

STANFORD UNIVERSITY  
DEPARTMENT of ELECTRICAL ENGINEERING  
EE 102B    Spring 2013  
Problem Set #2

Assigned: April 10, 2013

Due Date: April 17, 2013

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Reading: In *SP First*, Chapter 5 on *FIR Filters* and Chapter 6 on *Frequency Response*.

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

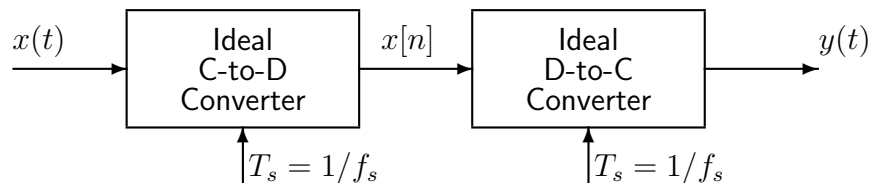
ALL of the **STARRED** problems will have to be turned in for grading. A solution will be posted to the web. Some problems have solutions similar to those found on the CD-ROM.

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**Your homework is due by 5pm on April 17, 2013.** It can be handed in up to 5pm on Friday, April 19 with a 10% penalty per day late. No credit will be given after that time.

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PROBLEM 2.1\*:



We can do some interesting things with sampling. One of them is that we can change the period of a periodic waveform. This problem illustrates how this can be done for the specific periodic input signal

$$x(t) = 2 \cos(2\pi(33)t) + \cos(2\pi(99)t).$$

In all the following parts, assume that the sampling frequency is  $f_s = 30$  Hz. Note that this sampling rate *does not* satisfy the conditions of the Shannon sampling theorem, so aliasing will occur.

- Plot the spectrum of the periodic continuous-time signal  $x(t)$ . What is the fundamental frequency of  $x(t)$ ?
- Determine an expression for the discrete-time signal  $x[n]$  as a sum of discrete-time cosine signals. Be sure that all of the normalized frequencies are positive and less than  $\pi$  radians. Plot the spectrum of  $x[n]$  over the range of normalized frequencies  $-\pi \leq \hat{\omega} \leq \pi$ .
- Now the continuous-time output signal  $y(t)$  that is created by the ideal D-to-C converter operating with sampling rate  $f_s = 30$  Hz will also be a sum of cosine signals. Write an expression for  $y(t)$  and plot its spectrum. What is the fundamental frequency of  $y(t)$ ?
- How are the fundamental frequencies of  $x(t)$  and  $y(t)$  related? Do you think that it would be possible to change the fundamental frequency by a different factor by using a different sampling rate?

PROBLEM 2.2:

Consider a continuous-time signal

$$x(t) = A \cos(\omega_0 t + \phi)$$

We know that this signal is periodic with period  $T_0 = 2\pi/\omega_0$ ; i.e.  $x(t + T_0) = x(t)$  for all  $t$ . Now suppose that  $x(t)$  is sampled to obtain the sequence

$$x[n] = x(nT_s) = A \cos(\omega_0 nT_s + \phi) = A \cos(\hat{\omega}_0 n + \phi)$$

where  $\hat{\omega}_0 = \omega_0 T_s$ .

Now a discrete-time signal is periodic with period  $N$  if  $x[n + N] = x[n]$  for all  $n$ , where  $N$  is necessarily an integer.

- (a) Will  $x[n]$  be periodic for all possible sampling rates? If not, what condition on  $T_s$  is sufficient to ensure that  $x[n]$  is periodic with period  $N$ ?
- (b) If  $\omega_0 = 2000\pi$ , what value of  $T_s$  will result in a periodic sequence with period  $N = 100$ ?

PROBLEM 2.3\*:

This problem is concerned with finding the output of an FIR filter for a given input signal. A linear time-invariant system is described by the difference equation

$$y[n] = \sum_{k=0}^3 (0.5)^k x[n - k] \tag{1}$$

- (a) Determine the filter coefficients  $\{b_k\}$  of this FIR filter.
- (b) Find the impulse response,  $h[n]$ , for this FIR filter. The impulse response is a discrete-time signal, so make a (stem) plot of  $h[n]$  versus  $n$ .
- (c) Use the difference equation in (1) to compute the output  $y[6]$  (i.e., the output at time  $n = 6$ ) when the input is

$$x[n] = \begin{cases} 0 & n < 0 \\ (n+1) & 0 \leq n \leq 4 \\ -4 & 5 \leq n \leq 10 \\ 0 & n \geq 11. \end{cases}$$

Show how you computed  $y[6]$  from the given information. A sketch of  $x[n]$  may be useful for this.

