Lecture 24: Downsampling & Upsampling

Foundations of Digital Signal Processing

Outline

- Downsampling
- Downsampling with Anti-Aliasing Filter
- Upsampling
- Upsampling with Reconstruction Filter
- Resampling

News

Homework Assignment #8

Due <u>Friday</u> by 11:59 PM

Coding Assignment #4

- Due <u>Today</u> by 11:59 PM
- Submit via canvas
 - Submit answers as a PDF
 - Submit code as .m files

Exam #2

In one week

Lecture 24: Downsampling & Upsampling

Foundations of Digital Signal Processing

Outline

- Downsampling
- Downsampling with Anti-Aliasing Filter
- Upsampling
- Upsampling with Reconstruction Filter
- Resampling

Reduce Sampling Rates





Downsample by 2



Image from Martin Vertelli's notes

Reduce Sampling Rates

$$x[n] \longrightarrow y[n] = x[Mn]$$

Downsample by M



Reduce Sampling Rates

$$x[n] \longrightarrow y[n] = x[2n]$$

Downsample by 2



Reduce Sampling Rates

$$x[n] \longrightarrow y[n] = x[2n]$$

Downsample by 2



Reduce Sampling Rates

$$x[n] \longrightarrow y[n] = x[2n]$$

Downsample by 2



$$x[n] \longrightarrow y[n] = x[2n]$$

- Sketch each signal period by 2 AROUND every 2π
- Reduce amplitude by 2 times

$$\mathcal{F}\{x[2n]\} = \frac{1}{2} \left[X(\omega/2) + X(\omega/2 - \pi) \right]$$

- Sketch each signal period by 2 AROUND every 2π
- Reduce amplitude by 2 times





- Sketch each signal period by 2 AROUND every 2π
- Reduce amplitude by 2 times





- Sketch each signal period by 2 AROUND every 2π
- Reduce amplitude by 2 times





- Sketch each signal period by 2 AROUND every 2π
- Reduce amplitude by 2 times





- Sketch each signal period by 2 AROUND every 2π
- Reduce amplitude by 2 times





- Sketch each signal period by 2 AROUND every 2π
- Reduce amplitude by 2 times

 x[n]
 Downsample by 2



- Sketch each signal period by 2 AROUND every 2π
- Reduce amplitude by 2 times





Question

What is the condition to prevent aliasing for downsampling by 2?



Question

• What is the condition to prevent aliasing for downsampling by 2?





$$x[n] \longrightarrow y[n] = x[Mn]$$

- Sketch each signal period by M AROUND every 2π
- Reduce amplitude by M times

$$\mathcal{F}\{x[Mn]\} = \frac{1}{M} \sum_{k=0}^{M-1} X\left(\frac{\omega}{M} - \frac{2\pi}{M}k\right)$$

- Sketch each signal period by M AROUND every 2π
- Reduce amplitude by M times x[n]Downsample by M=4 n -20 -10 10 20 x(_ω) 0.4 0.3 0.2 0.1 ω 0.5 _π -0.5 π

- Sketch each signal period by M AROUND every 2π
- Reduce amplitude by M times





Question

• What is the condition to prevent aliasing for downsampling by 2?



Question

• What is the condition to prevent aliasing for downsampling by 2?



Question: How do we prevent aliasing?

$$x[n] \longrightarrow M \longrightarrow y[n] = x[Mn]$$

- Sketch each signal period by M AROUND every 2π
- Reduce amplitude by M times

$$\mathcal{F}\{x[Mn]\} = \frac{1}{M} \sum_{k=0}^{M-1} X\left(\frac{\omega}{M} - \frac{2\pi}{M}k\right)$$

Question: How do we prevent aliasing?

$$x[n] \longrightarrow y[n] = x[Mn]$$

- Sketch each signal period by M AROUND every 2π
- Reduce amplitude by M times

Question:

What is the requirement to prevent aliasing?

