



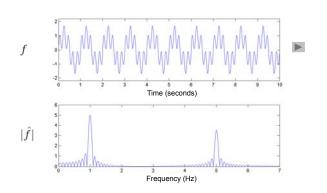
Lecture **Music Processing** 

#### **Audio Features**

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# **Fourier Transform**



# **Fourier Transform**

Signal  $f:\mathbb{R}\to\mathbb{R}$ 

Fourier representation  $f(t)=\int\limits_{\omega\in\mathbb{R}}c_{\omega}e^{2\pi i\omega t}d\omega$ ,  $c_{\omega}=\hat{f}(\omega)$ Fourier transform  $\hat{f}(\omega)=\int\limits_{t\in\mathbb{R}}f(t)e^{-2\pi i\omega t}dt$ 

# **Fourier Transform**

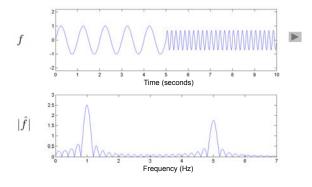
Signal  $f: \mathbb{R} \to \mathbb{R}$ 

Fourier representation  $f(t)=\int\limits_{\omega\in\mathbb{R}}c_{\omega}e^{2\pi i\omega t}d\omega$  ,  $c_{\omega}=\hat{f}(\omega)$ 

 $\hat{f}(\omega) = \int_{t \in \mathbb{R}} f(t)e^{-2\pi i\omega t} dt$ Fourier transform

- Tells which notes (frequencies) are played, but does not tell when the notes are played
- Frequency information is averaged over the entire time interval
- Time information is hidden in the phase

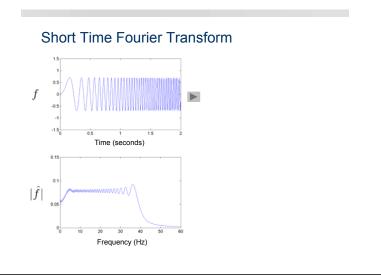
# **Fourier Transform**

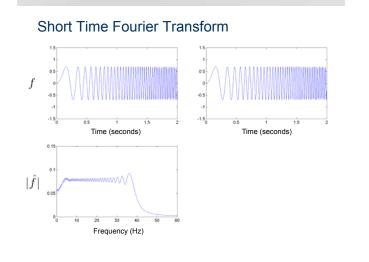


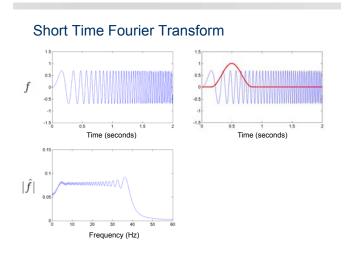
#### **Short Time Fourier Transform**

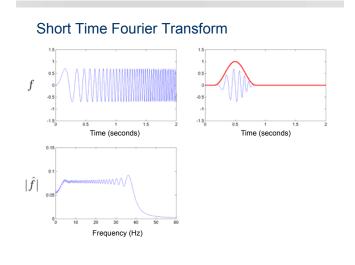
Idea (Dennis Gabor, 1946):

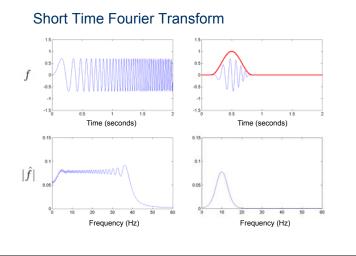
- Consider only a small section of the signal for the spectral analysis
  - → recovery of time information
- Short Time Fourier Transform (STFT)
- Section is determined by pointwise multiplication of the signal with a localizing window function

