

# Digital Signal Processing for Music

Part 23: Time-stretching and Pitch-shifting

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## Intro

### »» Time Stretching

Change playback speed/tempo without changing pitch

### »» Pitch Shifting

Change pitch without changing tempo / playback speed

### »» Terms

»» Time / pitch scaling

»» Time expansion / compression

## Applications

- » **Beat Matching:** Align tempo of two or more audio files (mashup)
- » **Key lock:** "Align" pitch of two or more audio files (mashup)
- » **Pitch / time correction:** Edit intonation, frequency deviation, vibrato, glissando
- » **Video frame rate conversion**
- » **Sample player / libraries**
- » **Sound design**
- » **Educational software:** Pitch and timing visualization

# Stretch and Pitch Factors

$$s = \frac{t_{output}}{t_{input}}$$

$$p = \frac{f_{output}}{f_{input}}$$

## Examples:

» *Half speed:  $s = 2$*

» *Half pitch:  $p = \frac{1}{2}$*

» *Semitone up / down:*

$$p_u = 2^{\frac{1}{12}} = 1.059 \quad p_d = 2^{-\frac{1}{12}} = 0.9439$$

» *100 BPM  $\rightarrow$  75 BPM:  $s = \frac{4}{3}$*


# Resampling


## »» Traditional: resampling

»» Change inter-sample 'distance' by interpolation

»» Keep playback sample rate constant

»» Audio example

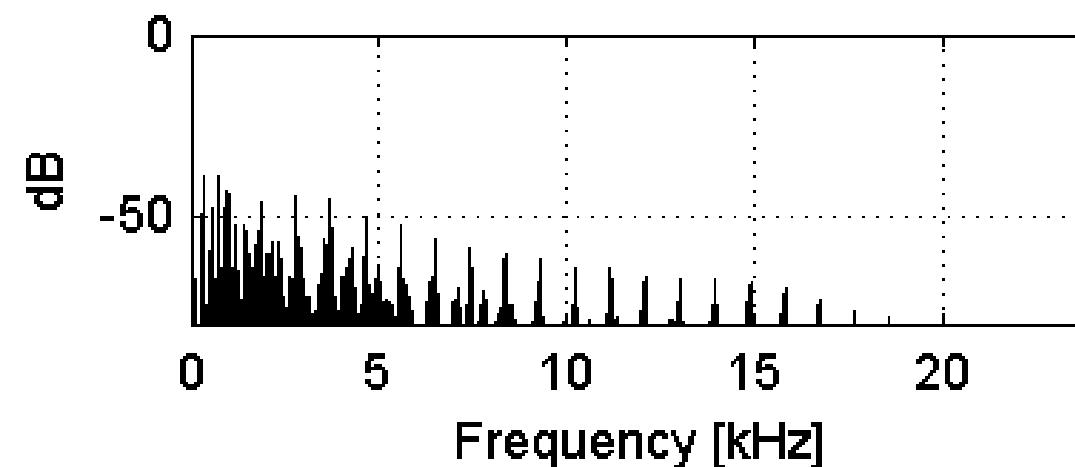
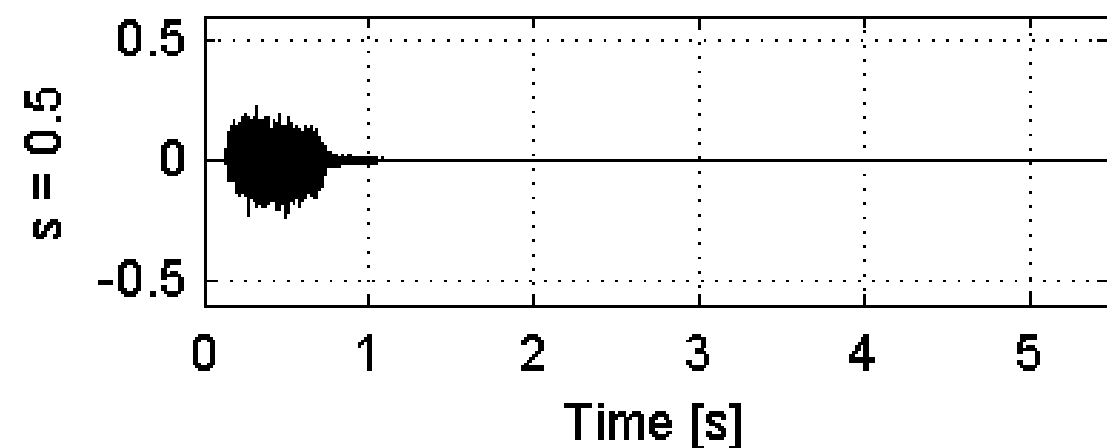
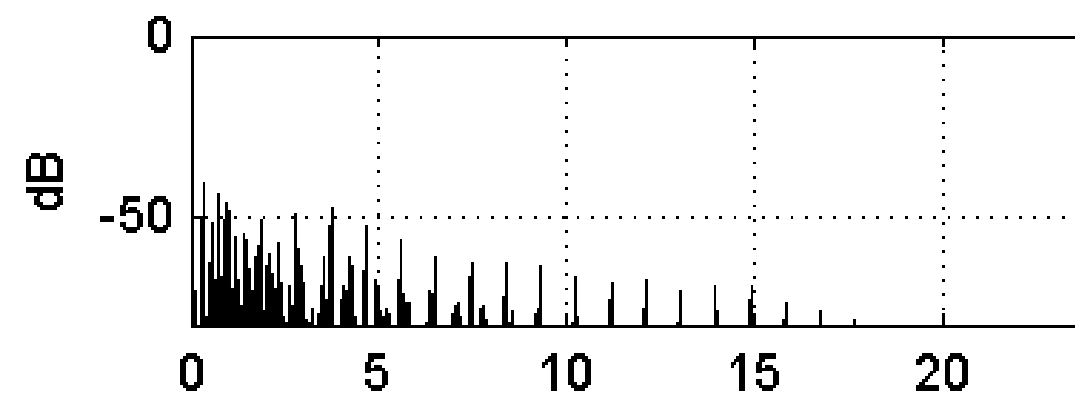
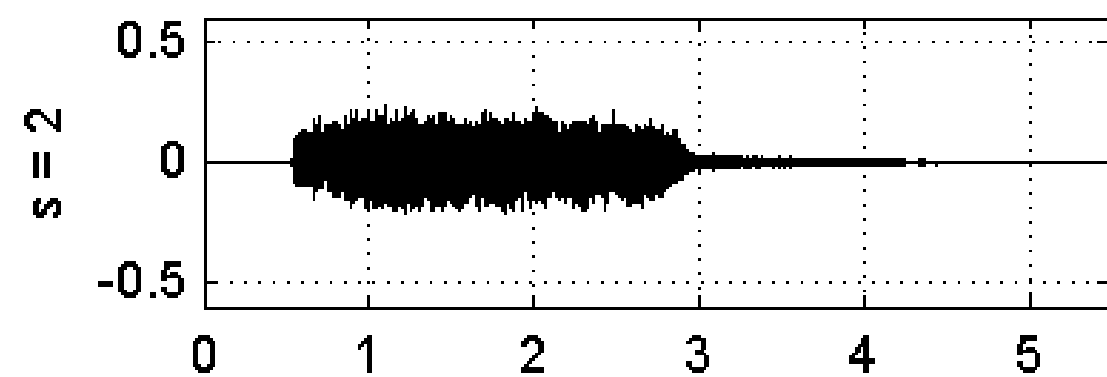
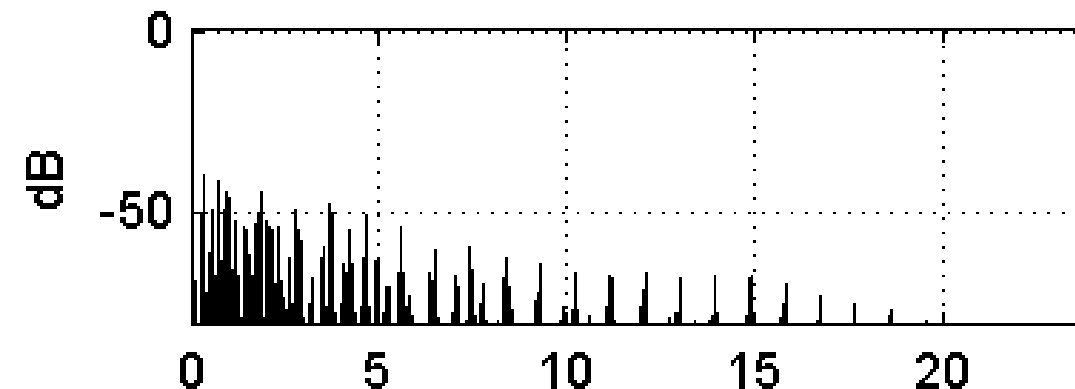
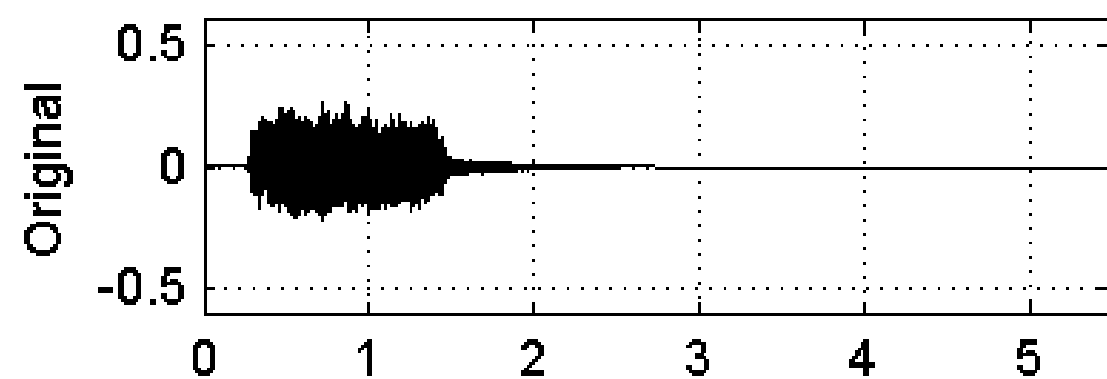
»» Original: 

