The Application of Psychoacoustic Audio Analysis Techniques to Electroacoustic Music for the Purpose of Visualization

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Introduction

Objective

- Motivation
 - Electroacoustic Music
 - Use of Human Hearing Models
 - Lack of Source Segregation
- Human Hearing Model Resources
- Progress
 - Brainstorming Results
 - Graphical Score Prototype

Objective

The Goal

create algorithms which use models of human hearing to extract audio properties from recorded electroacoustic music

- identification of appropriate audio properties
- design of audio property algorithms
- validation of those algorithms against human performance

- Electroacoustic Music
- Use of Human Hearing Models
- Lack of Source Segregation

Electroacoustic Music — Unusual Type of Music

- definition involves electronic technology for the compositional manipulation of sound
- not restricted by physics
- the blurry lines of perception
- no Western musical score

Why use models of human hearing?

- Physical representations lose aspects of what humans hear.
 - missing fundamental
 - streaming
 - perceived loudness

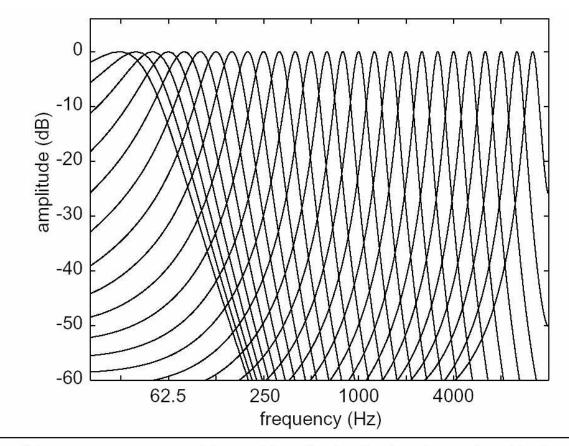
Why not segregate sources?

- Source Segregation
 - Holy Grail translate a continuous pressure variation into Western musical notation
 - first separate the energy from different auditory sources
- Sans Source Segregation
 - the nature of electroacoustic music
 - example Dr. Scheirer's tempo algorithm

Human Hearing Model Algorithms

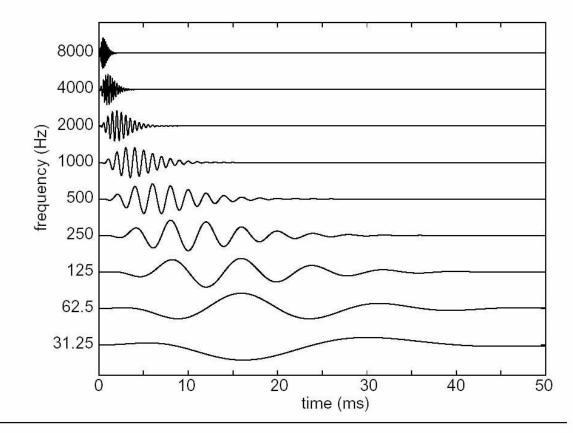
- Dr. Patterson's Auditory Image Model (AIM)
- Drs. Meddis and Hewitt's Correlogram
- Dr. Ellis' Weft

Dr. Patterson's Gammatone Filterbank



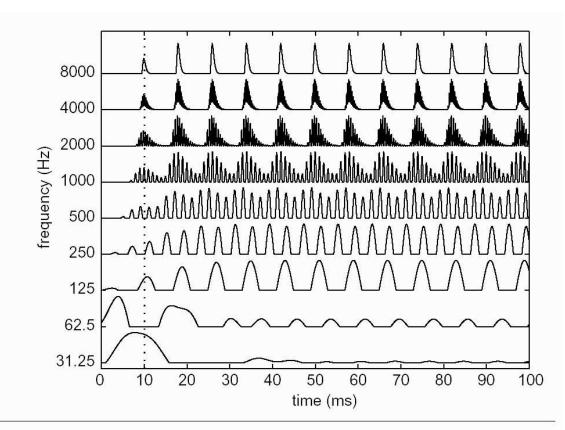
Overall frequency response of the cochlear filterbank, plotted on a logarithmic frequency scale (every second filter is shown).

Basilar Membrane Motion



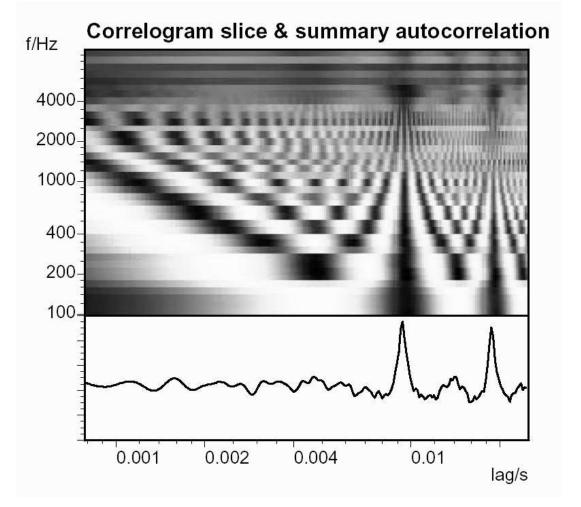
Impulse responses of nine cochlea bandpass filters (one filter is shown per octave). Their amplitudes have been normalized to a uniform scale for display purposes.