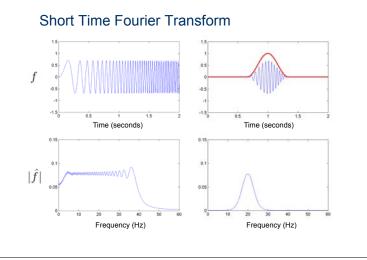




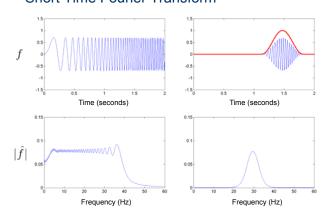








# **Short Time Fourier Transform**



# **Short Time Fourier Transform**

#### Definition

- Signal  $f: \mathbb{R} \to \mathbb{R}$
- Window function  $g:\mathbb{R}\to\mathbb{R}$   $(g\in L^2(\mathbb{R}),\|g\|=1)$
- STFT  $\tilde{f}(\omega, t) := \int_{\mathbb{R}} f(u)\bar{g}(u t)e^{-2\pi i\omega u}du = \langle f|g_{\omega,t}\rangle$

with 
$$g_{\omega,t}(u) := e^{2\pi i \omega u} g(u-t), \quad u \in \mathbb{R}$$

# **Short Time Fourier Transform**

#### Intuition:

•  $g_{\omega,t}$  is "musical note" of frequency  $\omega$ , which oscillates within the translated window  $u \to g(u-t)$ 





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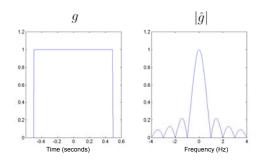




Innere product  $\ \langle f|g_{\omega,t} \rangle$  measures the correlation between the musical note  $\ g_{\omega,t}$  and the signal  $\ f.$ 

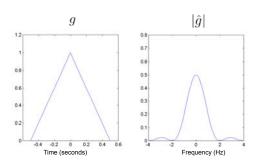
# Window Function

#### Box window



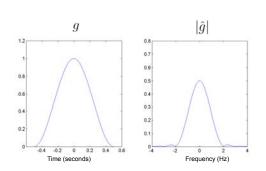
# Window Function

#### Triangle window

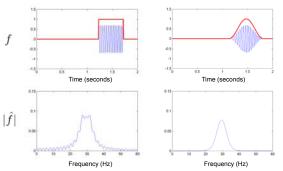


# Window Function

Hann window

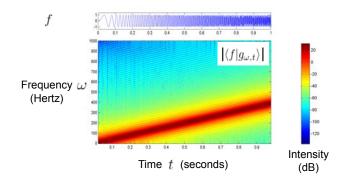


# Window Function

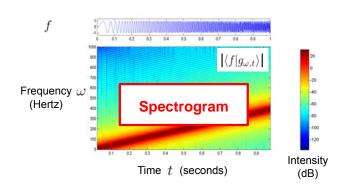


Trade off between smoothing and "ringing"

# Time-Frequency Representation

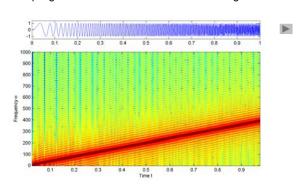


# Time-Frequency Representation



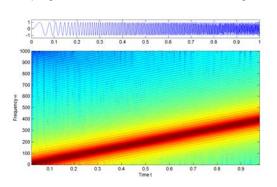
# Time-Frequency Representation

Chirp signal and STFT with box window of length 0.05



# Time-Frequency Representation

Chirp signal and STFT with Hann window of length 0.05



# **Time-Frequency Localization**

 Size of window constitutes a trade-off between time resolution and frequency resolution:

Large window: poor time resolution

good frequency resolution

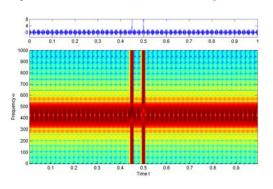
Small window: good time resolution

poor frequency resolution

 Heisenberg Uncertainty Principle: there is no window function that localizes in time and frequency with arbitrary position.

