



# Introduction to **Audio Content Analysis**

module B.6: frequency resolution & instantaneous frequency

alexander lerch

# introduction

## overview

### corresponding textbook section

- section 7.3.1
- appendix B.6

### ■ lecture content

- frequency detection error for sampled signals
- instantaneous frequency/frequency reassignment

### ■ learning objectives

- list the factors influencing frequency resolution in time and frequency domains
- explain the frequency error in Cent
- implement an instantaneous frequency estimate



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# pitch detection resolution

## introduction

- (fundamental) frequency detection on digital signals (discrete in time and frequency)

⇒ quantized result

error being made due to *discrete* signal processing

- **time domain:**

- detection of *period length*

⇒ maximum error depends on distance between two samples (sample rate)

- **frequency domain:**

- detection of *bin frequency*

⇒ maximum error depends on distance between two frequency bins (block length and sample rate)

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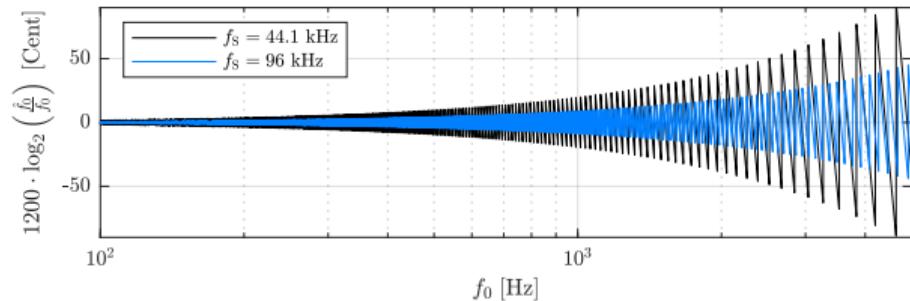
# pitch detection resolution

time domain (e.g., ACF)

period length quantized to multiple of inter-sample interval  $T_S$

$$T_Q = j \cdot T_S$$

$$\Rightarrow f_Q = \frac{1}{j \cdot T_S}$$



# pitch detection resolution

## frequency domain (e.g., HPS)

frequency quantized to multiple of inter-bin interval

$$f_Q = k \cdot \frac{f_S}{\mathcal{K}}$$

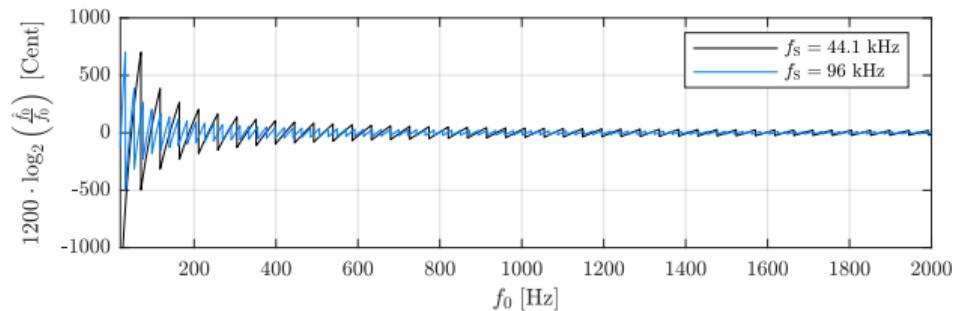
$\mathcal{K}$	$\Delta f$ [Hz]	$k_{ST}$	$f(k_{ST})$ [Hz]
<b>256</b>	187.5	35	6562.5
<b>512</b>	93.75	35	3281.25
<b>1024</b>	46.875	35	1640.625
<b>2048</b>	23.4375	35	820.3125
<b>4096</b>	11.7188	35	410.1563
<b>8192</b>	5.8594	35	205.0781
<b>16384</b>	2.9297	35	102.5391

# pitch detection resolution

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# pitch detection resolution

## simple fix

■ **assumption:** pitch is stationary with minor deviations over time

■ **simple solution:**

- average pitch observations over blocks
- the more blocks are averaged, the more result might approximate the *real* (population) mean

■ **problems:**

- 1 adds significant latency (non-realtime)
- 2 will not work for time-variant signals (speech, music)

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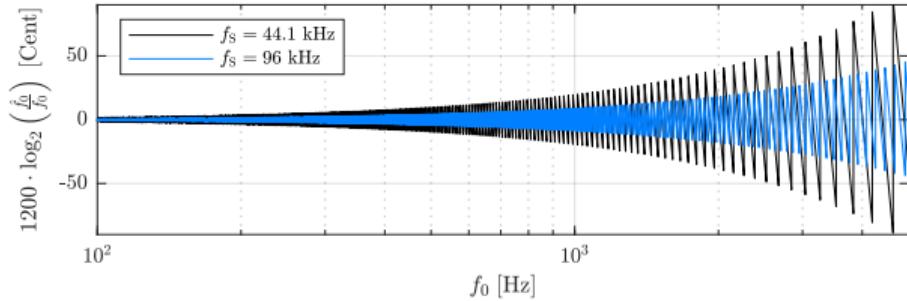
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# pitch detection resolution

