Introduction to Audio Content Analysis

Module 3.4.2: Time-Frequency Representations — Constant Q Transform

alexander lerch



corresponding textbook section

Section 3.4.2

- lecture content
 - constant-Q transform (CQT)
- learning objectives
 - discussing advantages and disadvantages of different time-frequency transforms
 - explaining the principles of the CQT and auditory filterbanks



corresponding textbook section

Section 3.4.2

- lecture content
 - constant-Q transform (CQT)
- **■** learning objectives
 - discussing advantages and disadvantages of different time-frequency transforms
 - explaining the principles of the CQT and auditory filterbanks



- Fourier transform continues to be much-used tool in audio signal processing and MIR
- but there are disadvantages, e.g.
 - frequency axis does not directly map to (perceptual) pitch axis
 - frequency and time resolution inversely related
 - ⇒ **alternative transforms** can be used

- Fourier transform continues to be much-used tool in audio signal processing and MIR
- but there are disadvantages, e.g.
 - frequency axis does not directly map to (perceptual) pitch axis
 - frequency and time resolution inversely related
 - ⇒ alternative transforms can be used

•000

- DFT has a *linear* frequency axis:
 - not perceptually meaningful: *logarithmic* is better match
 - low pitch resolution at low frequencies
- - space frequencies logarithmically (constant Q)
 - resulting abscissa resolution is pitch-related
 - parameter c adjusts number of bins per octave

- DFT has a *linear* frequency axis:
 - not perceptually meaningful: logarithmic is better match
 - low pitch resolution at low frequencies
- ⇒ compute DFT-like transform at specific frequencies
 - space frequencies logarithmically (constant Q)
 - resulting abscissa resolution is pitch-related
 - parameter c adjusts number of bins per octave

$$Q = \frac{f}{\Delta f} = \frac{1}{2^{1/c} - 1}$$

$$X_{\text{CQ}}(k,n) = \frac{1}{\mathcal{K}(k)} \sum_{i=i_{\text{s}}(n)}^{i_{\text{e}}(n)} w_k(i-i_{\text{s}}) \cdot x(i) e^{j2\pi \frac{\mathcal{Q} \cdot (i-i_{\text{s}})}{\mathcal{K}(k)}} = \mathcal{K}(k)$$
: frequency of bin index k $\mathcal{K}(k)$: blocklength for bin index k \mathcal{Q} : measure of pitch res. $\mathcal{K}(k) = \frac{f_{\text{S}}}{f(k)} \mathcal{Q}$ \mathcal{W}_k : window function

 \mathbf{w}_k : window function

 \bullet i_s , i_e : start and stop time indices of block

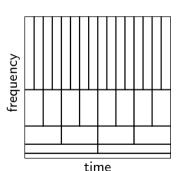
 \blacksquare f_S : sample rate

- long window for low frequencies (high freq res, low time res)
- short window for high frequencies (low freq res, high time res)

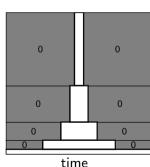
0000

constant Q transform implementation 2/2

non-overlapping



overlapping



differences

- outputs at multiple vs. one time resolution
- multiple different FFT lengths vs. one FFT length (zero-padded)
- dependent vs. independent definition of block and hop length

- + perceptually/musically adapted frequency resolution
- time resolution depends on frequency
- not invertible
- no optimized implementation (compare FFT)

- + perceptually/musically adapted frequency resolution
- time resolution depends on frequency
- not invertible
- no optimized implementation (compare FFT)

- + perceptually/musically adapted frequency resolution
- time resolution depends on frequency
- not invertible
- no optimized implementation (compare FFT)

0000

- + perceptually/musically adapted frequency resolution
- time resolution depends on frequency
- not invertible
- no optimized implementation (compare FFT)

■ DFT has disadvantages

- low frequency resolution for low pitches
- non-logarithmic/perceptually relevant pitch resolution

CQT

- similar to Fourier Transform but logarithmically spaced frequency bins
- not invertible and inefficient