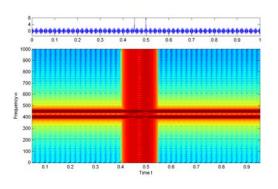
# **Short Time Fourier Transform**

Signal and STFT with Hann window of length 0.02



# **Short Time Fourier Transform**

Signal and STFT with Hann window of length 0.1



# Heisenberg Uncertainty Principle

Window function  $\ g \in L^2(\mathbb{R}) \ \text{with} \quad \|g\| = 1$ 

Center

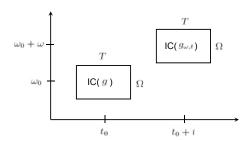
\/\idth

$$\begin{split} t_0 &= t_0(g) := \int_{-\infty}^{\infty} t |g(t)|^2 dt & T(g) := \left( \int_{-\infty}^{\infty} (t-t_0)^2 |g(t)|^2 dt \right)^{\frac{1}{2}} \\ \omega_0 &= \omega_0(g) := \int_{-\infty}^{\infty} \omega |\hat{g}(\omega)|^2 d\omega & \Omega(g) := \left( \int_{-\infty}^{\infty} (\omega-\omega_0)^2 |\hat{g}(\omega)|^2 d\omega \right)^{\frac{1}{2}} \end{split}$$

$$T(g)\cdot\Omega(g)\geq\frac{1}{4\pi}$$

# Information Cells

$$g_{\omega,t}(u) := e^{2\pi i \omega u} g(u-t)$$
 "musical note"



# **MATLAB**

- MATLAB function SPECTROGRAM
- N = window length (in samples)
- M = overlap (usually N/2)
- Compute DFT<sub>N</sub> for every windowed section
- Keep lower N/2 Fourier coefficients
- ightarrow Sequence of spectral vectors (for each window a vector of dimension N/2 )

# Example

Let x be a discrete time signal x(n) = f(Tn)

 $\begin{array}{ll} \mbox{Sampling rate:} & 1/T = 22050 \ \mbox{Hz} \\ \mbox{Window length:} & N = 4096 \\ \mbox{Overlap:} & N/2 = 2048 \end{array}$ 

Hopsize: window length – overlap

Let  $v_0 := (x(0), x(1), \dots, x(4095))$   $v_1 := (x(2048), \dots, x(6143))$  $v_2 := (x(4096), \dots, x(8191))$ 

 $v_m$  corresponds to window  $\left[m \cdot 2048 : m \cdot 2048 + 4095\right]$ 

# Example

Time resolution:

$$\frac{\text{hopsize}}{\text{sampling rate}} = \frac{4096 - 2048}{22050} = 0.093 = 93 \ ms$$

Frequency resolution:

$$v = v_0$$
,  $\hat{v} := DFT_N(v)$ 

$$\hat{v}(k) \approx \frac{1}{T} \cdot \hat{f}\left(\frac{k}{N} \cdot \frac{1}{T}\right)$$

$$\omega = \frac{k}{N} \cdot \frac{1}{T} = k \cdot \frac{22050}{4096} = k \cdot 5.38 \ \text{Hz}$$

#### Pitch Features

Model assumption: Equal-tempered scale

• MIDI pitches:  $p \in [1:128]$ 

• Piano notes: p = 21 (A0) to p = 108 (C8)

• Concert pitch: p = 69 (A4)

• Center frequency:  $f_{\mathrm{MIDI}}(p) = 2^{\frac{p-69}{12}} \cdot 440 \;\; \mathrm{Hz}$ 

→ Logarithmic frequency distribution Octave: doubling of frequency

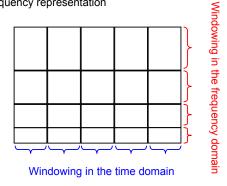
# Pitch Features

Idea: Binning of Fourier coefficients

Divide up the fequency axis into logarithmically spaced "pitch regions" and combine spectral coefficients of each region to a single pitch coefficient.