Drs. Meddis and Hewitt's Cochleogram



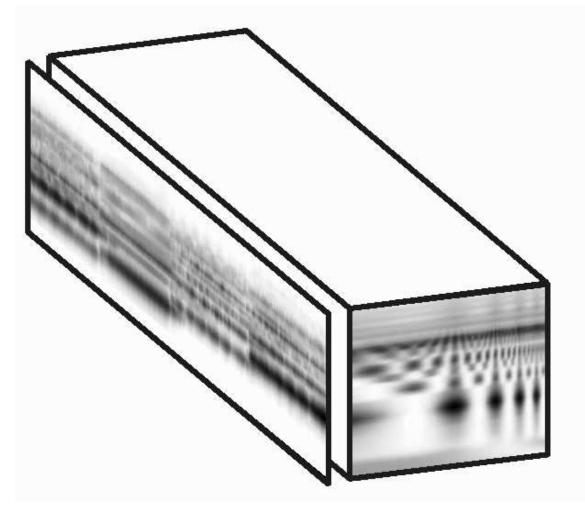
Responses of nine cochlea bandpass filters (one filter is shown per octave) to the 125 Hz sawtooth signal after half-wave rectification and light smoothing intended to model inner hair cell transduction. The output amplitudes have been normalized to a uniform scale for display purposes.

Drs. Meddis and Hewitt's Correlogram

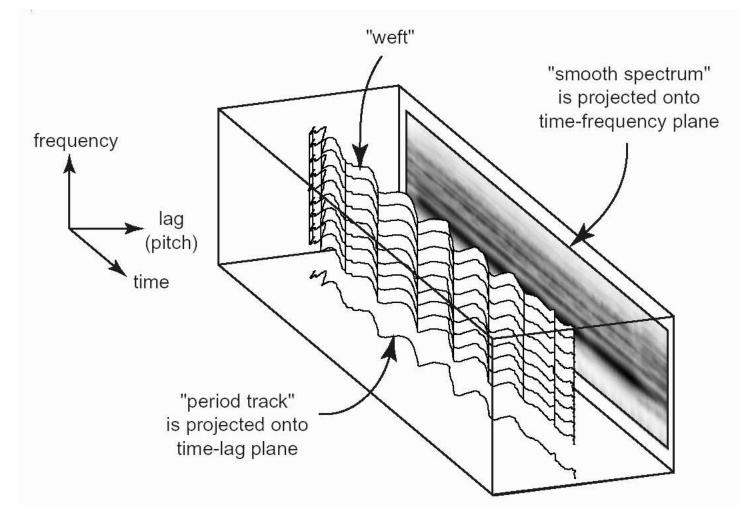


D. P. W. Ellis, Prediction Driven Computational Auditory Scene Analysis, PhD thesis, Massachusetts Institute of Technology, June 1996.

Drs. Meddis and Hewitt's Correlogram



Dr. Ellis' Weft



Progress

- Brainstorming Results
- Graphical Score prototype

Progress

Brainstorming Results

 Low Level amplitude brightness presence of harmonics

loudness panning energy compression spectral centroid SNR perceived energy

Medium Level
impulsiveness
commodulation
ecological sounds
harshness

self-similarity noise/tonal-ness event durations

texture/event-ness

backwardness fundamental frequencies number of sound sources energeticness

• High Level

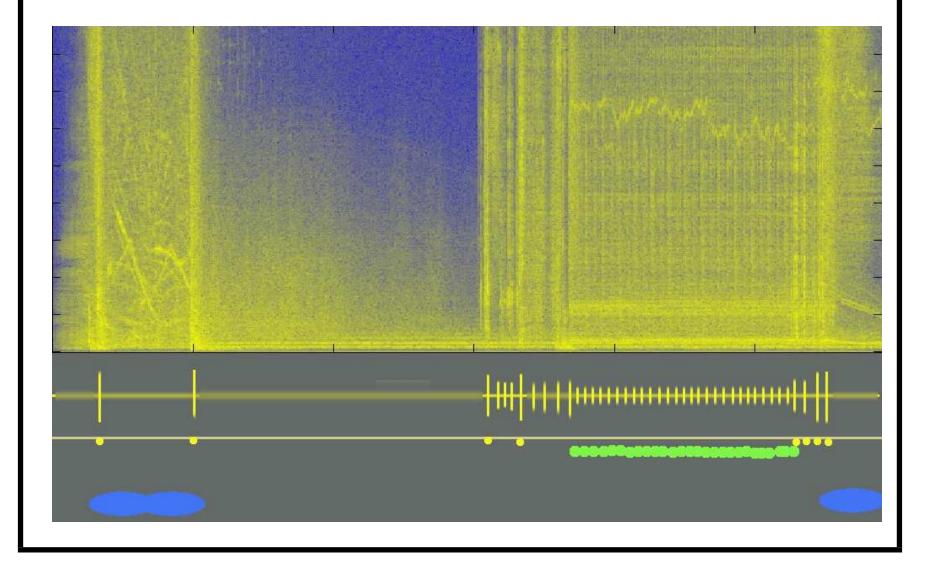
impulses

reverb important moments number of layers conventionalness

event types mood sound quality musical key predictableness time signature presence of effects text

Progress

Graphical Score Prototype



Conclusion

- create algorithms to extract audio properties from recorded electroacoustic music
- use human hearing models by Patterson, Meddis and Hewitt, and Ellis
- begun the process of designing the algorithms