### **Digital Signal Processing**

Prof. Nizamettin AYDIN

naydin@yildiz.edu.tr

http://www.yildiz.edu.tr/~naydin

### **Digital Signal Processing**

Lecture 8

**Sampling & Aliasing** 

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### **READING ASSIGNMENTS**

- This Lecture:
  - Chap 4, Sections 4-1 and 4-2
    - Replaces Ch 4 in DSP First, pp. 83-94
- Other Reading:
  - Recitation: Strobe Demo (Sect 4-3)
  - Next Lecture: Chap. 4 Sects. 4-4 and 4-5

### LECTURE OBJECTIVES

- SAMPLING can cause ALIASING
  - Sampling Theorem
  - Sampling Rate > 2(Highest Frequency)
- Spectrum for digital signals, x[n]
  - Normalized Frequency

$$\hat{\omega} = \omega T_s = \frac{2\pi f}{f_s} + 2\pi \ell$$

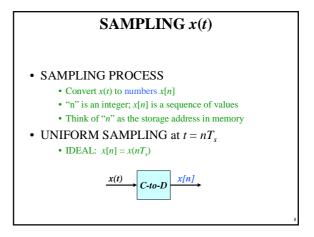
$$\hat{ALIASING}$$

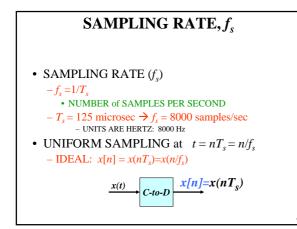
## **SYSTEMS Process Signals**

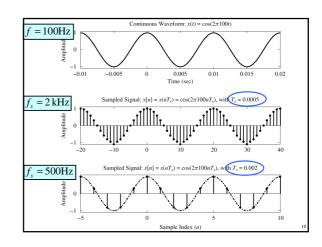


- PROCESSING GOALS:
  - Change x(t) into y(t)
    - For example, more BASS
  - Improve x(t), e.g., image deblurring
  - Extract Information from x(t)

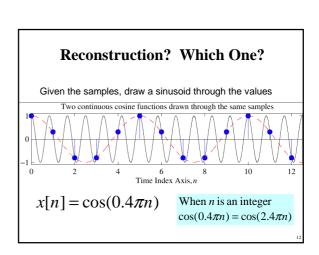
# System IMPLEMENTATION • ANALOG/ELECTRONIC: • Circuits: resistors, capacitors, op-amps x(t) ELECTRONICS • DIGITAL/MICROPROCESSOR • Convert x(t) to numbers stored in memory x(t) A-to-D x[n] COMPUTER y[n] D-to-A y(t)







## • HOW OFTEN? - DEPENDS on FREQUENCY of SINUSOID - ANSWERED by SHANNON/NYQUIST Theorem - ALSO DEPENDS on "RECONSTRUCTION" Shannon Sampling Theorem A continuous-time signal x(t) with frequencies no higher than $f_{\max}$ can be reconstructed exactly from its samples $x[n] = x(nT_s)$ , if the samples are taken at a rate $f_s = 1/T_s$ that is greater than $2f_{\max}$



### STORING DIGITAL SOUND

- x[n] is a SAMPLED SINUSOID
  - A list of numbers stored in memory
- EXAMPLE: audio CD
- CD rate is 44,100 samples per second
  - 16-bit samples
  - Stereo uses 2 channels
- Number of bytes for 1 minute is
  - $-2 \times (16/8) \times 60 \times 44100 = 10.584$  Mbytes

### **DISCRETE-TIME SINUSOID**

• Change x(t) into x[n] **DERIVATION** 

$$x(t) = A\cos(\omega t + \varphi)$$

$$x[n] = x(nT_s) = A\cos(\omega nT_s + \varphi)$$

$$x[n] = A\cos((\omega T_s)n + \varphi)$$

$$x[n] = A\cos(\hat{\omega}n + \varphi)$$

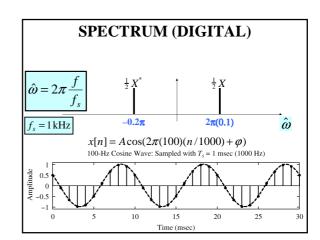
$$\hat{\omega} = \omega T_s = \frac{\omega}{f}$$
 DEFINE DIGITAL FREQUENCY