- (d) Use `conv( )` and `subplot( )` write a MATLAB program (and turn a printout of the program with a copy of the plot) to plot the input and output as a function of  $n$ .

**PROBLEM 2.4:**

Consider a system defined by 
$$y[n] = \sum_{k=N_1}^{N_2} b_k x[n-k]$$

In other words, only the coefficients  $b_{N_1}, b_{N_1+1}, \dots, b_{N_2}$  are non-zero.

Suppose that the input  $x[n]$  is non-zero only for  $N_3 \leq n \leq N_4$ . Use sketches of  $h[k]$  and  $x[n-k]$  for different values of  $n$  to show that  $y[n]$  is non-zero at most over a finite interval of the form  $N_1 + N_3 \leq n \leq N_2 + N_4$ . How many non-zero samples can there be in the output sequence  $y[n]$ ?

*Hint: consult Figs. 5.4, 5.5 and 5.6 in the book for the sliding window interpretation of the FIR filter.*

**PROBLEM 2.5\*:**

The diagram in Fig. 1 depicts a *cascade connection* of two linear time-invariant systems; i.e., the output of the first system is the input to the second system, and the overall output is the output of the second system.

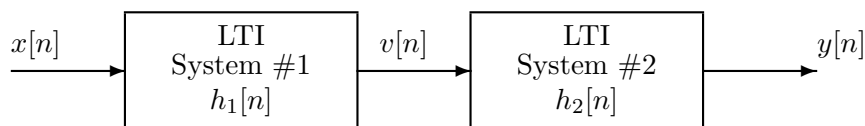


Figure 1: Cascade connection of two LTI systems.

Suppose that System #1 has impulse response,

$$h_1[n] = \begin{cases} 0 & n < 0 \\ 1 & n = 0 \\ -1 & n = 1 \\ 0 & n > 1 \end{cases}$$

and System #2 is described by the difference equation

$$y[n] = 0.25v[n] + 0.25v[n-1] + 0.25v[n-2] + 0.25v[n-3] \quad (2)$$

- Determine the difference equation of System #1; i.e., the equation that relates  $v[n]$  to  $x[n]$ .
- When the input signal  $x[n]$  is an impulse,  $\delta[n]$ , determine the signal  $v[n]$  and make a plot. Show that the resulting output is the given impulse response  $h_1[n]$ .
- From the difference equation in (2), determine  $h_2[n]$ , the impulse response of System #2.
- Determine the impulse response of the overall cascade system, i.e., find  $y[n]$  when  $x[n] = \delta[n]$ .
- From the impulse response of the overall cascade system as obtained in part (d), obtain a single difference equation that relates  $y[n]$  directly to  $x[n]$  in Fig. 1.

**PROBLEM 2.6\*:**

For each of the following systems, determine if they are (1) linear; (2) time-invariant; (3) causal.

- (a)  $y[n] = x[n] \cos(0.2\pi n)$
- (b)  $y[n] = -x[n+1] + x[n] - x[n-1]$
- (c)  $y[n] = |x[-n]|$

**PROBLEM 2.7\*:**

Consider a system implemented by the following MATLAB program:

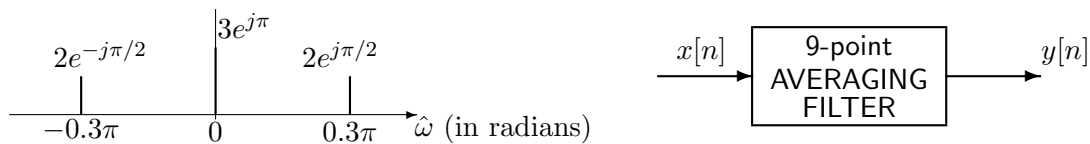
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% xx.mat is a binary file containing the  
% vector of input samples called 'xx'  
load xx  
yy1 = conv(ones(1,4),xx);  
yy2 = conv([1,-1,-1,1],xx);  
ww = yy1 + yy2;  
yy = conv(ones(1,3),ww);
```

The overall system from input  $\mathbf{xx}$  to output  $\mathbf{yy}$  is an LTI system composed of three LTI systems.

- (a) Draw a block diagram of the system that is implemented by the program above. Be sure to indicate the impulse responses of each of the three component systems.
- (b) The overall system is an LTI system. What is its impulse response, and what is the difference equation that is satisfied by the input  $x[n]$  and output  $y[n]$ ?

**PROBLEM 2.8:**

A discrete-time signal  $x[n]$  has the two-sided spectrum representation shown below.



- (a) Write an equation for  $x[n]$ . Make sure to express  $x[n]$  as a real-valued signal.
- (b) Determine a formula for the output signal  $y[n]$ .