

Introduction

- Aim to transcribe polyphonic drum sequences from audio to a symbolic representation.
- Overlapping sounds are problematic with many current systems.
- Individual drum hits can be recognised reliably.
- Use source separation to segregate individual instruments to own streams.

Signal model

- Signal represented as magnitude spectrogram.
- Drum sounds' spectral content is relatively static over the whole hit.
- Sum of N sources with fixed spectra $S_n(f)$ and timevarying gains $a_{n,t}$

$$X_t(f) \approx \sum_{n=1}^N a_{n,t} S_n(f)$$
(1)

• Spectrogram wih very coarse frequency resolution: 5 approximately log-spaced bands.



FIGURE 1: An example decomposition of a drum loop into three sources, Bark band frequency resolution.

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Drum Transcription with Non-negative Spectrogram Factorisation

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Method

- \bullet Calculate template source spectra S from training samples. -Normal NMF used to separate each example hit to one source.
- -Source spectra of each drum type are averaged over examples.
- Estimate corresponding time-varying gains a to fit model and input signal together.
- -Minimise cost function (divergence) between original spectrogram X and the estimate M.
- Iterative updates until convergence:

$$a_{n,t} \leftarrow a_{n,t} \frac{\sum_{f} X_t(f) S_n(f) / M_t(f)}{\sum_{f} S_n(f)}.$$
 (2)

• Detect onset times from estimated gains a.



Demonstrational signals are available at http://www.cs.tut.fi/~paulus/research.html

- Spectral content fixed.
- \rightarrow Enough to analyse only time-varying gains a.
- Threshold value learned from training data.

- System evaluated with acoustic recordings. -Four data sets with differring instruments and acoustic environments.
- -4-fold cross-validations, performance metrics calculated over all folds.
- Performance compared with two other systems with same material.
- Prior Subspace Analysis (PSA) - FitzGerald et al., "Prior subspace analysis for drum transcription", AES 2003.
- Same signal model as in proposed method.
- Solve time-varying gains with matrix inverse and ICA, onset detection with thresholding.
- Event-based recognition

Onset detection Motivated by auditory system inspired sub-band amplitude envelope method.

- -View relative difference instead of absolute.
- \rightarrow Each sub-band envelope of identical form.



FIGURE 3: Onset detection method.

Evaluations

- Gillet & Richard, "Automatic transcription of drum loops", ICASSP 2004.







- and hi-hats.
- sation.
- good.
- a failure.

- More traditional pattern recognition approach. - Detect onsets from acoustic signal, segment, extract features, classify with N binary SVMs.



Conclusions

• Aim to transcribe signals containing bass drum, snare drum

• Use fixed template spectra and estimate time-varying gains with algorithm stemming from non-negative matrix factori-

• Detect onsets from gains with auditory motivated method. • When target signal and used model match, the result is

- Signals containing also other instruments will result into