Introduction to Audio Content Analysis

Module 3.6: Instantaneous Features

alexander lerch



introduction overview

corresponding textbook section

Section 3.6

■ lecture content

- introduction to the concept of features
- timbre
- spectral shape instantaneous features

■ learning objectives

- describe the process of feature extraction
- list possible pre-processing option and explain potential use cases
- describe the general impact of spectral shape on timbre perception
- summarize features, describe their computation, and discuss their meaning



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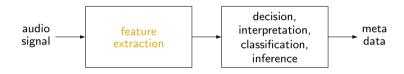
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instantaneous features introduction



remember the flow chart of a general ACA system:



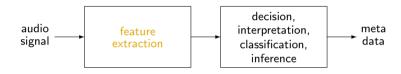
feature

- terminology
 - audio descriptor
 - instantaneous/short-term/low-level feature
- characteristics:
 - not necessarily musically, perceptually, or semantically meaningful
 - low-level: usually one value per block

instantaneous features introduction



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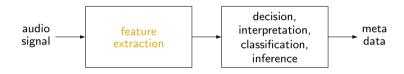
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instantaneous features introduction



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rview intro timbre spectral features MFCCs tonalness technical learned summary ○●○○ ○○ ○○○○ ○○○○○ ○○○○○ ○○○○ ○○○

instantaneous features

a feature . . .

- is task-specific, i.e. holds descriptive power relevant to the task,
- may be custom-designed, chosen from a set of established features, or learned from data,
- can be a representation of any data (audio, meta data, other features, ...),
- is not necessarily musically, perceptually, or semantically meaningful or interpretable
- also: non-redundant, invariant to irrelevancies

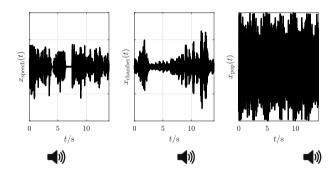


view intro timbre spectral features MFCCs tonalness technical learned summary $00 \bullet 0$ 00 00000 000000 0000 00 0000 0

instantaneous features feature example

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waveform envelope of three different signals



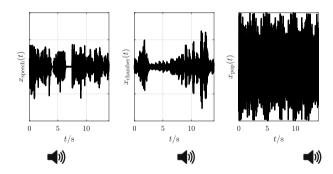
envelopes of waveforms can have distinct shape

⇒ a feature describing envelope shape could help to distinguish these signal types



instantaneous features feature example

waveform envelope of three different signals

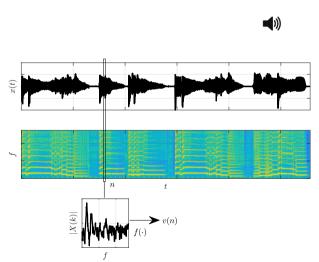


- envelopes of waveforms can have distinct shape
- \Rightarrow a feature describing envelope shape could help to distinguish these signal types $^{
 m I}$



instantaneous features feature extraction

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- repeat for every block
- repeat for every feature: Spectral Centroid, RMS, MFCCs, . . .
- \Rightarrow feature matrix per audio input



definition (American Standards Association)

...that attribute of sensation in terms of which a listener can judge that two sounds having the same loudness and pitch are dissimilar

What is the problem with this definition?



¹A. S. Bregman, Auditory Scene Analysis. MIT Press, 1994.

²S. McAdams and A. Bregman, "Hearing Musical Streams," Computer Music Journal, vol. 3, no. 4, pp. 26–60, Dec. 1979, ISSN: 0148-9267.