## time domain observations

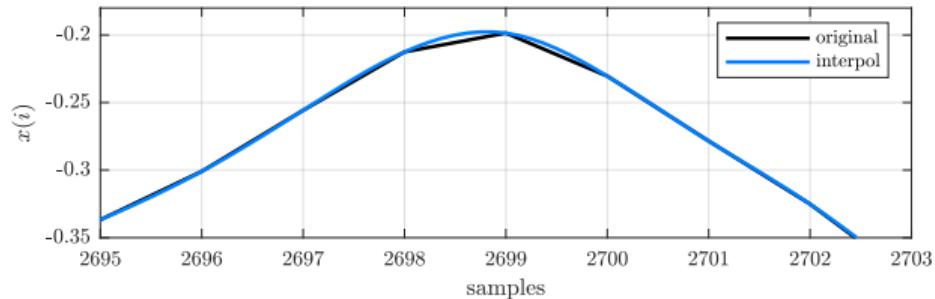


- error depends on fundamental frequency
- error depends on sample rate

# pitch detection resolution

## time domain workarounds

virtually increase time resolution by upsampling



- + higher virtual resolution
- significant workload increase

# pitch detection resolution

## frequency domain workarounds

different ways of increasing frequency resolution in the frequency domain

- 1 increasing the FFT window length (decreases time resolution)
- 2 interpolating the spectrum
- 3 applying frequency reassignment

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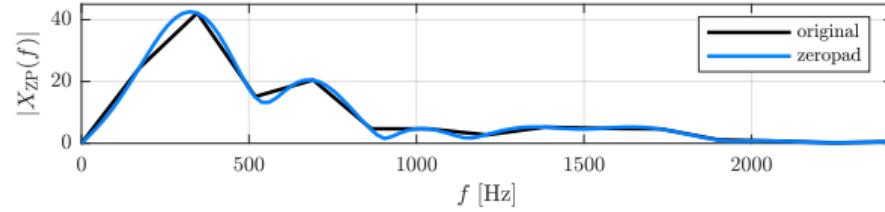
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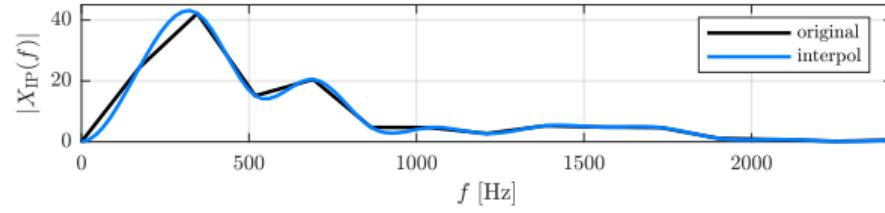
# pitch detection resolution

## spectrum interpolation

1 zeropad in time domain



2 use standard interpolation on magnitude spectrum



# pitch detection resolution

frequency reassignment: relation of phase and frequency 1/2

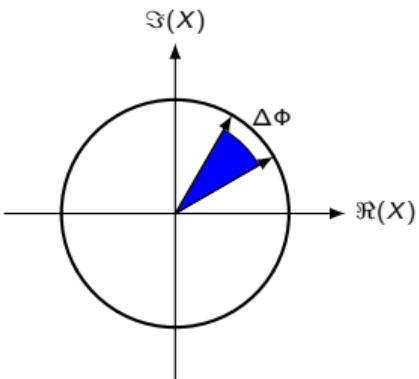
## ■ phasor representation:

- 1 sine value is defined by magnitude and phase
- 2 decreasing the amplitude  $\Rightarrow$  shorter vector
- 3 increasing the frequency  $\Rightarrow$  increasing speed



# pitch detection resolution

frequency reassignment: relation of phase and frequency 2/2



## ■ relation of frequency and phase change:

- time for full rotation is period length  $T$  with

$$f = \frac{1}{T}$$

- time for fractional rotation  $\Delta\Phi$  is corresponding fraction of period length

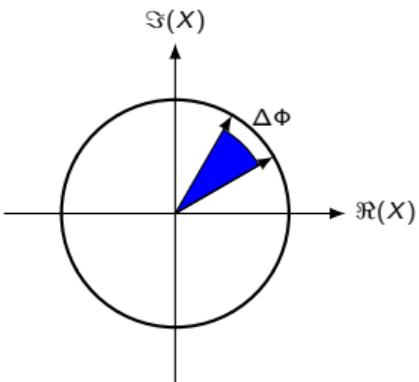
$$f = \frac{\Delta\Phi}{\Delta t}$$

- in other words:

$$\begin{aligned}\Phi(t) &= \omega \cdot t \\ \Rightarrow \frac{d\Phi(t)}{dt} &= \omega = 2\pi f\end{aligned}$$

