

GEORGIA INSTITUTE OF TECHNOLOGY  
SCHOOL of ELECTRICAL and COMPUTER ENGINEERING

**ECE 2025 Fall 2001**  
**Problem Set #5**

Assigned: 21-Sept-01

Due Date: Week of 1-Oct-01

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Reading: Chapter 4 (**Revised**) on *Sampling*, and Chapter 5, pp. 119–133.

• Please check the “Bulletin Board” often. All official course announcements are posted there.

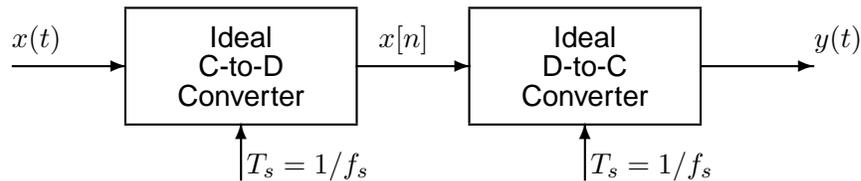
ALL of the **STARRED** problems will have to be turned in for grading. A solution will be posted to the web. Some problems have solutions similar to those found on the CD-ROM or in the class archives.

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Your homework is due in recitation at the beginning of class. After the beginning of your assigned recitation time, the homework is considered late and will be given a zero.

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**PROBLEM 5.1\*:**



Shown in the figure above is an ideal C-to-D converter that samples  $x(t)$  with a sampling period  $T_s$  to produce the discrete-time signal  $x[n]$ . The ideal D-to-C converter then forms a continuous-time signal  $y(t)$  from the samples  $x[n]$ . Suppose that  $x(t)$  is given by

$$x(t) = [15 + 30 \sin(250\pi t)] \cos(1000\pi t)$$

- Sketch the two-sided spectrum of this signal. Be sure to label important features of the plot. *Hint: Recall the AM spectrum from a previous homework set.*
- Is this waveform periodic? If so, what is the period?
- What is the minimum sampling rate  $f_s$  that can be used in the above system so that  $y(t) = x(t)$ ?

**PROBLEM 5.2:**

A non-ideal D-to-C converter takes a sequence  $y[n]$  as input and produces a continuous-time output  $y(t)$  according to the relation

$$y(t) = \sum_{n=-\infty}^{\infty} y[n]p(t - nT_s)$$

where  $T_s = 0.1$  second. The input sequence is given by the formula

$$y[n] = \begin{cases} 1 & 0 \leq n \leq 4 \\ (0.5)^{(n-4)} & 5 \leq n \leq 9 \\ 0 & \text{otherwise} \end{cases}$$

- (a) Plot  $y[n]$  versus  $n$ .  
 (b) For the pulse shape

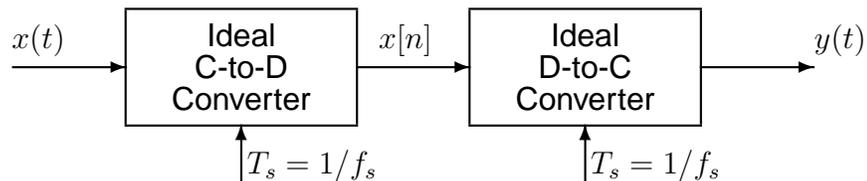
$$p(t) = \begin{cases} 1 & -0.05 \leq t \leq 0.05 \\ 0 & \text{otherwise} \end{cases}$$

carefully sketch the output waveform  $y(t)$  over its nonzero region.

- (c) For the pulse shape

$$p(t) = \begin{cases} 1 - 10|t| & -0.1 \leq t \leq 0.1 \\ 0 & \text{otherwise} \end{cases}$$

carefully sketch the output waveform  $y(t)$  over its nonzero region.

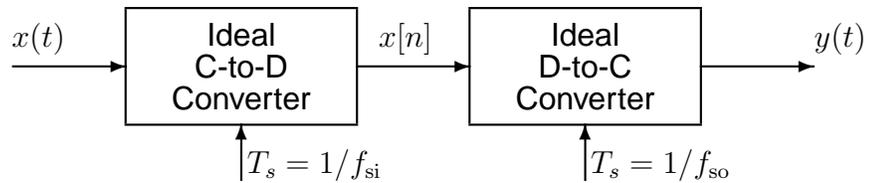
**PROBLEM 5.3:**

Chirps are very useful signals for probing the behavior of sampling operations and illustrating the “folding” type of aliasing (see Fig. 4.4 in the book).

- (a) If the input to the ideal C/D converter is  $x(t) = 7 \cos(1800\pi t + \pi/4)$ , and the sampling frequency is 1000 Hz, then the output  $y(t)$  is a sinusoid. Determine the formula for the output signal.  
 (b) Suppose that the input signal is a chirp signal defined as follows:

$$x(t) = \cos(2000\pi t - 400\pi t^2) \quad \text{for } 0 \leq t \leq 5 \text{sec.}$$

If the sampling rate is  $f_s = 1000$  Hz, then the output signal  $y(t)$  will have time-varying frequency content. Draw a graph of the resulting analog *instantaneous* frequency (in Hz) versus time of the signal  $y(t)$  **after reconstruction**. Hint: this could be done in MATLAB by putting a sampled chirp signal into the MATLAB function `specgram()`, or the DSP-First function `plotspec()`.