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What is the problem with this definition? Bregman:¹

- implies that timbre only exists for sounds with pitch!
- 2 only says that timbre is not loudness and pitch



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- implies that timbre only exists for sounds with pitch!
- 2 only says that timbre is not loudness and pitch
- → [timbre is] "...the psychoacoustician's multidimensional waste-basket category for everything that cannot be labeled pitch or loudness." ²

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timbre is

- a function of temporal envelope
 - attack time characteristics
 - amplitude modulations
 - . . .
- a function of spectral distribution
 - spectral envelope
 - number of partials
 - energy distribution of partials
 - ...

when dealing with complex mixtures of sound, it is very difficult (maybe impossible?) to extract detailed temporal information for individual tones

timbre introduction 2/2

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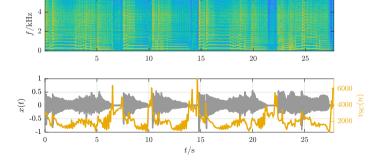
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spectral shape features spectral centroid

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$$v_{\text{SC}}(n) = \frac{\sum_{k=0}^{K/2-1} k \cdot |X(k,n)|}{\sum_{k=0}^{K/2-1} |X(k,n)|}$$







spectral shape features spectral centroid



$$v_{\rm SC}(n) = \frac{\sum_{k=0}^{K/2-1} k \cdot |X(k,n)|}{\sum_{k=0}^{K/2-1} |X(k,n)|}$$

common variants:

- power spectrum
- logarithmic frequency scale

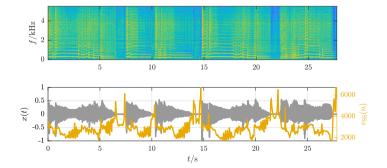
$$v_{\text{SC,log}}(n) = \frac{\sum\limits_{k=k(f_{\min})}^{\mathcal{K}/2-1} \log_2\left(\frac{f(k)}{f_{\text{ref}}}\right) \cdot |X(k,n)|^2}{\sum\limits_{k=k(f_{\min})}^{\mathcal{N}/2-1} |X(k,n)|^2}$$

spectral shape features spectral spread

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$$v_{\rm SS}(n) = \sqrt{\frac{\sum\limits_{k=0}^{\mathcal{K}/2-1} (k - v_{\rm SC}(n))^2 \cdot |X(k,n)|^2}{\sum\limits_{k=0}^{\mathcal{K}/2-1} |X(k,n)|^2}}$$







spectral shape features spectral spread

$$v_{ ext{SS}}(n) = \sqrt{rac{\sum\limits_{k=0}^{\mathcal{K}/2-1} (k - v_{ ext{SC}}(n))^2 \cdot |X(k,n)|^2}{\sum\limits_{k=0}^{\mathcal{K}/2-1} |X(k,n)|^2}}$$

common variants:

■ same variants as with *Spectral Centroid*, e.g. logarithmic:

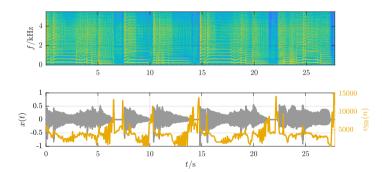
$$v_{\mathrm{SS,log}}(\textit{n}) = \sqrt{rac{\sum\limits_{k=k(f_{\mathrm{min}})}^{\mathcal{K}/2-1} \left(\log_2\left(rac{f(k)}{1000\,\mathrm{Hz}}
ight) - v_{\mathrm{SC}}(\textit{n})
ight)^2 \cdot |X(k,\textit{n})|^2}{\sum\limits_{k=k(f_{\mathrm{min}})}^{\mathcal{K}/2-1} |X(k,\textit{n})|^2}}$$

spectral shape features spectral rolloff

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$$v_{ ext{SR}}(extbf{ extit{n}}) = i \quad ext{at } \sum_{k=0}^i |X(k, extit{n})| = \kappa \cdot \sum_{k=0}^{\mathcal{K}/2-1} |X(k, extit{n})|$$







spectral shape features spectral rolloff



$$v_{\mathrm{SR}}(\textit{n}) = i \quad \mathsf{at} \; \sum_{k=0}^{i} |X(k,\textit{n})| = \kappa \cdot \sum_{k=0}^{\mathcal{K}/2-1} |X(k,\textit{n})|$$

common variants:

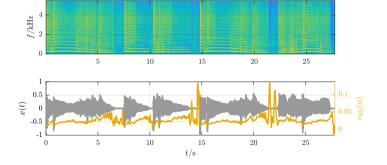
- scaled to frequency
- power spectrum

spectral shape features spectral decrease

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$$v_{\text{SD}}(n) = \frac{\sum_{k=1}^{\kappa/2-1} \frac{1}{k} \cdot (|X(k,n)| - |X(0,n)|)}{\sum_{k=1}^{\kappa/2-1} |X(k,n)|}$$







spectral shape features spectral decrease

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common variants:

restricted frequency range:

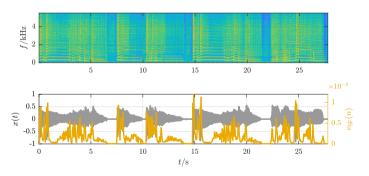
$$v_{ ext{SD}}(n) = rac{\sum\limits_{k=k_{ ext{l}}}^{k_{ ext{u}}} rac{1}{k} \cdot ig(|X(k,n)| - |X(k_{ ext{l}}-1,n)| ig)}{\sum\limits_{k=k_{ ext{l}}}^{k_{ ext{u}}} |X(k,n)|}$$

spectral shape features spectral flux

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$$v_{ ext{SF}}(n) = rac{\sqrt{\sum\limits_{k=0}^{\mathcal{K}/2-1} ig(|X(k,n)| - |X(k,n-1)|ig)^2}}{\mathcal{K}/2}$$







spectral shape features



$$v_{ ext{SF}}(n) = rac{\sqrt{\sum\limits_{k=0}^{\mathcal{K}/2-1} \left(|X(k,n)| - |X(k,n-1)|
ight)^2}}{\mathcal{K}/2}$$

common variants:

$$egin{array}{lcl} v_{
m SF}(n,eta) &=& rac{\sqrt[eta]{\sum\limits_{k=0}^{\mathcal{K}/2-1}}\left(|X(k,n)|-|X(k,n-1)|
ight)^eta}{\mathcal{K}/2} \ &v_{
m SF},_{\sigma}(n) &=& \sqrt{rac{2}{\mathcal{K}}\sum\limits_{k=0}^{\mathcal{K}/2-1}\left(\Delta X(k,n)-\mu_{\Delta X}
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m SF},_{
m log}(n) &=& rac{2}{\mathcal{K}}\sum\limits_{k=0}^{\mathcal{K}/2-1}\log_2\left(rac{|X(k,n)|}{|X(k,n-1)|}
ight) \end{array}$$

signal model:

convolution of excitation signal and transfer function

$$x(i) = e(i) * h(i)$$

$$X(j\omega) = E(j\omega) \cdot H(j\omega)$$

$$\log (X(j\omega)) = \log (E(j\omega) \cdot H(j\omega))$$
$$= \log (E(j\omega)) + \log (H(j\omega))$$

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$$c_{x}(i) = \mathfrak{F}^{-1} \{ \log (X(j\omega)) \}$$

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$$\hat{c}_{x}(i_{s}(n) \dots i_{e}(n)) = \sum_{k \neq j} \log (|X(k,n)|) e^{iki\Delta\Omega}$$

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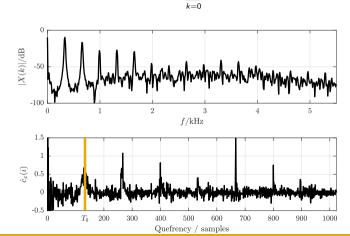
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$$\hat{c}_{x}(i_{s}(n)\dots i_{e}(n)) = \sum_{k=1}^{K/2-1} \log(|X(k,n)|) e^{jki\Delta\Omega}$$



■ summary:

- cepstrum 'replaces' time domain convolution operation with addition
- result is the *unfiltered* excitation signal *plus* the filter IR (both logarithmic)
- can be used for, e.g., spectral envelope extraction or pitch detection
- more naming silliness: cepstrum, quefrency, liftering, . . .