»» Resample: 

»» Tempo changes results in pitch change (and vice versa)

$$s = \frac{1}{p}$$

# Stretching: Effect on Frequency Domain



## OLA: Introduction

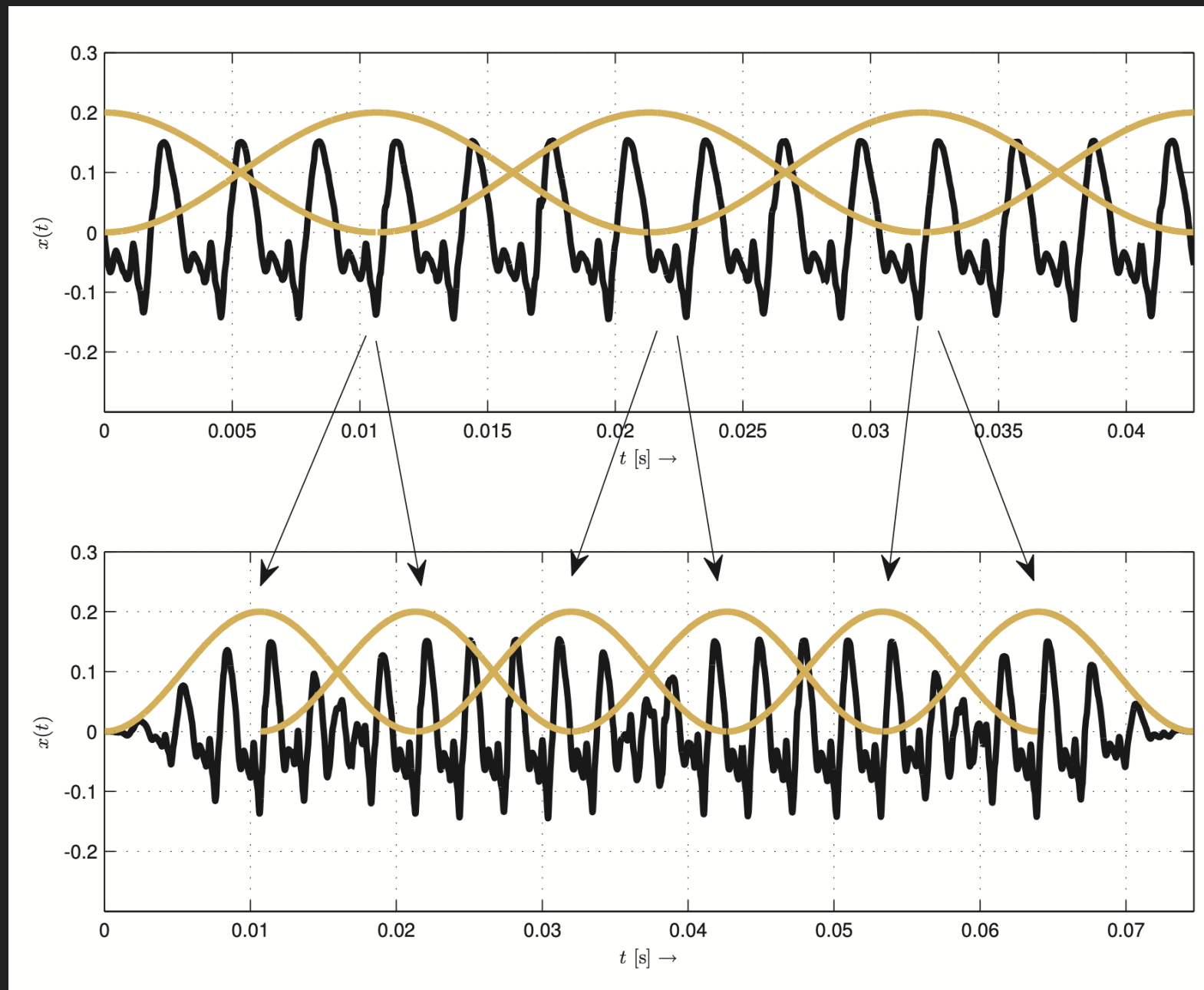
Overlap and add approaches for:

- » Granular synthesis
- » Time /frequency synthesis and processing
- » Time-stretching and pitch-shifting

# Time Stretching

## Overlap and add

1. **Split input** signal into overlapping blocks
2. **Duplicate or discard blocks** depending on stretch factor



» Original:

▶ 0:00 / 0:15

»  $s = \frac{4}{3}$ :

