

Introduction to Audio Content Analysis

module 3.5: instantaneous features

alexander lerch



overview intro timbre spectral features MFCCs tonalness technical summar

introduction overview



corresponding textbook section

section 3.5

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



module 3.5; instantaneous features 1/26

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introduction overview



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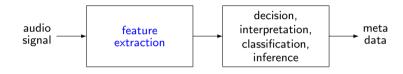


module 3.5: instantaneous features 1 / 26

instantaneous features introduction



remember the flow chart of a general ACA system:



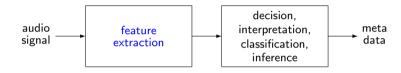
teature

- 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



remember the flow chart of a general ACA system:



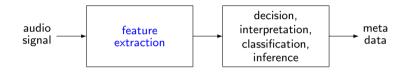
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instantaneous features introduction



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instantaneous features feature



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

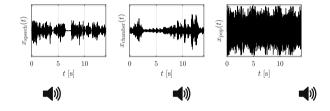


module 3.5: instantaneous features

instantaneous features feature example



waveform envelope of three different signals

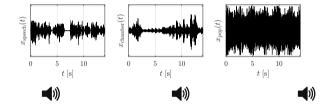


- envelopes of waveforms can have distinct shape
- \Rightarrow a feature describing envelope shape could help to distinguish these signal types



instantaneous features feature example

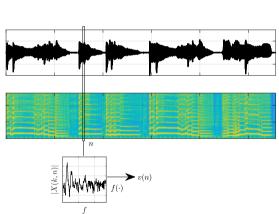
waveform envelope of three different signals



- envelopes of waveforms can have distinct shape
- \Rightarrow a feature describing envelope shape could help to distinguish these signal types







- 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?



module 3.5: instantaneous features 6 / 26

A. 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
- \rightarrow [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 is

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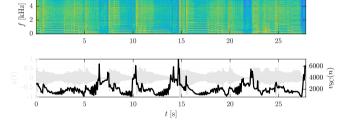
when dealing with complex mixtures of sound, it is very difficult (maybe impossible?) to extract detailed temporal information for individual tones

spectral shape features spectral centroid



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







spectral shape features spectral centroid



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

common variants:

- power spectrum
- logarithmic frequency scale

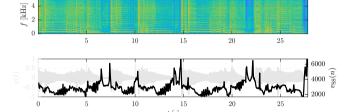
$$v_{\text{SC,log}}(n) = \frac{\sum\limits_{k=k(f_{\min})}^{\mathcal{K}/2-1} \log_2\left(\frac{f(k)}{f_{\text{ref}}}\right) \cdot |X(k,n)|^2}{\sum\limits_{k=k(f_{\min})}^{N/2-1} |X(k,n)|^2}$$

spectral shape features spectral spread

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







spectral shape features spectral spread



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

common variants:

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

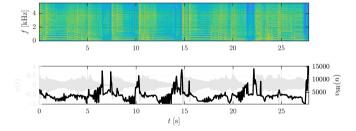
$$v_{\rm SS,log}(n) = \sqrt{\frac{\sum\limits_{k=k(f_{\rm min})}^{\mathcal{K}/2-1} \left(\log_2\left(\frac{f(k)}{1000\,{\rm Hz}}\right) - v_{\rm SC}(n)\right)^2 \cdot |X(k,n)|^2}{\sum\limits_{k=k(f_{\rm min})}^{\mathcal{K}/2-1} |X(k,n)|^2}}$$

spectral shape features spectral rolloff

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$$v_{\mathrm{SR}}(n) = k_{\mathrm{r}} \left| \sum_{\substack{k_{\mathrm{r}} \\ k=0}}^{k_{\mathrm{r}}} |X(k,n)| = \kappa \cdot \sum_{k=0}^{\mathcal{K}/2} |X(k,n)| \right|$$







spectral shape features spectral rolloff



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common variants:

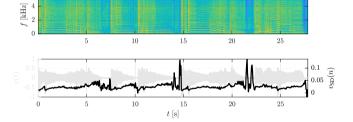
- scaled to frequency
- power spectrum

spectral shape features spectral decrease



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







spectral shape features spectral decrease



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

common variants:

restricted frequency range:

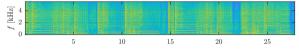
$$v_{ ext{SD}}(n) = rac{\sum\limits_{k=k_{ ext{l}}}^{k_{ ext{u}}} rac{1}{k} \cdot ig(|X(k,n)| - |X(k_{ ext{l}}-1,n)| ig)}{\sum\limits_{k=k_{ ext{l}}}^{k_{ ext{u}}} |X(k,n)|}$$