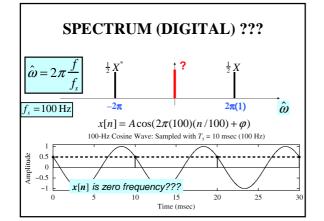
## **DIGITAL FREQUENCY**



- $\hat{\omega}$  VARIES from 0 to  $2\pi$ , as f varies from 0 to the sampling frequency
- UNITS are radians, not rad/sec
  - DIGITAL FREQUENCY is NORMALIZED

$$\hat{\omega} = \omega T_s = \frac{2\pi f}{f}$$





### The REST of the STORY

- Spectrum of x[n] has more than one line for each complex exponential
  - Called **ALIASING**
  - MANY SPECTRAL LINES
- SPECTRUM is PERIODIC with period =  $2\pi$

$$A\cos(\hat{\omega}n + \varphi) = A\cos((\hat{\omega} + 2\pi)n + \varphi)$$

### ALIASING DERIVATION

• Other Frequencies give the same  $\hat{\omega}$   $x_1(t) = \cos(400\pi t) \quad \text{sampled at } f_s = 1000 \,\text{Hz}$   $x_1[n] = \cos(400\pi \frac{n}{1000}) = \cos(0.4\pi n)$   $x_2(t) = \cos(2400\pi t) \quad \text{sampled at } f_s = 1000 \,\text{Hz}$   $x_2[n] = \cos(2400\pi \frac{n}{1000}) = \cos(2.4\pi n)$   $x_2[n] = \cos(2.4\pi n) = \cos(0.4\pi n + 2\pi n) = \cos(0.4\pi n)$   $\Rightarrow x_2[n] = x_1[n]$   $2400\pi - 400\pi = 2\pi(1000)$ 

### **ALIASING DERIVATION-2**

• Other Frequencies give the same  $\hat{\omega}$ If  $x(t) = A\cos(2\pi(f + \ell f_s)t + \varphi)$   $t \leftarrow \frac{n}{f_s}$ and we want :  $x[n] = A\cos(\hat{\omega}n + \varphi)$ then :  $\hat{\omega} = \frac{2\pi(f + \ell f_s)}{f_s} = \frac{2\pi f}{f_s} + \frac{2\pi \ell f_s}{f_s}$  $\hat{\omega} = \omega T_s = \frac{2\pi f}{f_s}$ 

### **ALIASING CONCLUSIONS**

- ADDING  $f_s$  or  $2f_s$  or  $-f_s$  to the FREQ of x(t) gives exactly the same x[n]
  - The samples,  $x[n] = x(n/f_s)$  are EXACTLY THE <u>SAME VALUES</u>
- GIVEN x[n], WE CAN'T DISTINGUISH  $f_o$  FROM  $(f_o + f_s)$  or  $(f_o + 2f_s)$

## NORMALIZED FREQUENCY

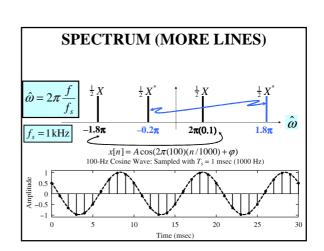
• DIGITAL FREQUENCY
Normalized Radian Frequency

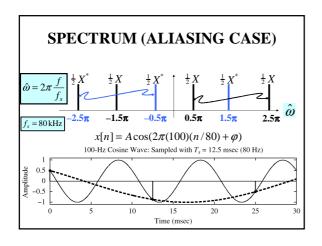
$$\hat{\omega} = \omega T_s = \frac{2\pi f}{f_s} + 2\pi \ell$$

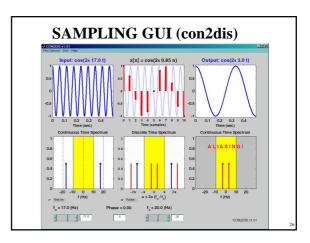
Normalized Cyclic Frequency  $\hat{f} = \hat{\omega}/(2\pi) = fT_s = f/f_s$ 

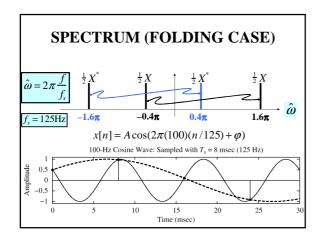
## SPECTRUM for x[n]

