

PROBLEM FALL-04-F.1:

In each of the following cases, *simplify the expression as much as possible.*

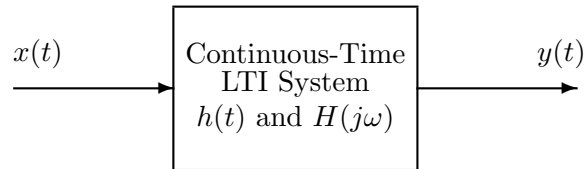
(a) $e^{-2t}\delta(t+3) =$

(b) $u(t-3) * \delta(t-2) =$

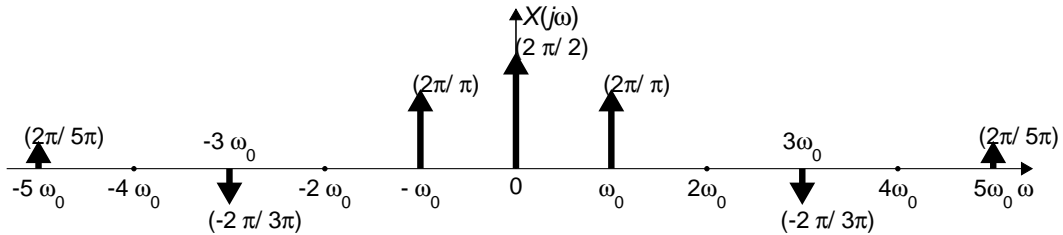
(c) $\int_{-10}^5 \cos(10\tau)\delta(\tau-3)d\tau =$

(d) $x[n] = 5\sqrt{2}\cos(0.3\pi n + \pi/4) + 5\cos(0.3\pi n + \pi) =$

PROBLEM FALL-04-F.2:



The periodic input to the above LTI system has the Fourier transform $X(j\omega)$ drawn below:



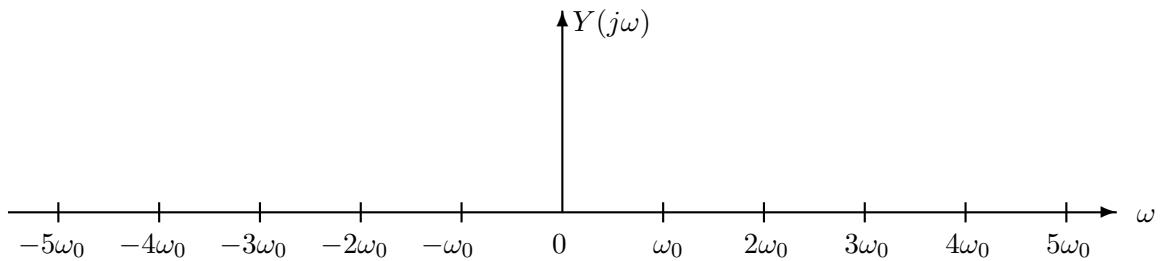
where the dark arrows denote impulses.

- (a) If the frequency response of the filter is given by

$$H(j\omega) = \begin{cases} e^{-j\omega} & \omega_0/2 < |\omega| < 3\omega_0/2 \\ 0 & \text{otherwise} \end{cases}$$

determine $y(t)$. Your answer should be written as a real time function, i.e., there should be no j 's in your answer.

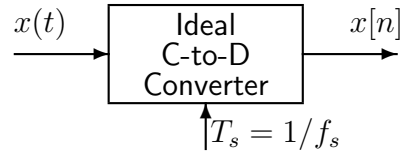
- (b) If $y(t) = x(t) - \frac{1}{2}$ sketch $Y(j\omega)$, the spectrum of $y(t)$, on the axes below.



PROBLEM FALL-04-F.3:

The individual parts of this problem are independent.