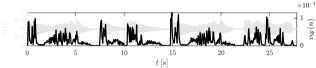
spectral shape features spectral flux

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$$v_{\mathrm{SF}}(n) = \frac{\sqrt{\sum\limits_{k=0}^{\mathcal{K}/2} \left(|X(k,n)| - |X(k,n-1)|\right)^2}}{\frac{\mathcal{K}/2 + 1}{}}$$









spectral shape features spectral flux



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

common variants:

$$v_{\rm SF}(n,\beta) = \frac{\sqrt[\beta]{\sum_{k=0}^{K/2-1} (|X(k,n)| - |X(k,n-1)|)^{\beta}}}{\frac{K/2}{K}}$$

$$v_{\rm SF,\sigma}(n) = \sqrt{\frac{2}{K} \sum_{k=0}^{K/2-1} (\Delta X(k,n) - \mu_{\Delta X})^{2}}$$

$$v_{\rm SF,log}(n) = \frac{2}{K} \sum_{k=0}^{K/2-1} \log_{2} \left(\frac{|X(k,n)|}{|X(k,n-1)|}\right)$$



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



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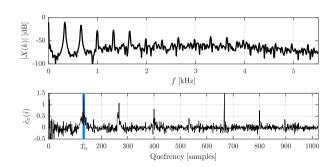
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■ summary:

- 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, . . .



summary:

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spectral shape features mel frequency cepstral coefficients 1/4

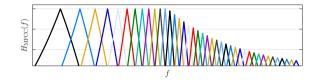


- typical processing steps for the mel frequency cepstral coefficients (MFCCs):
 - 1 compute magnitude spectrum
 - 2 convert linear frequency scale to logarithmic
 - 3 group bins into bands
 - 4 apply logarithm to all bands
 - **5** compute (inverse) cosine transform (DCT)

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

MFCCs

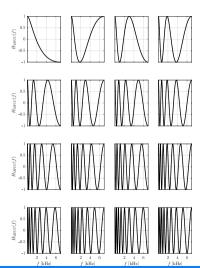
spectral shape features mel frequency cepstral coefficients 2/4



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

spectral shape features mel frequency cepstral coefficients 3/4

mel-warped cosine bases for DCT

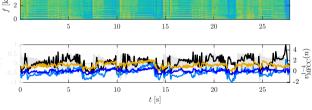


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spectral shape features mel frequency cepstral coefficients 4/4

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Property	DM	HTK	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
		l kalin keka	

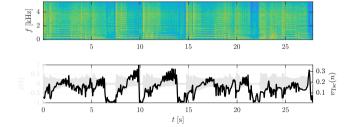


tonalness features spectral crest factor

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







tonalness features spectral crest factor



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

common variants:

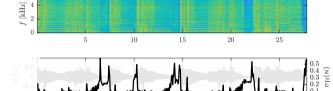
- normalization
- power spectrum
- measure per band instead of whole spectrum

tonalness features spectral flatness

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







tonalness features spectral flatness



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

common variants:

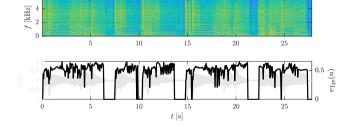
- power vs. magnitude spectrum
- smoothed spectrum (avoid spurious 0-bins)
- measure *per band* instead of whole spectrum

tonalness features spectral tonal power ratio



$$v_{\mathrm{Tpr}} = rac{E_{\mathrm{T}}(n)}{\sum\limits_{i=0}^{\mathcal{K}/2} |X(k,n)|^2}$$







tonalness features spectral tonal power ratio



$$v_{\mathrm{Tpr}} = rac{E_{\mathrm{T}}(n)}{\sum\limits_{i=0}^{\mathcal{K}/2} |X(k,n)|^2}$$

common variants:

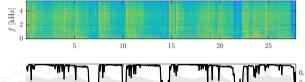
- definition of tonal/non-tonal components
 - local maxima
 - peak salience
 - in periodic (harmonic) pattern
 - .

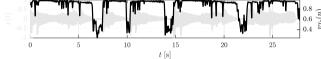
tonalness features maximum of ACF

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$$v_{\mathrm{Ta}}(n) = \max_{0 \leq \eta \leq \mathcal{K}-1} |r_{\mathsf{xx}}(\eta, n)|$$









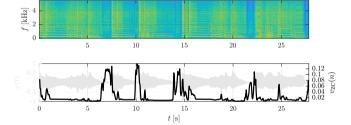
technical features

zero crossing rate

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





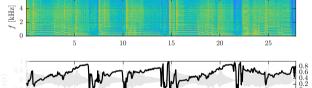


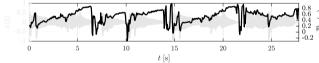
technical features ACF coefficients

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







erview intro timbre spectral features MFCCs tonalness technical summary 0000 00 0000 000 000

summary lecture content



feature

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

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
- often extracted from the magnitude spectrum, describing timbre
- condensing various properties of the spectral shape into single values
- there exist multiple variants of "the same" feature



module 3.5; instantaneous features 26 / 26