# Pitch Features

Time-frequency representation



# Pitch Features

#### Details:

Let  $\hat{v}$  be a spectral vector obtained from a spectrogram w.r.t. a sampling rate 1/T and a window length N. The spectral coefficient  $\hat{v}(k)$  corresponds to the frequency

$$f_{\text{coeff}}(k) := \frac{k}{N} \cdot \frac{1}{T}$$

Let

 $S(p) := \{k: f_{\mathrm{MIDI}}(p-0.5) \leq f_{\mathrm{coeff}}(k) < f_{\mathrm{MIDI}}(p+0.5)\}$  be the set of coefficients assigned to a pitch  $p \in [1:128]$  Then the pitch coefficient P(p) is defined as

$$P(p) := \sum_{k \in S(p)} |\hat{v}(k)|^2$$

# Pitch Features

Example: A4, p = 69

• Center frequency:  $f(p=69) = 2^{\frac{0}{12}} \cdot 440 = 440 \ Hz$ 

• Lower bound:  $f(p = 68.5) = 2^{\frac{-0.5}{12}} \cdot 440 = 427.5 \ Hz$ 

• Upper bound:  $f(p = 69.5) = 2^{\frac{0.5}{12}} \cdot 440 = 452.9 \; Hz$ 

• STFT with N = 4096 , 1/T = 22050

# Pitch Features

Example: A4, p = 69

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• STFT with N = 4096 , 1/T = 22050

$$\begin{array}{lll} \vdots & \vdots & \vdots \\ f(k=79) & = & 425.3 \; Hz \\ f(k=80) & = & 430.7 \; Hz \\ f(k=81) & = & 436.0 \; Hz \\ f(k=82) & = & 441.4 \; Hz \\ f(k=83) & = & 446.8 \; Hz \\ f(k=84) & = & 452.2 \; Hz \\ f(k=85) & = & 457.6 \; Hz \\ \vdots & & \vdots & & \\ \end{array} \right\} \quad \mathcal{S}(p=69) \\ P(p=69) = \sum_{k=80}^{84} |\hat{v}(k)|^2$$

# Pitch Features

Note	MIDI pitch	Center [Hz] frequency	Left [Hz] boundary	Right [Hz] boundary	Width [Hz]
A3	57	220.0	213.7	226.4	12.7
A#3	58	233.1	226.4	239.9	13.5
B3	59	246.9	239.9	254.2	14.3
C4	60	261.6	254.2	269.3	15.1
C#4	61	277.2	269.3	285.3	16.0
D4	62	293.7	285.3	302.3	17.0
D#4	63	311.1	302.3	320.2	18.0
E4	64	329.6	320.2	339.3	19.0
F4	65	349.2	339.3	359.5	20.2
F#4	66	370.0	359.5	380.8	21.4
G4	67	392.0	380.8	403.5	22.6
G#4	68	415.3	403.5	427.5	24.0
A4	69	440.0	427.5	452.9	25.4

# Pitch Features

#### Note:

- $P \in \mathbb{R}^{128}$
- For some pitches, S(p) may be empty. This particularly holds for low notes corresponding to narrow frequency bands.
- $\rightarrow$  Linear frequency sampling is problematic!

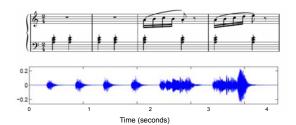
#### Solution:

Multi-resolution spectrograms or multirate filterbanks

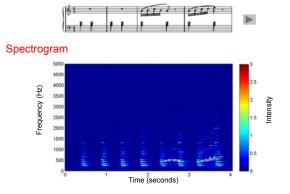
# Pitch Features

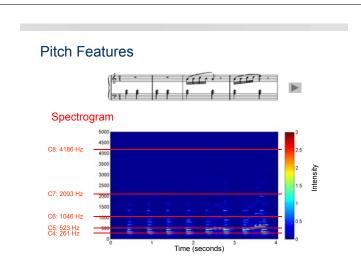
Example: Friedrich Burgmüller, Op. 100, No. 2

