

GEORGIA INSTITUTE OF TECHNOLOGY
SCHOOL of ELECTRICAL and COMPUTER ENGINEERING

EE 2200 Spring 1999
Problem Set #4

Assigned: 30 April 99
Due Date: 7 May 99 (FRIDAY)

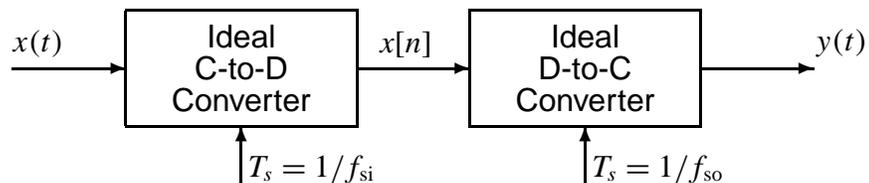
Reading: In *DSP First*, all of Chapter 4 on *Sampling*.

A lab quiz is planned for the lab on 13-May; and also for 20-May.

⇒ The five(5) **STARRED** problems will have to be turned in for grading.

Next week a solution will be posted. Some similar problems solutions can be found on the CD-ROM, especially the “unstarred” problems.

PROBLEM 4.1*:

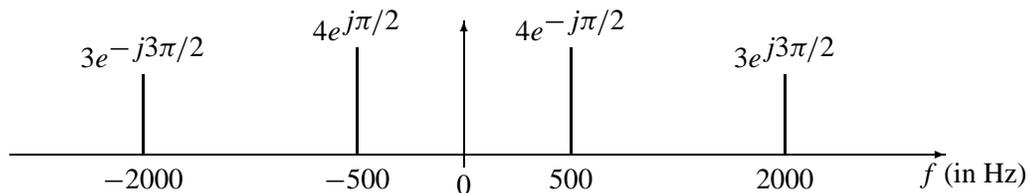


- (a) Suppose that the discrete-time signal $x[n]$ is given by the formula

$$x[n] = 10 \cos(0.20\pi n - \pi/3)$$

If the sampling rate of the C-to-D converter is $f_{si} = 1500$ samples/second, many *different* continuous-time signals $x(t) = x_\ell(t)$ could have been inputs to the above system. Determine two such inputs with frequency less than 1500 Hz; i.e., find $x_1(t)$ and $x_2(t)$ such that $x[n] = x_1(nT_{si}) = x_2(nT_{si})$ if $T_{si} = 1/1500$.

- (b) If the input $x(t)$ is given by the two-sided spectrum representation shown below,



Determine the spectrum for $x[n]$ when $f_{si} = 1500$ samples/sec. Make a plot for your answer, but label the frequency, amplitude and phase of each spectral component.

- (c) Using the discrete-time spectrum from part (b), determine the analog frequency components in the output $y(t)$ when the sampling rate of the D-to-C converter is $f_{so} = 900$ Hz. In other words, the sampling rates of the two converters are different.

PROBLEM 4.2*:

Suppose that a MATLAB function has been written to calculate a sum of discrete-time sinusoids, e.g., something similar to the `makecos()` that was written for the lab. Here is the actual function:

```
function xn = makedcos(omegahat,ZZ,Length)
%MAKEDCOS make a discrete-time sinusoid for x[n]
%
xn = real( exp( j*(0:Length-1)*omegahat(:) ) * ZZ(:) );
```

If the following MATLAB command is used to make an output sound:

```
soundsc( makedcos(pi*(0.5:0.4:1.5),[-2i,1i,3-4i],1000000), 1000 )
```

- Draw a plot of the discrete-time spectrum (vs. $\hat{\omega}$) of the discrete-time signal defined by this MATLAB operation. Make sure that you include all the spectrum components in the $-\pi$ to $+\pi$ interval.
- Draw a plot of the continuous-time spectrum (vs. f in Hz) of the analog output signal defined by the `soundsc()` function.

PROBLEM 4.3*:

A linear-FM “chirp” signal is one that sweeps in frequency from $\omega_1 = 2\pi f_1$ to $\omega_2 = 2\pi f_2$ as time goes from $t = 0$ to $t = T_2$. We can define the *instantaneous frequency* of the chirp as the derivative of the phase of the sinusoid:

$$x(t) = A \cos(\alpha t^2 + \beta t + \phi) \quad (1)$$

where the cosine function operates on a time-varying argument

$$\psi(t) = \alpha t^2 + \beta t + \phi$$

The derivative of the argument $\psi(t)$ is the *instantaneous frequency* which is also the audible frequency heard from the chirp *if the chirping frequency does not change too rapidly*.

$$\omega_i(t) = \frac{d}{dt} \psi(t) \quad \text{radians/sec} \quad (2)$$

There are examples on the CD-ROM in the Chapter 3 demos.

- For the linear-FM “chirp” in (1), determine formulas for the beginning instantaneous frequency (ω_1) and the ending instantaneous frequency (ω_2) in terms of α , β and T_2 . For this problem, assume that the starting time of the “chirp” is $t = 0$.
- For the “chirp” signal

$$x(t) = \Re \left\{ e^{j2\pi(29t^2 - 100t)} \right\}$$

derive a formula for the *instantaneous frequency* versus time. Should your answer for the frequency be a positive number?

- For the signal in part (b), make a plot of the *instantaneous frequency* (in Hz) versus time over the range $0 \leq t \leq 1$ sec.

PROBLEM 4.4*:

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.

- Assume that the disk is rotating in the counter-clockwise direction at a constant speed of 600 rpm (revolutions per minute). Express the movement of the spot on the disk as a rotating complex phasor.
- 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.
- If the flashing rate is 9 times per second, explain how the spot will move and write a complex phasor that gives the position of the spot at each flash.
- Draw a spectrum plot of the discrete-time signal in part (c) to explain your answer.

PROBLEM 4.5*:

A linear time-invariant system (FIR Filter) is described by the difference equation: $y[n] = \sum_{k=0}^3 x[n-k]$

The input to this system is a *finite-length* complex exponential signal:

$$x[n] = e^{j\pi n} \quad 0 \leq n \leq 5$$

- Make a plot of $x[n]$ vs. n .
- Compute $y[n]$, over the a range of n that includes all of its non-zero values.

PROBLEM 4.6:

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 $T_s = 0.001 = 10^{-3}$ second. The input sequence is given by the formula

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

- Plot $y[n]$ versus n .
- For the pulse shape

$$p(t) = \begin{cases} 1 & -0.0005 \leq t \leq 0.0005 \\ 0 & \text{otherwise} \end{cases}$$

carefully sketch the output waveform $y(t)$ over its non-zero region.

- For the pulse shape

$$p(t) = \begin{cases} 1 - 1000|t| & -0.001 \leq t \leq 0.001 \\ 0 & \text{otherwise} \end{cases}$$

carefully sketch the output waveform $y(t)$ over its non-zero region.