- (a) If the output from an ideal C/D converter is $x[n] = 33 \cos(0.5\pi n)$, and the sampling rate is 2000 samples/sec, then determine two possible positive values of the input frequency of $x(t)$ that are less than 2000Hz.:



ANS 1: =

ANS 2: =

- (b) Suppose that a student writes the following MATLAB code to generate a sine wave:

```
nn = 0:44100;  
xx = (3/pi) * cos(pi*1.25*nn + pi/3);  
soundsc(xx,fsamp)
```

Although the sinusoid was not written to have a frequency of 2300 Hz, it is possible to play out the vector **xx** so that it sounds like a 2300 Hz tone. Determine the value of **fsamp** that should be used to play the vector **xx** as a 2300 Hz tone. Write your answer as an integer.

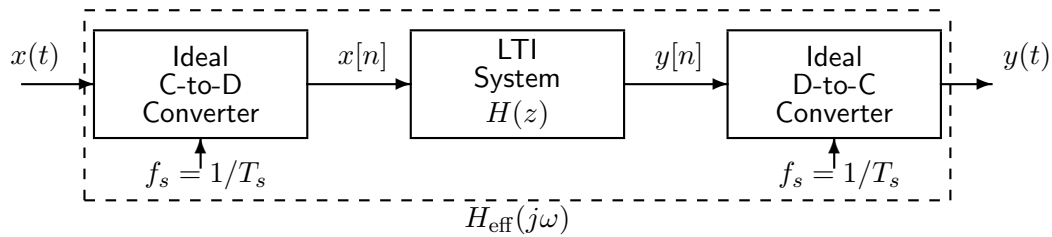
fsamp = Hz

- (c) Determine the Nyquist rate for sampling the signal $x(t)$ defined by:
 $x(t) = \Im\{e^{-j1200\pi t} + e^{j2000\pi t}\}.$

ANS = samples/sec.

PROBLEM FALL-04-F.4:

Consider the following system for discrete-time filtering of a continuous-time signal:

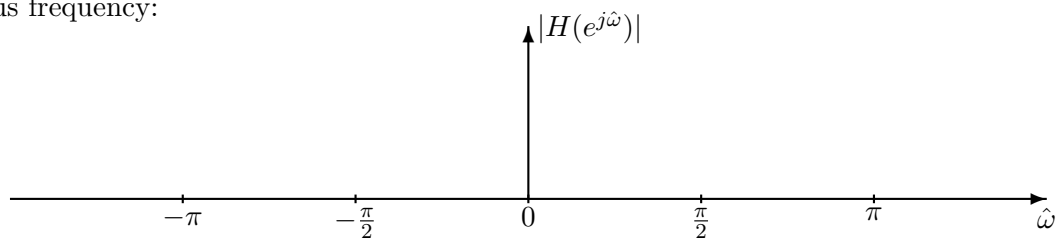


Assume that the discrete-time system is implemented using the MATLAB command:

```
yy=firfilt([0,1,0,-1],xx)
```

where **xx** is an array of samples of $x[n]$ and **yy** holds samples of $y[n]$.

- (a) Determine the frequency response of the discrete-time filter and plot its magnitude response versus frequency:



- (b) Assume that the input signal $x(t)$ is a sum of cosines:

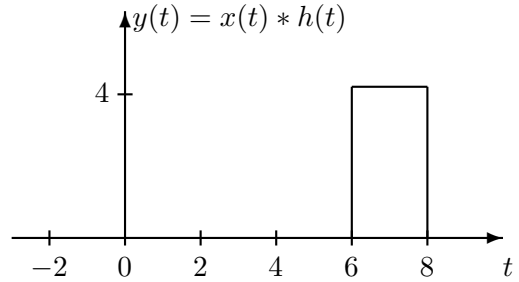
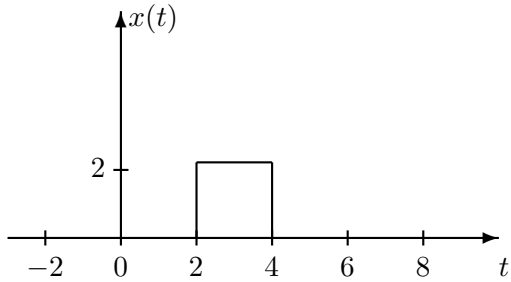
$$x(t) = 8 \cos(960\pi t + \pi/3) + 5 \cos(600\pi t - \pi/4)$$

For this input signal, determine the output signal $y(t)$ when the sampling rate is $f_s = 480$ **samples/sec**. Your answer should be expressed as a sum of cosines.

