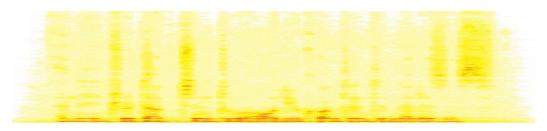
### Introduction to Audio Content Analysis

Module 5.3: Instantaneous Frequency

#### alexander lerch





### introduction

overview



#### corresponding textbook section

- Chapter 2 Fundamentals: pp. 21–23
- Chapter 5 Tonal Analysis: pp. 92–93
- 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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- (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)

## pitch detection resolution introduction



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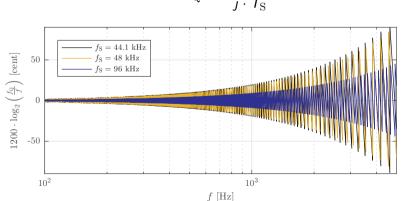
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# pitch detection resolution time domain (e.g., ACF)

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period length quantized to multiple of inter-sample interval  $T_{
m S}$ 

$$egin{array}{ll} {\mathcal T}_{
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m S} \ \Rightarrow & f_{
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# pitch detection resolution frequency domain (e.g., HPS)

frequency quantized to multiple of inter-bin interval

$$f_{\mathrm{Q}} = k \cdot \frac{f_{\mathrm{S}}}{\mathcal{K}}$$

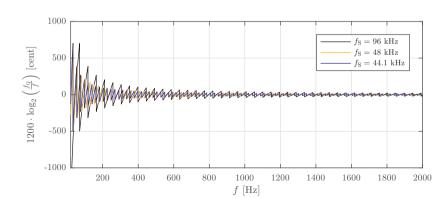
$\mathcal{K}$	$\Delta f$ [Hz]	$k_{ m ST}$	$f(k_{ m ST})$ [Hz]
256	187.5	35	6562.5
512	93.75	35	3281.25
1024	46.875	35	1640.625
2048	23.4375	35	820.3125
4096	11.7188	35	410.1563
8192	5.8594	35	205.0781
16384	2.9297	35	102.5391

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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:
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  - will not work for time-variant signals (speech, music)

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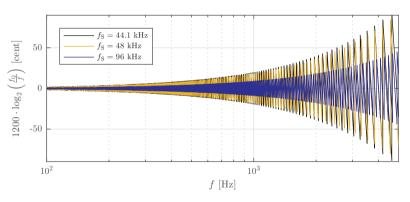
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# pitch detection resolution time domain observations

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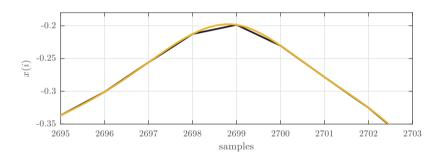


- error depends on fundamental frequency
- error depends on sample rate

### pitch detection resolution time domain workarounds

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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

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

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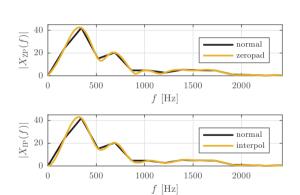
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### pitch detection resolution spectrum interpolation

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zeropad in time domain

use standard interpolation on magnitude spectrum



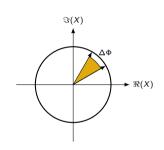


- phasor representation:
  - sine value is defined by magnitude and phase
  - ② decreasing the amplitude ⇒ shorter vector
  - increasing the frequency ⇒ increasing speed



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frequency reassignment: relation of phase and frequency 2/2



- relation of frequency and phase change:
  - time for full rotation is period length T with

$$f = \frac{1}{7}$$

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

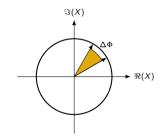
$$f = \frac{\Delta d}{\Delta t}$$

• in other words:

$$\Phi(t) = \omega \cdot t$$

$$\Rightarrow \frac{d\Phi(t)}{dt} = \omega = 2\tau$$

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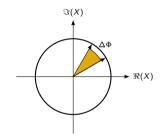
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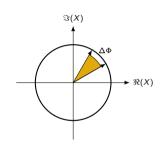
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# pitch detection resolution frequency reassignment: principles

frequency domain:

instead of using the bin frequency

$$f(k) = k * \frac{f_{\rm S}}{\mathcal{K}}$$

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

$$\omega_{\rm I}(k,n) \propto \Phi(k,n) - \Phi(k,n-1)$$

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frequency reassignment: scaling factor

- instantaneous frequency calculation has to take into account
  - ullet hop size  ${\cal H}$
  - ullet sample rate  $f_{
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$$\omega_{\mathrm{I}}(k,n) = \frac{\Delta\Phi_{\mathrm{u}}(k,n)}{\mathcal{H}} \cdot f_{\mathrm{S}}$$

problem: phase ambiguity

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

⇒ phase unwrapping

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frequency reassignment: phase unwrapping

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

unwrap phase by shifting current phase to estimate's range

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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



frequency reassignment: problems

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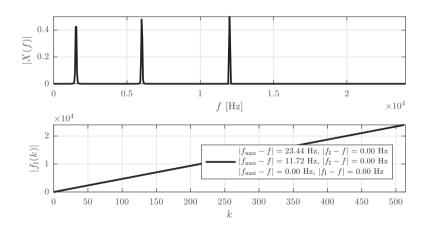
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frequency reassignment: example





## pitch detection resolution frequency reassignment: applications

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- 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)