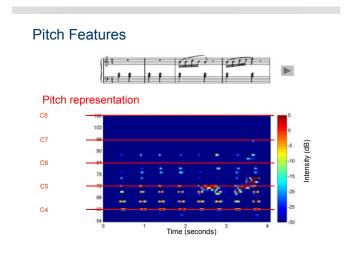


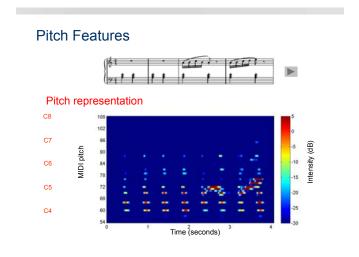


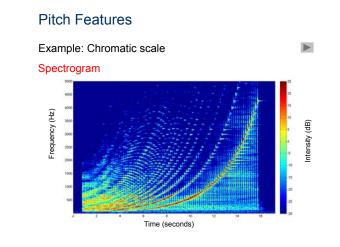
# Pitch Features

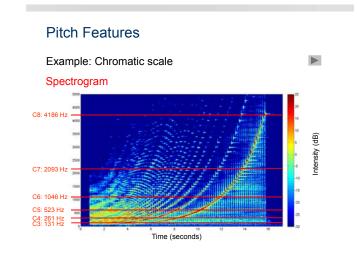


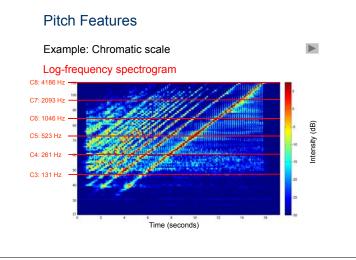


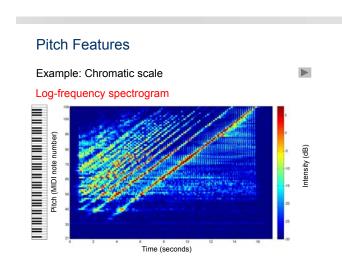


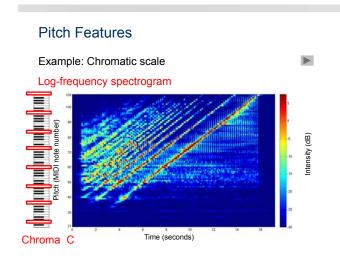


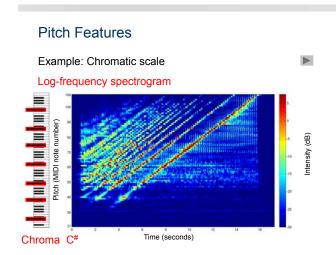


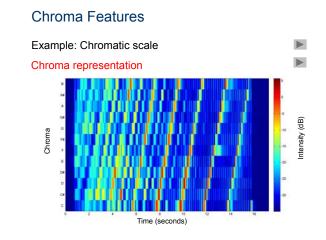


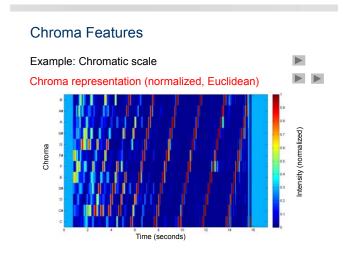












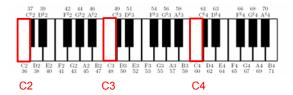
# **Chroma Features**

- Human perception of pitch is periodic in the sense that two pitches are perceived as similar in color if they differ by an octave.
- Seperation of pitch into two components: tone height (octave number) and chroma.
- Chroma: 12 traditional pitch classes of the equaltempered scale. For example:
  - Chroma C  $\widehat{=}$  {..., C0, C1, C2, C3, ...}
- Computation: pitch features → chroma features
   Add up all pitches belonging to the same class
- Result: 12-dimensional chroma vector.

