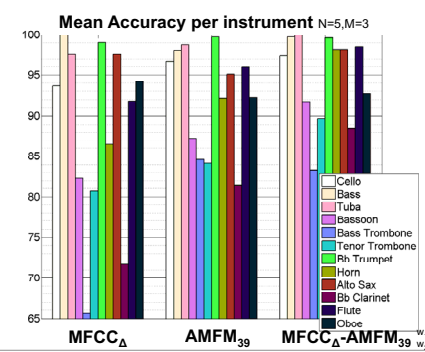
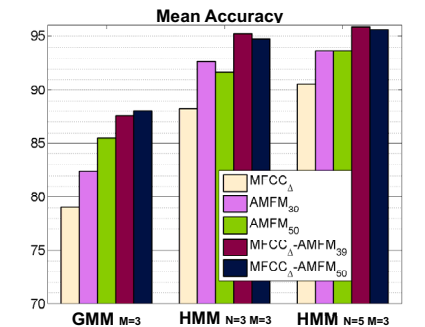
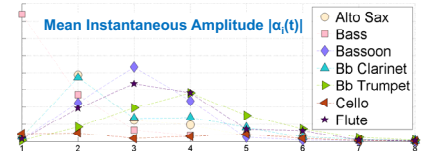
- Input signal: $F_s = 44.1$ KHz.
- Dynamic range: piano to forte.
- **12 Gabor Filters** with mel placing and 50% overlap. Optimum joint time-frequency resolution.
- Analysis on 30 ms segments, updated every 15 ms.

Experimental Setup

- 2070 notes, 12 instruments
- 5 feature sets
 1. 24 AMFM (12 m-IAM + 12 m-IFM)
 2. 72 AMFM $_{\Delta}$
 3. AMFM $_{30}$ (18 m-IAM + 32 m-IFM) after PCA
 4. AMFM $_{39}$ (12 m-IAM + 27 m-IFM) after PCA
 5. 39 MFCC $_{\Delta}$
- AMFM $_{\Delta}$ + MFCC $_{\Delta}$
- AMFM $_{30}$ + MFCC $_{\Delta}$
- AMFM $_{39}$ + MFCC $_{\Delta}$
- 2 classification methods
 - GMM M[1-3] mixtures
 - Multi-stream HMM N[3-9] states, M[1-3] mixtures

Results

- Error Rate Reduction (ERR) = 56% (AMFM $_{39}$)
- Enhanced discrimination except for: Bass, Bass Trombone & Oboe
- Improves: Bass Trombone, Horn, Clarinet
- ERR = 15%, 38%, 33% (AMFM $_{39}$)
- Better recognition on all, except Bass, Saxophone & Oboe
- High recognition: Trumpet, Tuba, Bass & Cello
- Low recognition: Clarinet & Flute, Bass & Tenor Trombone
- High recognition: Bass, Trumpet, Tuba & Sax
- Low recognition: Clarinet, Flute & Oboe, Bass & Tenor Trombone, Bassoon & Horn



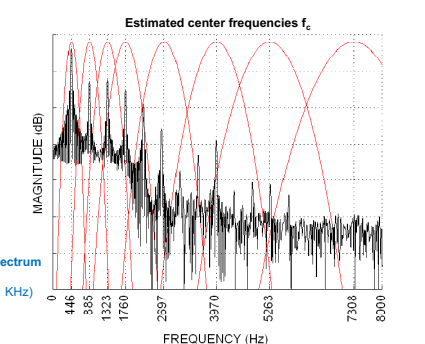
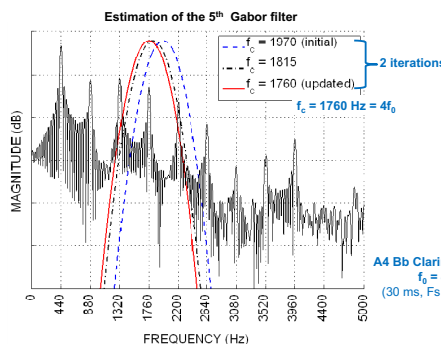
4. Iterative-ESA

- Iterative application of ESA to the Gabor filtered signal.
- Reduces the importance of having good initial estimates of the filterbank's center frequencies, since they are iteratively adjusted after every iteration.

- f_c : tend to converge on frequencies close to integer multiples of the fundamental frequency, i.e., the harmonics.

- Some Gabor filters favored to converge at the same center frequency.

- Possible estimation of the harmonic content of a tone.



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