Question: How do we prevent aliasing?

$$x[n] \longrightarrow y[n] = x[Mn]$$

- Sketch each signal period by M AROUND every 2π
- Reduce amplitude by M times

Question:

- What is the requirement to prevent aliasing?
- $M < \pi/M$

Lecture 24: Downsampling & Upsampling

Foundations of Digital Signal Processing

Outline

- Downsampling
- Downsampling with Anti-Aliasing Filter
- Upsampling
- Upsampling with Reconstruction Filter
- Resampling

Question: How do we prevent aliasing?

$$x[n] \longrightarrow G(z) \longrightarrow M \longrightarrow y[n] = x[Mn]$$
Low pass filter
Cutoff: π/M
Gain: 1

- Apply anti-aliasing
- Sketch each signal period by M AROUND every 2π
- Reduce amplitude by M times

Without Anti-Aliasing



With Anti-Aliasing



Lecture 24: Downsampling & Upsampling

Foundations of Digital Signal Processing

Outline

- Downsampling
- Downsampling with Anti-Aliasing Filter
- Upsampling
- Upsampling with Reconstruction Filter
- Resampling

Increase Sampling Rates

$$x[n] \longrightarrow 1 \longrightarrow y[n]$$

How do we increase the sampling rate?

Increase Sampling Rates

$$x[n] \longrightarrow y[n] ? x[n/N]$$

How do we increase the sampling rate?

Increase Sampling Rates

$$x[n] \longrightarrow y[n] ? x[n/N]$$

Upsample by M



Increase Sampling Rates

$$x[n] \longrightarrow 12 \longrightarrow y[n] ? x[n/2]$$

Upsample by 2



Increase Sampling Rates

$$x[n] \longrightarrow 12 \longrightarrow y[n] ? x[n/2]$$

Upsample by 2



So what happens in frequency?



- Shrink entire frequency-domain signal by 2
- Amplitude does <u>not</u> change

 $\mathcal{F}\{y[n]\} = X(2\omega)$

- Shrink entire frequency-domain signal by 2
- Amplitude does <u>not</u> change





So what happens in frequency?

- Shrink entire frequency-domain signal by 2
- Amplitude does <u>not</u> change





So what happens in frequency?



- Shrink entire frequency-domain signal by N
- Amplitude does <u>not</u> change

 $\mathcal{F}\{y[n]\} = X(N\omega)$

- Shrink entire frequency-domain signal by 2
- Amplitude does <u>not</u> change





- Shrink entire frequency-domain signal by 3
- Amplitude does <u>not</u> change





- Shrink entire frequency-domain signal by 4
- Amplitude does <u>not</u> change





Lecture 24: Downsampling & Upsampling

Foundations of Digital Signal Processing

Outline

- Downsampling
- Downsampling with Anti-Aliasing Filter
- Upsampling
- Upsampling with Reconstruction Filter
- Resampling

Question: How do we get the higher sampling rate?

Question: How do we get the higher sampling rate?

$$x[n] \longrightarrow \uparrow N \longrightarrow H(z) \longrightarrow y[n]$$

Low pass filter
Cutoff: π/N
Gain: N

- Shrink entire frequency-domain signal by N
- Amplitude does <u>not</u> change
- Apply reconstruction filter

- Shrink entire frequency-domain signal by 2
- Amplitude does <u>not</u> change





So what happens in frequency?

Shrink entire frequency-domain signal by 2







So what happens in frequency?

Shrink entire frequency-domain signal by 2







So what happens in frequency?

- Shrink entire frequency-domain signal by 3
- Amplitude does <u>not</u> change





So what happens in frequency?

Shrink entire frequency-domain signal by 4







Lecture 24: Downsampling & Upsampling

Foundations of Digital Signal Processing

Outline

- Downsampling
- Downsampling with Anti-Aliasing Filter
- Upsampling
- Upsampling with Reconstruction Filter
- Resampling

Question: How do non-integer sampling rate changes?

Question: How do non-integer sampling rate changes?

$$x[n] \longrightarrow 1 \longrightarrow H(z) \longrightarrow y[n]$$

Low pass filter Cutoff: π/N Gain: N

Question: How do non-integer sampling rate changes?



Question: How do non-integer sampling rate changes?

$$x[n] \longrightarrow V(z) \longrightarrow y[n]$$
Low pass filter
$$Cutoff: \min\left(\frac{\pi}{N}, \frac{\pi}{M}\right)$$
Gain: N

Question: How do non-integer sampling rate changes?



Foundations of Digital Signal Processing Lecture 24: Multi-rate Processing

Question: How do non-integer sampling rate changes?