# **Chroma Features**

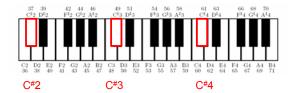


# **Chroma Features**



Chroma C

# **Chroma Features**



Chroma C#

# **Chroma Features**

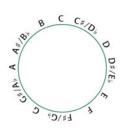


Chroma D

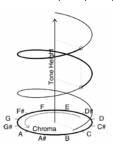
# **Chroma Features**

Chromatic circle

Shepard's helix of pitch perception

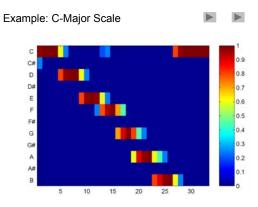


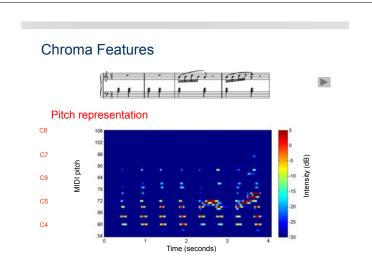
http://en.wikipedia.org/wiki/Pitch\_class\_space

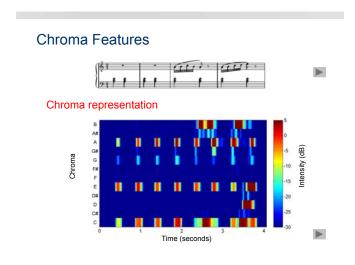


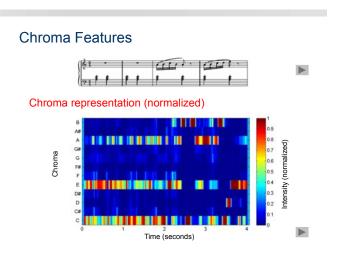
Bartsch/Wakefield, IEEE Trans. Multimedia, 2005

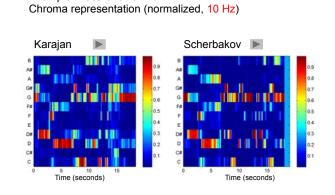
# **Chroma Features**





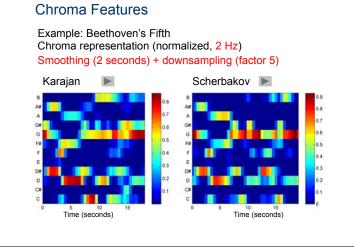


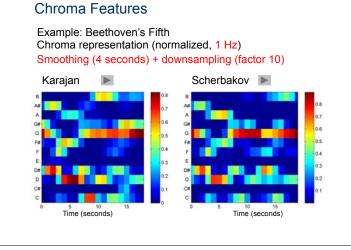




Chroma Features

Example: Beethoven's Fifth

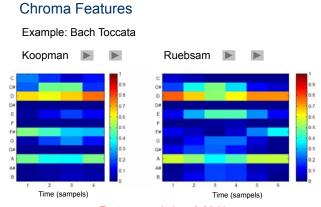




# Chroma Features Example: Bach Toccata Koopman Ruebsam Ruebsam Time (sampels)

# Chroma Features Example: Bach Toccata Koopman Ruebsam Ruebsam Feature resolution: 10 Hz

# Chroma Features Example: Bach Toccata Koopman Ruebsam Ruebsam Companies Ruebsam Ruebsa



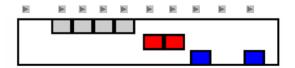
# Feature resolution: 0.33 Hz

# **Chroma Features**

- Sequence of chroma vectors correlates to the harmonic progression
- Normalization  $v \to \frac{v}{\|v\|}$  makes features invariant to changes in dynamics
- Further quantization and smoothing: CENS features
- Taking logarithm before adding up pitch coefficients accounts for logarithmic sensation of intensity

# **Chroma Features**

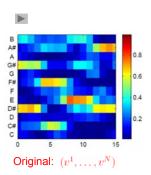
Example: Zager & Evans "In The Year 2525"



How to deal with transpositions?

# **Chroma Features**

Example: Zager & Evans "In The Year 2525"



# **Audio Features**

- There are many ways to implement chroma features
- Properties may differ significantly
- Appropriateness depends on respective application



- http://www.mpi-inf.mpg.de/resources/MIR/chromatoolbox/
- MATLAB implementations for various chroma variants
- ISMIR 2011, Poster Session (PS2), Tuesday 13-15

# **Chroma Features**

Example: Zager & Evans "In The Year 2525"