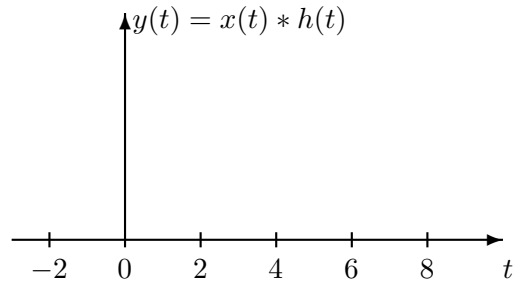
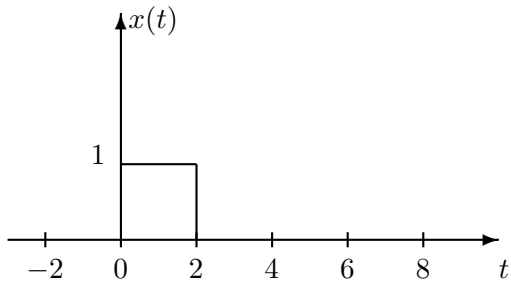
PROBLEM FALL-04-F.5:

The parts of this problem are completely independent.

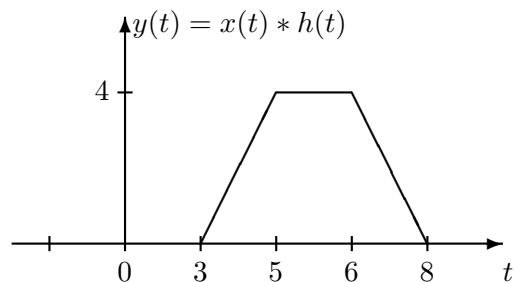
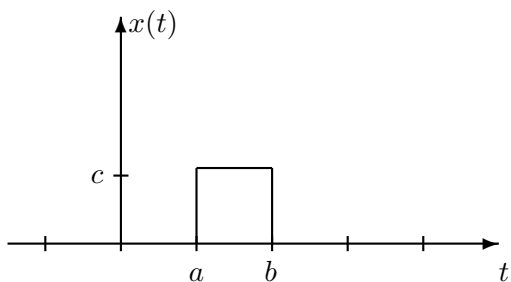
(a) Given that $y(t) = x(t) * h(t)$, find $h(t)$. $h(t) =$



(b) If $h(t) = u(t)$, plot $y(t) = x(t) * h(t)$ on the graph on the right. *Be sure to label the y(t) axis.*



(c) If $h(t) = u(t - 1) - u(t - 3)$ and $y(t) = x(t) * h(t)$, determine the values of a , b , and c in the graph of $x(t)$ on the left, if $y(t)$ is given by the graph on the right.