**PROBLEM 5.4\*:**

- (a) Suppose that the input  $x(t)$  is given by

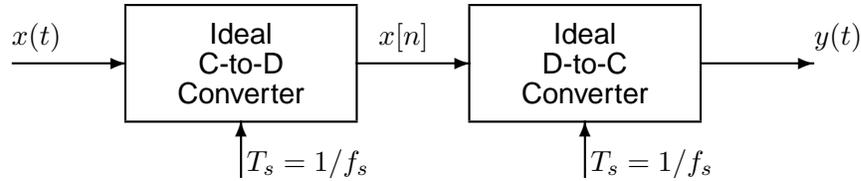
$$x(t) = 3 + 2 \cos(2\pi(3000)t - \pi/8) + 5 \cos(2\pi(8000)t + 3\pi/16)$$

Determine the spectrum for  $x[n]$  when  $f_{si} = 10000$  samples/sec. Make a plot for your answer, making sure to label the frequency, amplitude and phase of each spectral component.

- (b) Using the discrete-time spectrum from part (a), determine the analog frequency components in the output  $y(t)$  when the sampling rate of the D-to-C converter is  $f_{so} = 10000$  Hz.
- (c) Again using the discrete-time spectrum from part (a), determine the analog frequency components in the output  $y(t)$  when the sampling rate of the D-to-C converter is  $f_{so} = 20000$  Hz. In other words, the sampling rates of the two converters are different.

**PROBLEM 5.5\*:**

Consider the increasingly familiar ideal sampling and reconstruction system shown in the figure below

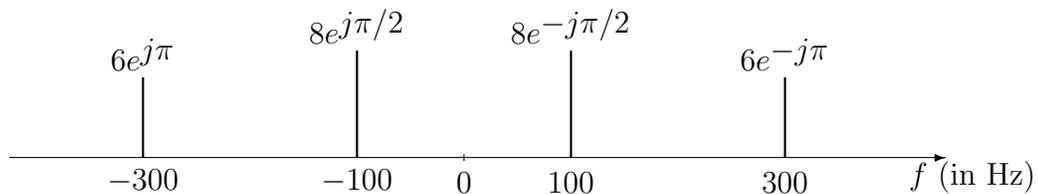


- (a) Suppose that the discrete-time signal  $x[n]$  in that figure is given by the formula

$$x[n] = 2 \cos(0.3\pi n - \pi/4)$$

If the sampling rate of the C-to-D converter is  $f_s = 10000$  samples/second, many *different* continuous-time signals  $x(t) = x_\ell(t)$  could have been inputs to that system. Determine two such inputs with frequency less than 10000 Hz; i.e., find  $x_1(t) = A_1 \cos(\omega_1 t + \phi_1)$  and  $x_2(t) = A_2 \cos(\omega_2 t + \phi_2)$  such that  $x[n] = x_1(nT_s) = x_2(nT_s)$  if  $T_s = 1/10000$  secs.

- (b) Now if the input  $x(t)$  to the system in the figure above has the two-sided spectrum representation shown below, what is the *minimum* sampling rate  $f_s$  such that the output  $y(t)$  is equal to the input  $x(t)$ ?



- (c) Determine the spectrum for  $x[n]$  when  $f_s = 300$  samples/sec. Make a plot for your answer, but label the frequency, amplitude and phase of each spectral component.

**PROBLEM 5.6\*:**

In the rotating disk and strobe demo described in Chapter 4 of *DSP First*, we observed that different flashing rates of the strobe light would make the spot on the disk stand still.

- (a) Assume that the disk is rotating in the counter-clockwise direction at a constant speed of 10 revolutions per second. Express the movement of the spot on the disk as a rotating complex phasor.
- (b) If the strobe light can be flashed at a rate of  $n$  flashes *per second* where  $n$  is an integer greater than zero, determine all possible flashing rates such that the disk can be made to stand still.  
NOTE: the only possible flashing rates are integers: 1 per second, 2 per second, 3 per second, etc.
- (c) Now assume that the flashing rate is fixed so that the interval between flashes is 60 milliseconds. Explain how the spot will move and write a complex phasor that gives the position of the spot at each flash.
- (d) Draw a spectrum plot of the discrete-time signal in part (c) to explain your answer.

**PROBLEM 5.7\*:**

Suppose that a MATLAB function is used to plot a sinusoidal signal. The following MATLAB code generates a signal  $x[n]$  and plots it. Unfortunately, the time axis of the plot is not labeled properly.

```
Ts = 0.01;
Duration = 0.3;
tt = 0 : Ts : Duration;
Fo = 392;
xx = 6*cos( 2*pi*Fo*tt - pi/2 );
%
stem( xx ) %<--- OOPS! there is no time axis
```

- (a) Make the `stem` plot of the signal. Either sketch it or plot it using MATLAB.
- (b) Plot the spectrum of the discrete-time signal.
- (c) For the plot in part (a) or the spectrum in (b), determine the correct formula for the discrete-time signal in the form:

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

- (d) Explain how aliasing affects the plot that you see.