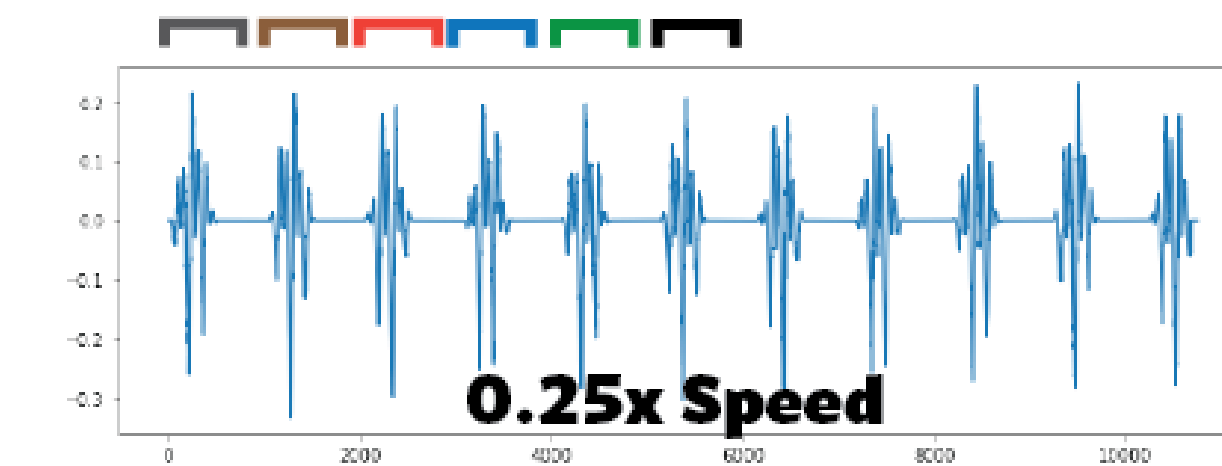
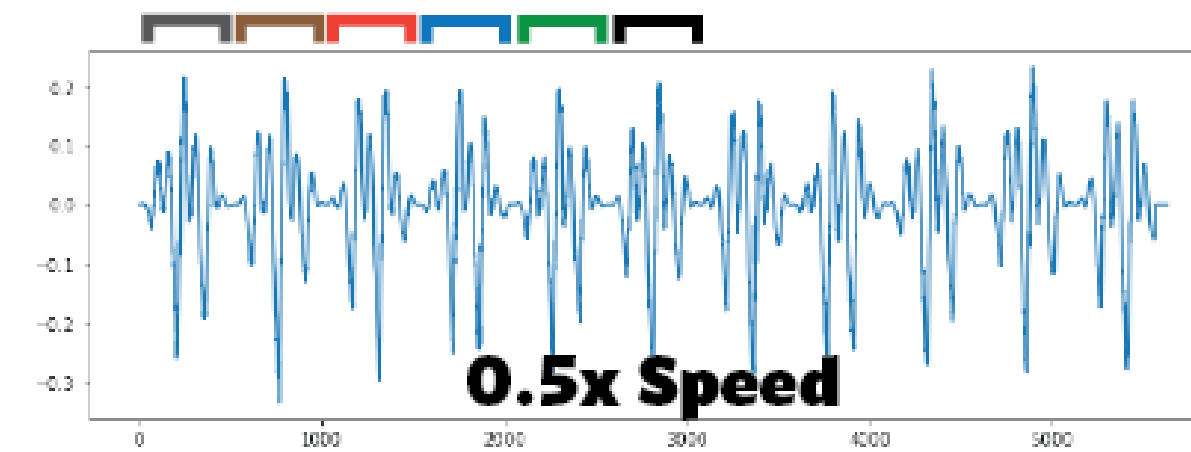
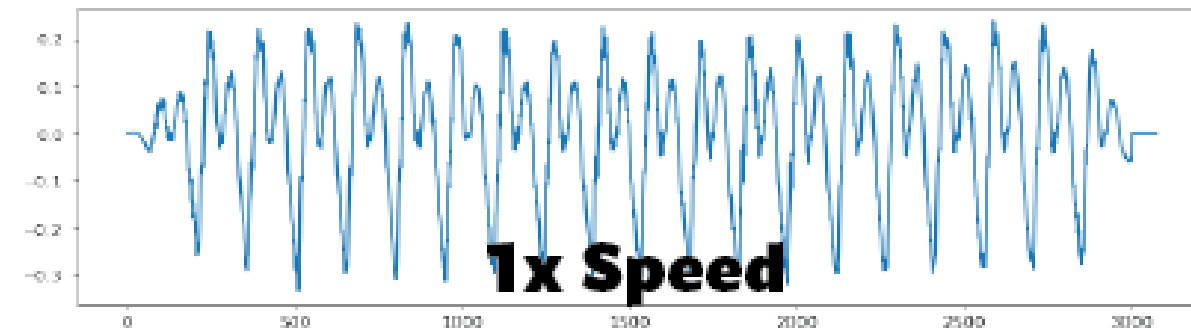
▶ 0:00 / 0:20