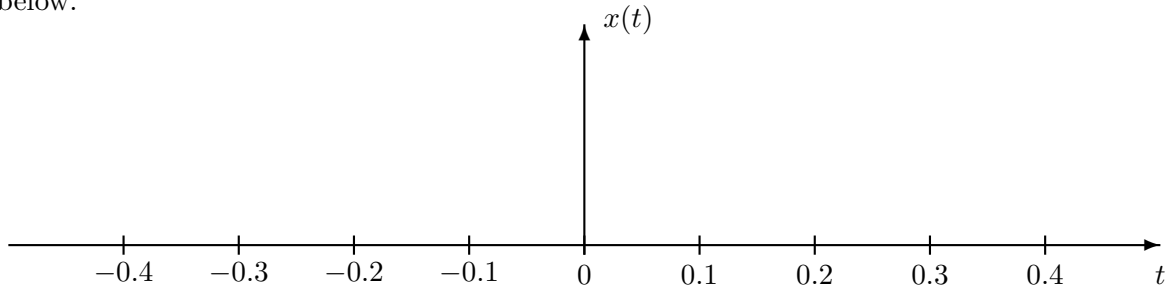
$a =$

$b =$

$c =$

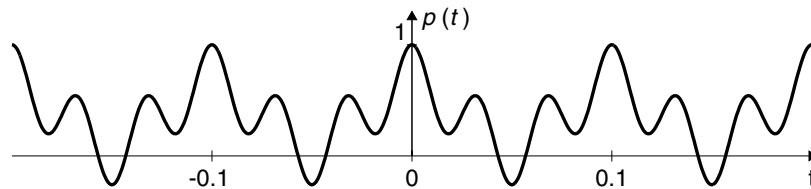
PROBLEM FALL-04-F.6:

- (a) Consider the signal $x(t) = \frac{\sin(5\pi t)}{2\pi t}$. Make a carefully labeled sketch of $x(t)$ in the space below.



- (b) Determine the Fourier transform of $y(t) = x(0.2 - t)$, using $x(t)$ from part (a).

- (c) Now consider the periodic signal $p(t)$ plotted below:

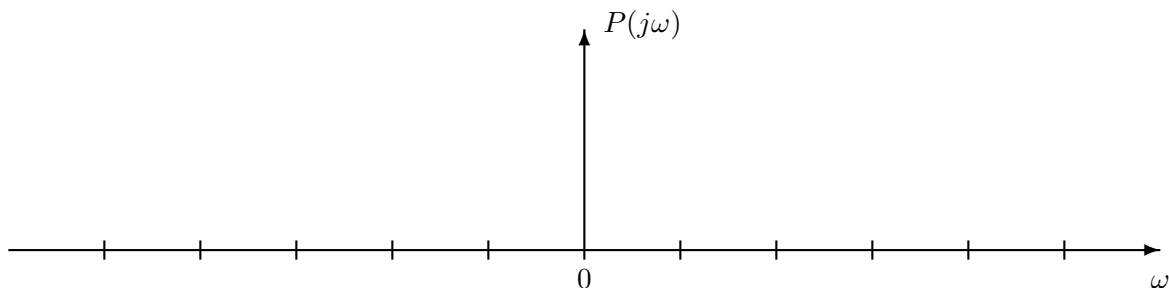


The Fourier series for this input can be simplified to the following form:

$$p(t) = \frac{1}{2} + \frac{2}{\pi} \cos(\omega_0 t) + \frac{2}{\pi} \cos(3\omega_0 t)$$

$\omega_0 =$ _____ rad/sec

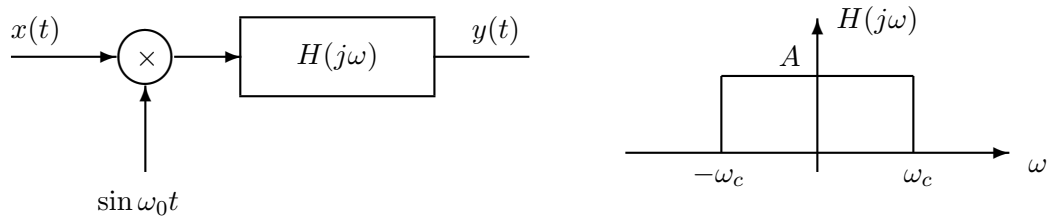
First determine the value of ω_0 and put your result in the box. Then, **either** write an equation for $P(j\omega)$, the Fourier transform of $p(t)$, in the space below, **or** plot it on the axes below. **You must label your plot carefully to receive full credit.**



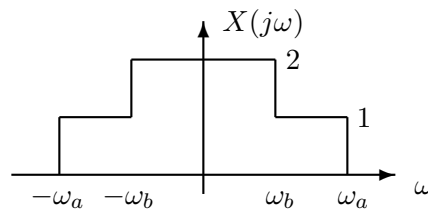
PROBLEM FALL-04-F.7:

The two parts of this problem are independent.