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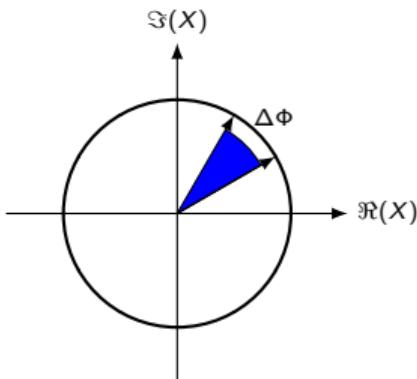
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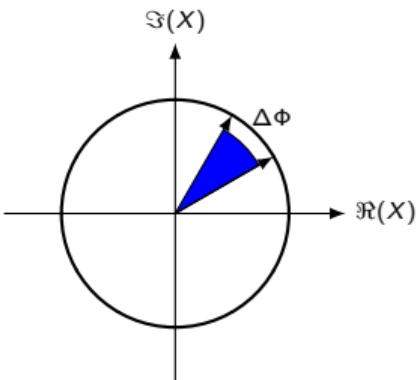
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# pitch detection resolution

## frequency reassignment: principles

frequency domain:

- instead of using the bin frequency

$$f(k) = k \cdot \frac{f_s}{K}$$

- we use the phase of each bin  $\Phi(k, n)$
- to compute the frequency from the phase difference of neighboring blocks

$$\omega_I(k, n) \propto \Phi(k, n) - \Phi(k, n - 1)$$

- $\omega_I(k, n)$  is called **instantaneous frequency** per block per bin

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# pitch detection resolution

frequency reassignment: scaling factor

- instantaneous frequency calculation has to take into account

- hop size  $\mathcal{H}$
- sample rate  $f_S$

$$\omega_I(k, n) = \frac{\Delta\Phi_u(k, n)}{\mathcal{H}} \cdot f_S$$

- problem: phase ambiguity

$$\Phi(k, n) = \Phi(k, n) + j \cdot 2\pi$$

⇒ *phase unwrapping*

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## frequency reassignment: phase unwrapping

### 1 compute unwrapped phase $\Phi_u(k, n)$

- estimate unwrapped bin phase

$$\hat{\Phi}(k, n) = \Phi(k, n - 1) + \underbrace{2\pi k \cdot \frac{\mathcal{H}}{\mathcal{K}}}_{= \omega_k \cdot \frac{\mathcal{H}}{f_s}}$$

- unwrap phase by shifting current phase to estimate's range

$$\Phi_u(k, n) = \hat{\Phi}(k, n) + \text{princarg} [\Phi(k, n) - \hat{\Phi}(k, n)]$$

### 2 compute unwrapped phase difference

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## frequency reassignment: problems

### ■ overlapping spectral components

- sinusoidal components often overlap (spectral leakage, several instruments playing the same pitch, ...)
  - ⇒ “incorrect” phase estimate
  - spectrum should be as sparse as possible, increase STFT length

### ■ inaccurate phase unwrapping

- unwrapping algorithm is based on assumption of similarity between predicted and measured phase
- decrease hop size

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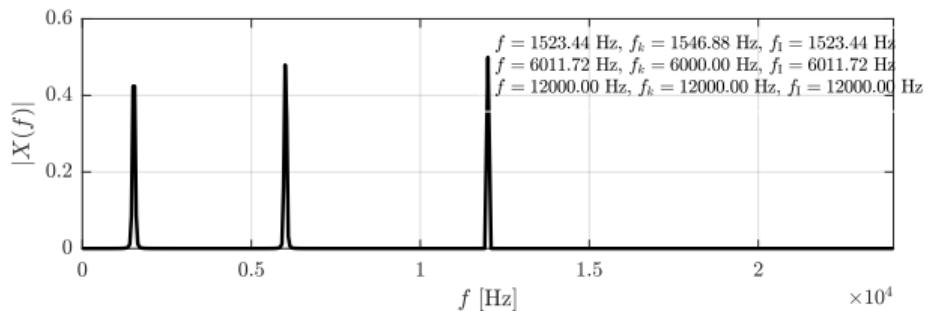
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# pitch detection resolution

## frequency reassignment: example



- FFT length: 1024
- sample rate: 48 kHz

- selected frequencies:
  - between bins (0.5)
  - between bins (0.25)
  - on bin

# pitch detection resolution

## frequency reassignment: applications

### ■ improving frequency resolution

- e.g., for detecting signal frequencies when using a filter bank

### ■ improving phase extrapolation

- e.g., for accurate phase estimation in the *phase vocoder*

### ■ grouping spectral bins

- spectral leakage sidelobes have the same instantaneous frequency

### ■ tonalness detection

- the instantaneous frequency should be reasonably close to the bin frequency for the component to be considered tonal

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# summary

## lecture content

### ■ frequency resolution of sampled signals depends on

- time domain: sample rate
- freq domain: sample rate, block size

### ■ pitch detection error in Cent also depends on input frequency

- time domain: high error at high frequencies
- freq domain: high error at low frequencies

### ■ possible solutions

- time domain:
  - ▶ upsampling/interpolation
- freq domain:
  - ▶ zeropadding/interpolation
  - ▶ frequency reassignment (instantaneous frequency)