# Outputting Grains

## Output

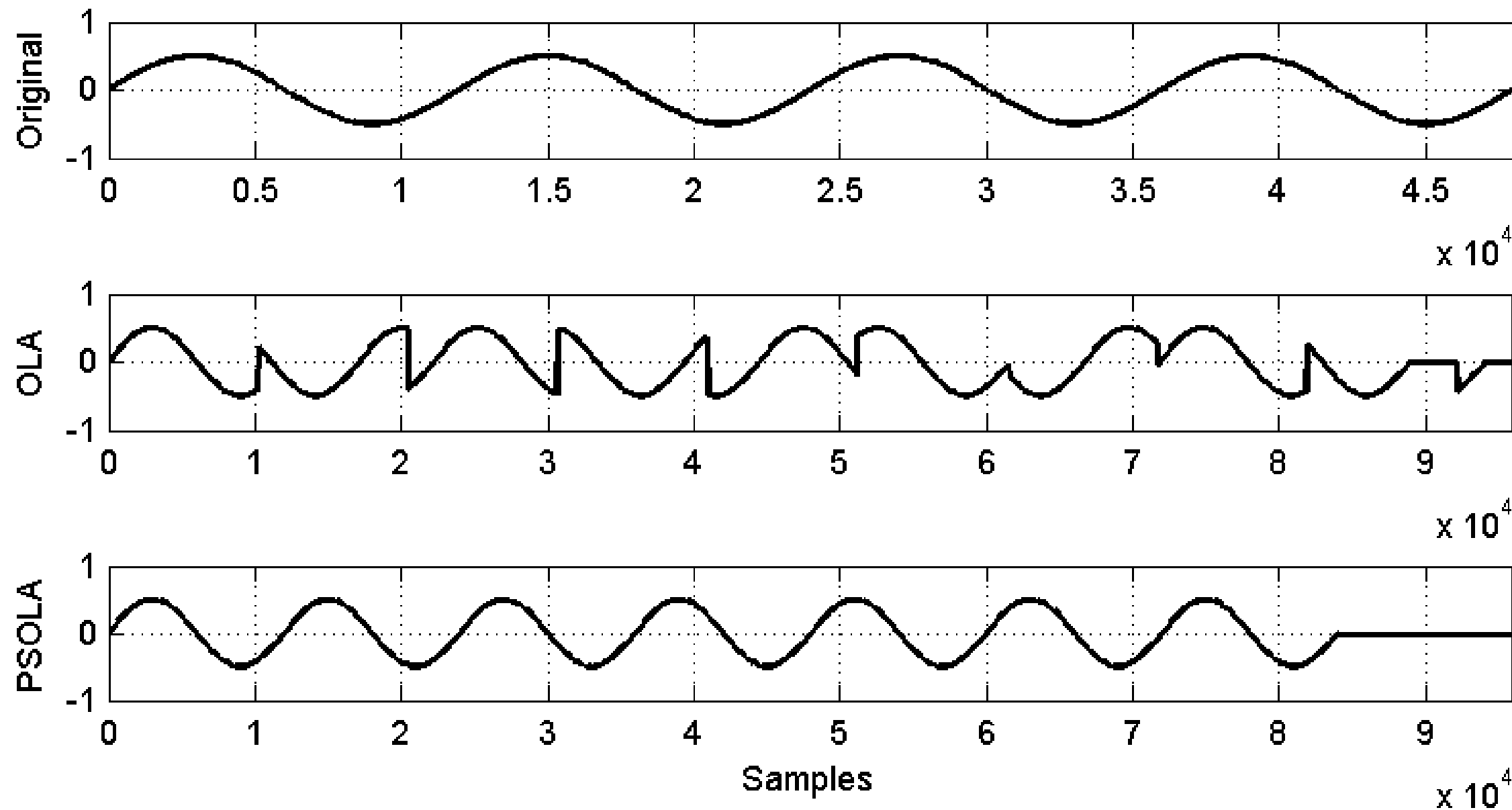


## Parameters

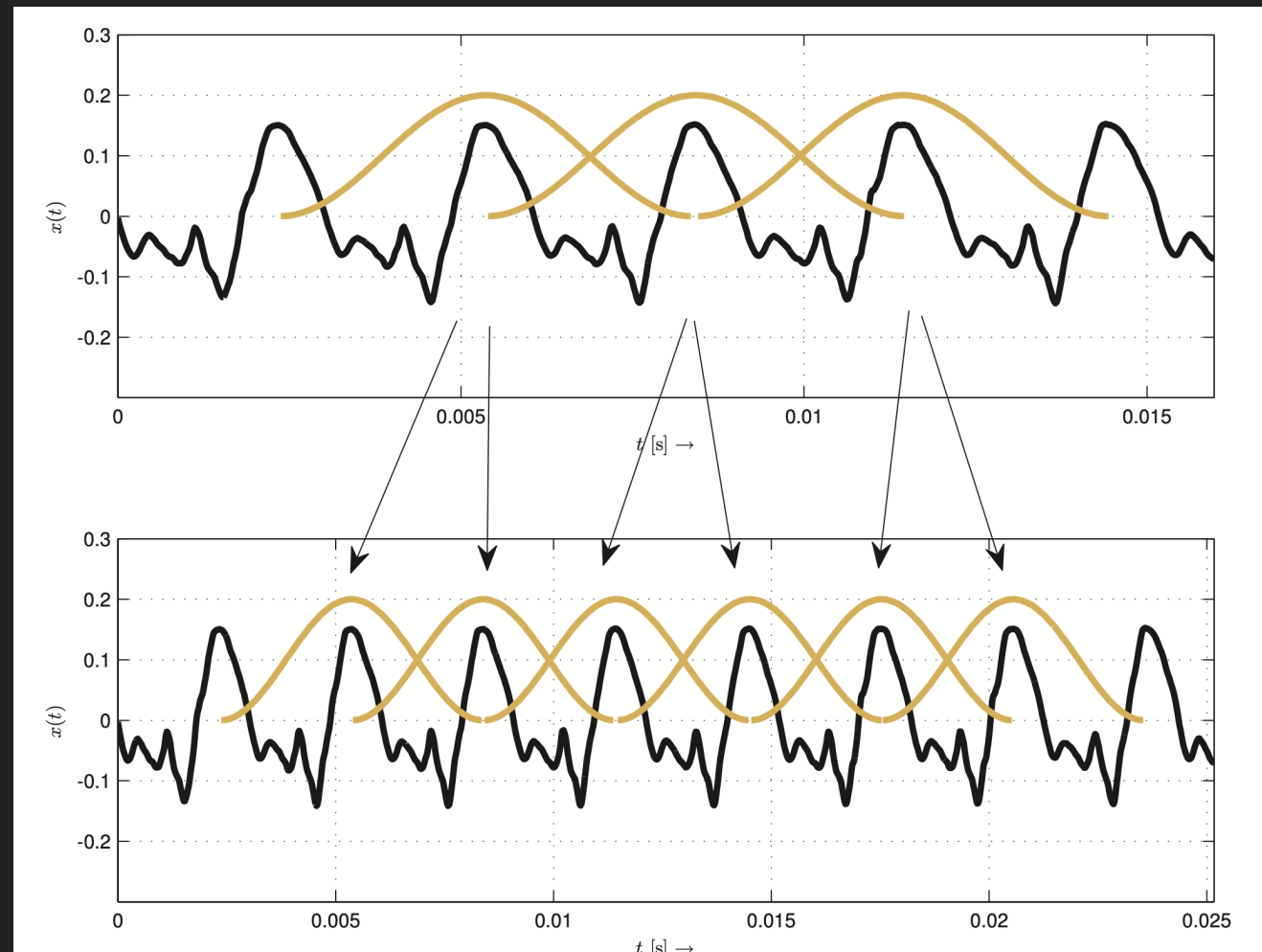
- » Grain Size
- » Hop Size
- » Time Scale
- » Pitch Scale
- » Time Variance
- » Pitch Variance
- » Interpolation Method

# Pitch Synchronous Overlap and Add

- » Use the OLA principle, but
- » **Adapt block length** to fundamental period length



# PSOLA: Example



» Original:

▶ 0:00 / 0:15

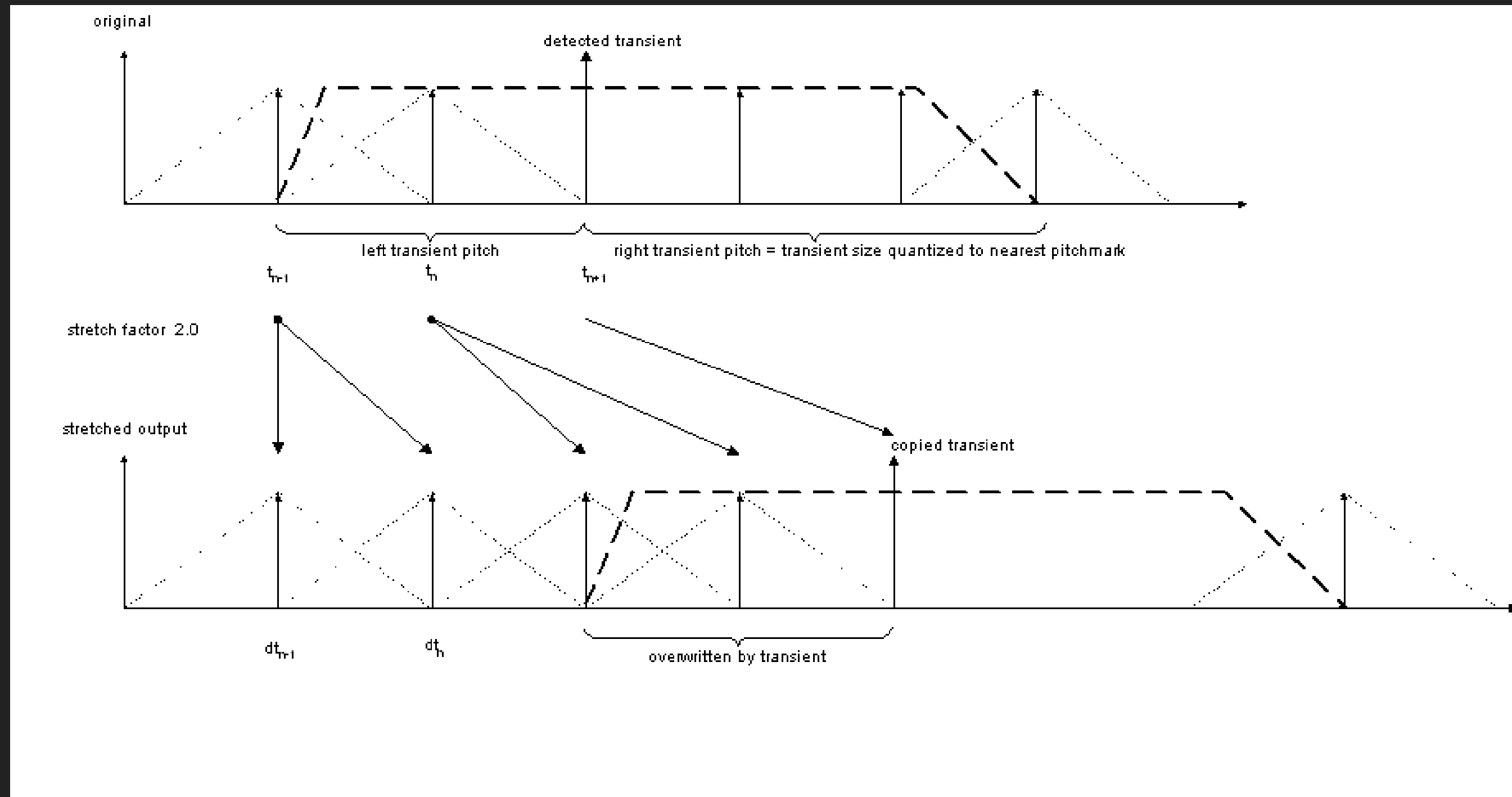


»  $s = \frac{4}{3}$ :

▶ 0:00 / 0:20



# PSOLA: Transient Copying



## PSOLA: Summary

### »» Processing Steps

Detect *fundamental frequency* / period length

Set *pitch marks*

Intelligently *select blocks* to be repeated / discarded

### »» Advantages

»» *High granularity* - Modify audio on period length resolution

»» *High quality*

# PSOLA: Summary

## »» Problems

- »» Quality depends on *pitch tracking* reliability
- »» Quality and timbre depends on *pitch mark positioning*
- »» Works only for *monophonic* input signals
  - »» Polyphonic and noisy segments
  - »» Reverberation and overlapping tones
- »» *Noise, plosives* require special consideration
- »» *Copying* artifacts (double transients, timing deviations)

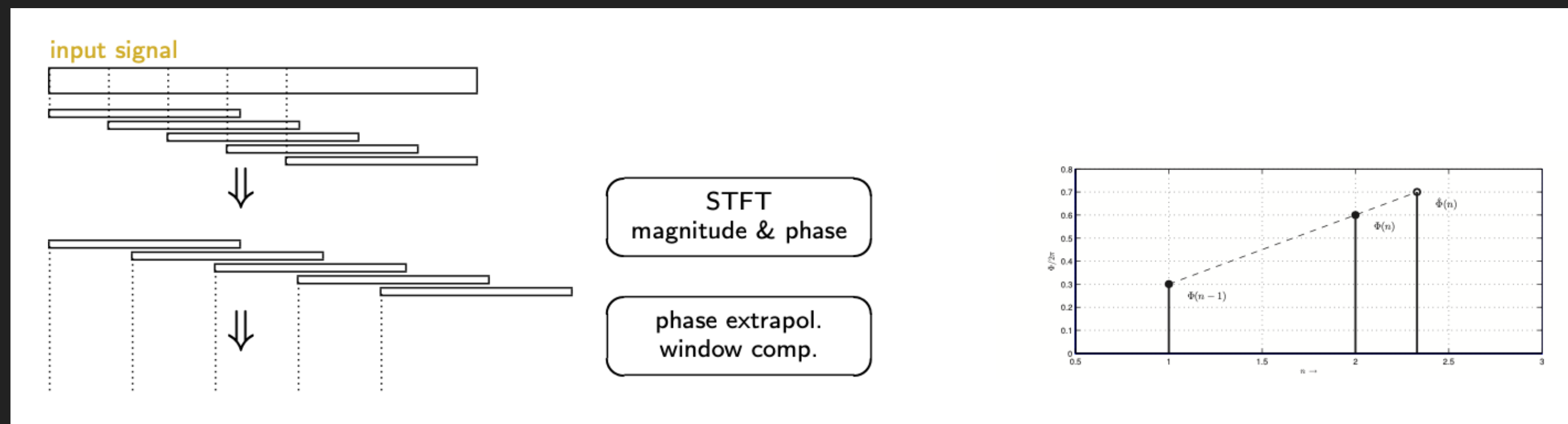
## »» Typical Applications

- »» **Standard approach for vocal editing tools**



# Phase Vocoder: Frequency Domain OLA

1. **Split input** signal into overlapping blocks
2. Compute **magnitude and phase spectrum** of each block
3. **Change overlap ratio** between blocks depending on stretch factor
4. Keep the magnitude, **adapt the phase per bin** to the blocks new time stamp



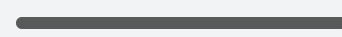
## Phase Vocoder: Audio Example



Original:



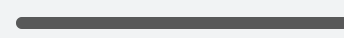
0:00 / 0:15



Resample:



0:00 / 0:20



# Frequency Reassignment: Relation of Phase and Frequency



## Phasor Representation:

- » Sine value is defined by magnitude and phase
- » Decreasing the amplitude → shorter vector
- » Increasing the frequency → Increasing speed

Frequency and phase change closely related

» 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

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

Frequency and phase change closely related

» In other words:

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

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

# Frequency Reassignment: Principles

Frequency domain

» Instead of using the bin frequency

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

» 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

# 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*

## Phase Unwrapping

- » 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 estimates range

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



## Phase Unwrapping

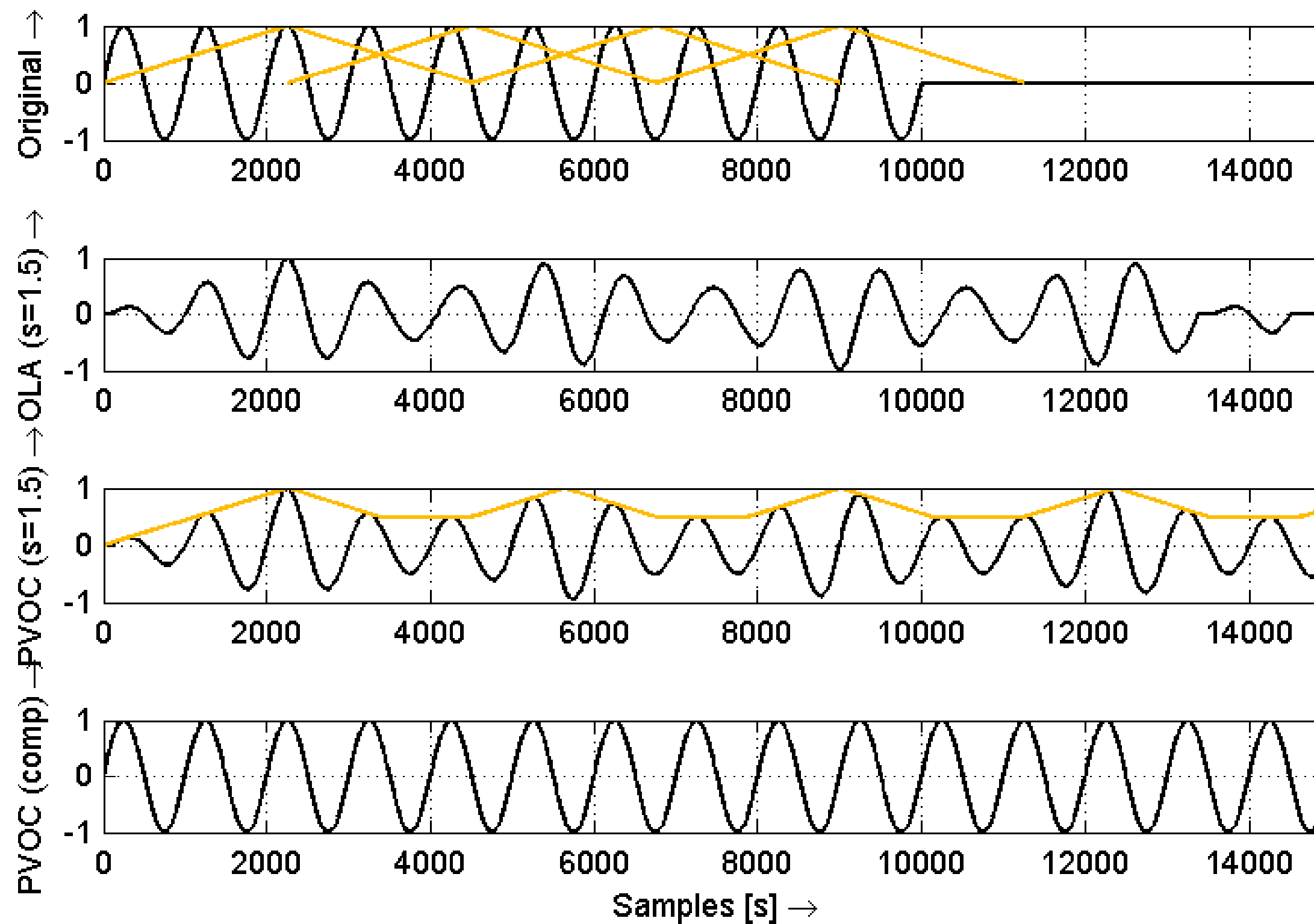
» Compute unwrapped phase difference

$$\begin{aligned}\Delta\Phi_u(k, n) &= \Phi_u(k, n) - \Phi(k, n - 1) \\ &= \hat{\Phi}(k, n) + \text{princarg}\left[\Phi(k, n) - \hat{\Phi}(k, n)\right] - \Phi(k, n - 1) \\ &= \frac{2\pi k}{\mathcal{K}}\mathcal{H} + \text{princarg}\left[\Phi(k, n) - \Phi(k, n - 1) - \frac{2\pi k}{\mathcal{K}}\right]\end{aligned}$$

## Frequency Reassignment: Problems

- » **Overlapping spectral components**
  - » Sinusoidal components often overlap (Spectral leakage, several instruments playing the same pitch, ...)
  - » Incorrect phase estimates
  - » Spectrum should be as sparse as possible, increase STFT length
- » **Inaccurate phase unwrapping**

# Phase Vocoder Window Compensation



# Phase Vocoder – Properties & Artifacts

## » Advantages

- » Allows *polyphonic input* (assumption: no overlapping harmonics)
- » Absolute *timing stability* (i.e., sample resolution)

## » Disadvantages

- » *Low granularity* - FFT block size
- » Artifacts: Phasing, Transient smearing / doubling