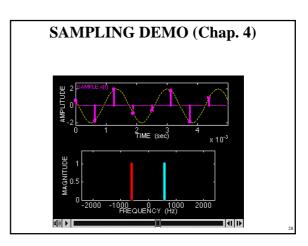
- PLOT versus NORMALIZED FREQUENCY
- INCLUDE **ALL** SPECTRUM LINES
  - ALIASES
    - ADD MULTIPLES of 2π
    - SUBTRACT MULTIPLES of  $2\pi$
  - FOLDED ALIASES
    - (to be discussed later)
    - ALIASES of NEGATIVE FREQS











### **ALIASING DERIVATION**

• Other Frequencies give the same  $\hat{\omega}$ If  $x(t) = A\cos(2\pi(\underline{f} + \ell f_s)t + \varphi)$ and we substitute:  $t \leftarrow \frac{n}{f_s}$ then:  $x[n] = A\cos(2\pi(f + \ell f_s)\frac{n}{f_s} + \varphi)$ or,  $x[n] = A\cos(2\pi f_s n + 2\pi \ell n + \varphi)$ 

• Other Frequencies give the same 
$$\hat{\omega}$$
  
If  $x(t) = A\cos(2\pi(f + \ell f_s)t + \varphi)$  and we want:  $x[n] = A\cos(\hat{\omega}n + \varphi)$   
then:  $\hat{\omega} = \frac{2\pi(f + \ell f_s)}{f_s} = \frac{2\pi f}{f_s} + \frac{2\pi \ell f_s}{f_s}$   
 $\hat{\omega} = \omega T_s = \frac{2\pi f}{f_s}$ 

**ALIASING DERIVATION-2** 

### FOLDING DERIVATION

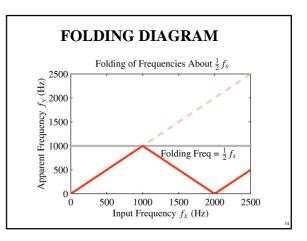
• Negative Freqs can give the same  $x(t) = A\cos(2\pi(-f + \ell f_s)t - \varphi)$   $x[n] = x(nT_s) = A\cos(2\pi(-f + \ell f_s)nT_s - \varphi)$   $x[n] = A\cos((-2\pi fT_s)n + (2\pi \ell f_sT_s)n - \varphi)$   $x[n] = A\cos((2\pi fT_s)n - 2\pi \ell n + \varphi)$   $x[n] = A\cos(\hat{\varphi}n + \varphi)$   $x[n] = A\cos(\hat{\varphi}n + \varphi)$ SAME DIGITAL SIGNAL

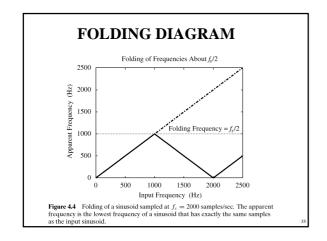
## FOLDING (a type of ALIASING)

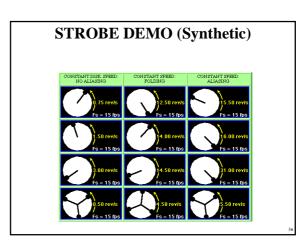
- MANY x(t) give IDENTICAL x[n]
- CAN'T TELL  $f_0$  FROM  $(f_s-f_0)$ 
  - Or,  $(2f_s-f_0)$  or,  $(3f_s-f_0)$
- EXAMPLE:
  - -y(t) has 1000 Hz component
  - SAMPLING FREQ = 1500 Hz
  - WHAT is the "FOLDED" ALIAS?

 $-1000+1500 \rightarrow 500$ 

# Normalized Radian Frequency $\hat{\omega} = \omega T_s = \frac{2\pi f}{f_s} + 2\pi \ell$ $\hat{\omega} = \omega T_s = -\frac{2\pi f}{f_s} + 2\pi \ell$ Folded Alias







## ALIASING DERIVATION

• Other Frequencies give the same  $\hat{\omega}$ 



$$x(t) = A\cos(2\pi(f + \ell f_s)t + \varphi)$$

$$x[n] = x(nT_s) = A\cos(2\pi(f + \ell f_s)nT_s + \varphi)$$

$$x[n] = A\cos((2\pi fT_s)n + (2\pi \ell f_s T_s)n + \varphi)$$

$$x[n] = A\cos((2\pi fT_s)n + \underline{2\pi\ell n} + \varphi)$$

$$x[n] = A\cos(\hat{\omega}n + \varphi)$$
  $\hat{\omega} = 2\pi f T_s = \frac{2\pi f}{f_s}$