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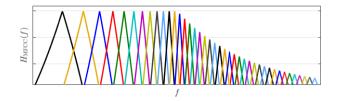
spectral shape features mel frequency cepstral coefficients 1/4

- typical processing steps for the mel frequency cepstral coefficients (MFCCs):
 - 1 compute magnitude spectrum
 - 2 convert linear frequency scale to logarithmic
 - 3 group bins into bands
 - 4 apply logarithm to all bands
 - 5 compute (inverse) cosine transform (DCT)

$$v_{\mathrm{MFCC}}^{j}(n) = \sum_{k'=1}^{\mathcal{K}'} \log\left(|X'(k',n)|\right) \cdot \cos\left(j \cdot \left(k' - \frac{1}{2}\right) \frac{\pi}{\mathcal{K}'}\right)$$

spectral shape features mel frequency cepstral coefficients 2/4

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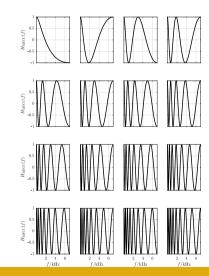
- constant Q filter spacing for higher frequencies (mel scale)
- FFT values are weighted and summed over bins for each band

verview intro timbre spectral features MFCCs tonalness technical learned summary

spectral shape features mel frequency cepstral coefficients 3/4

mel-warped cosine bases for DCT

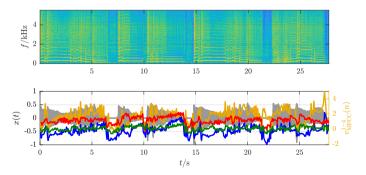




spectral shape features mel frequency cepstral coefficients 4/4

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Property	DM	НТК	SAT
Num. filters	20	24	40
Mel scale	lin/log	log	lin/log
Freq. range	[100; 4000]	[100; 4000]	[200; 6400]
Normalization	Equal height	Equal height	Equal area

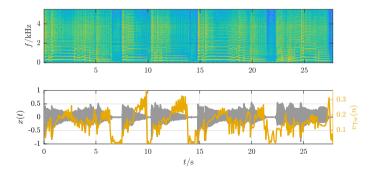


tonalness features spectral crest factor

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$$v_{\mathrm{Tsc}}(n) = rac{\displaystyle\max_{0 \leq k \leq \mathcal{K}/2-1} |X(k,n)|}{\sum\limits_{k=0}^{\mathcal{K}/2-1} |X(k,n)|}$$







tonalness features spectral crest factor



$$v_{\mathrm{Tsc}}(n) = \frac{\max\limits_{0 \leq k \leq K/2-1} |X(k,n)|}{\sum\limits_{k=0}^{K/2-1} |X(k,n)|}$$

common variants:

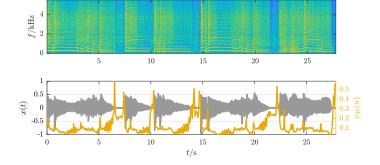
- normalization
- power spectrum
- measure *per band* instead of whole spectrum

tonalness features spectral flatness

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$$v_{\mathrm{Tf}}(n) = \frac{\sqrt[K/2-1]{\prod\limits_{k=0}^{K/2-1}|X(k,n)|}}{\frac{2}{K} \cdot \sum\limits_{k=0}^{K/2-1}|X(k,n)|}$$







tonalness features spectral flatness



$$v_{\mathrm{Tf}}(n) = \frac{\sqrt[\kappa/2]{\prod_{k=0}^{\kappa/2-1}|X(k,n)|}}{\sqrt[2]{\kappa} \cdot \sum_{k=0}^{\kappa/2-1}|X(k,n)|}$$

common variants:

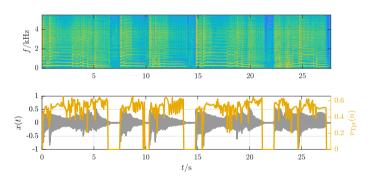
- power vs. magnitude spectrum
- smoothed spectrum (avoid spurious 0-bins)
- measure *per band* instead of whole spectrum

tonalness features spectral tonal power ratio

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$$u_{\mathrm{Tpr}} = \frac{E_{\mathrm{T}}(n)}{\sum\limits_{i=0}^{K/2-1} |X(k,n)|^2}$$







tonalness features spectral tonal power ratio

$$v_{\mathrm{Tpr}} = \frac{E_{\mathrm{T}}(n)}{\sum\limits_{i=0}^{\kappa/2-1} |X(k,n)|^2}$$

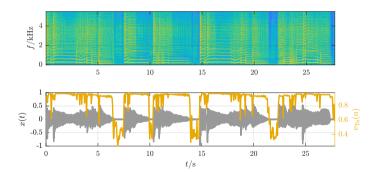
common variants:

definition of tonal/non-tonal components

tonalness features maximum of ACF

$$v_{\mathrm{Ta}}(n) = \max_{\eta_1 \leq \eta \leq \mathcal{K} - 1} |r_{\mathsf{xx}}(\eta, n)|$$







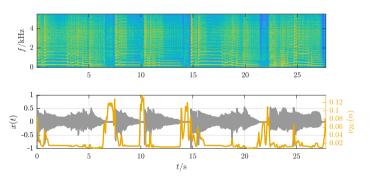
Tech 🛭 Technology

technical features

zero crossing rate

$$v_{ ext{ZC}}(n) = rac{1}{2 \cdot \mathcal{K}} \sum_{i=i_{ ext{s}}(n)}^{i_{ ext{e}}(n)} ig| \operatorname{sign}\left[x(i)
ight] - \operatorname{sign}\left[x(i-1)
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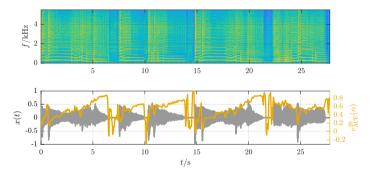




technical features ACF coefficients

$$v_{\text{ACE}}^{\eta}(n) = r_{xx}(\eta, n)$$
 with $\eta = 1, 2, 3, \dots$







feature learning introduction

hand-crafted features:

- arbitrary definitions
- simple to compute
- mostly focus on one technical property
- provide limited information

■ feature learning:

- automatically learn features from data-set
- meaning not obvious, can combine multiple properties

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feature learning

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principle

- 1 put (a lot of) raw data at input
- 2 learn a way of reducing dimensionality while keeping as much information as possible

advantages

- features might contain more useful information than provided by hand-crafted features
- no expert knowledge required

disadvantages

- usually time consuming
- limited ways of controlling the type of information learned

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■ dictionary learning (sparse coding, non-negative matrix factorization)

$$X = B \cdot A$$

X: input signal to be modeled (often spectrogram)

B: dictionary/template matrix (often set of single spectra that comprise the basic building blocks of X)

A: activation matrix indicating the weight and superposition of templates

- derive B,A, by minimizing a cost function, e.g. $||X BA||_2$
- → templates are trained, activations are used as feature vector (length: number of templates)

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■ clustering

- find clusters in data set (e.g., from magnitude spectra or simple features)
- store median of clusters (compare: template matrix)
- \rightarrow features
 - binary vector (length: number of clusters, zero except for closest cluster)
 - distance vector (distance to each cluster)
- neural networks and deep architectures
 - stack multiple layers of simple learning blocks
 - each layer uses the output of the previous layer as input
 - → feature: output of the highest layer

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summary lecture content

■ feature

- descriptor with condensed relevant information
- not necessarily interpretable by humans

■ low-level feature extraction

- usually extracted per short block of samples
- many features can be extracted from audio data, resulting in feature matrix

■ timbre

- mostly dependent on both spectral shape and time domain envelope characteristics
- multi-dimensional perceptual property not as clearly defined as pitch or loudness

■ instantaneous spectral shape features

- established set of baseline features
- usually extracted from the magnitude spectrum
- condensing various properties of the spectral shape into single values
- there exist multiple variants of "the same" feature



Module 3.6: Instantaneous Features