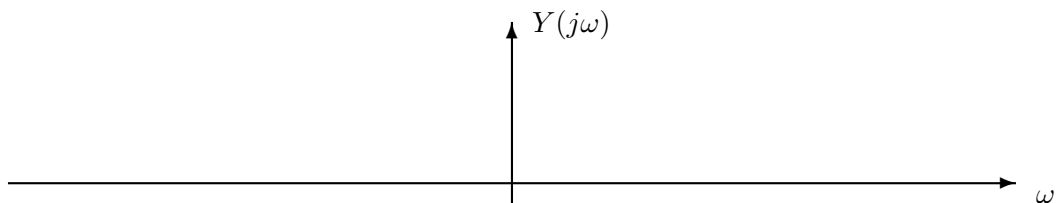
- (a) The system below is proposed as an alternative speech scrambler to the one in lab. Notice that the carrier signal is a sine instead of a cosine.



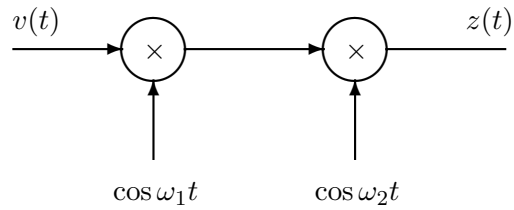
Assume that $x(t)$ has the spectrum, $X(j\omega)$ shown below



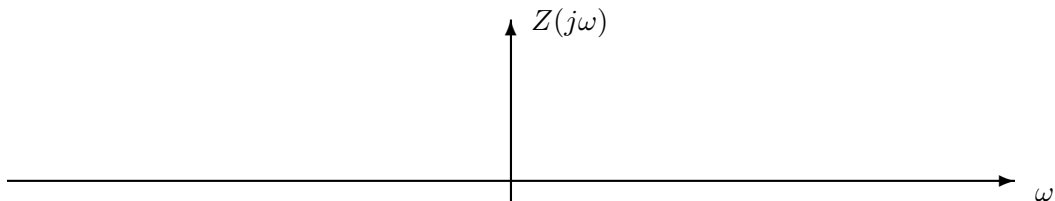
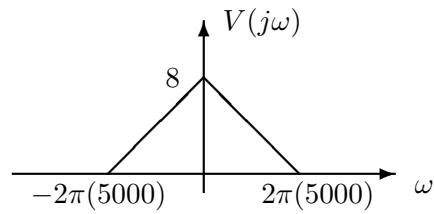
Sketch the spectrum, $Y(j\omega)$ of the output of the scrambler on the axes below if $\omega_a/(2\pi) = 5$ kHz, $\omega_b/(2\pi) = 2$ kHz, $\omega_c/(2\pi) = 10$ kHz, and $\omega_0/(2\pi) = 10$ kHz. Be sure to LABEL YOUR PLOT.



- (b) Signals are often repeatedly moved from one portion of the spectrum to another by repeated mixing. This process is called **heterodyning**. A simple example is the cascade of two mixers shown below.



Let $f_1 = \omega_1/(2\pi) = 45$ kHz and $f_2 = \omega_2/(2\pi) = 10$ kHz. Sketch the spectrum $Z(j\omega)$ assuming that $V(j\omega)$ has the shape shown in the figure below.



PROBLEM FALL-04-F.8:

A discrete-time system is defined by the following system function:

$$H(z) = H(z) = \frac{0.49 - z^{-2}}{1 + 0.49z^{-2}}.$$

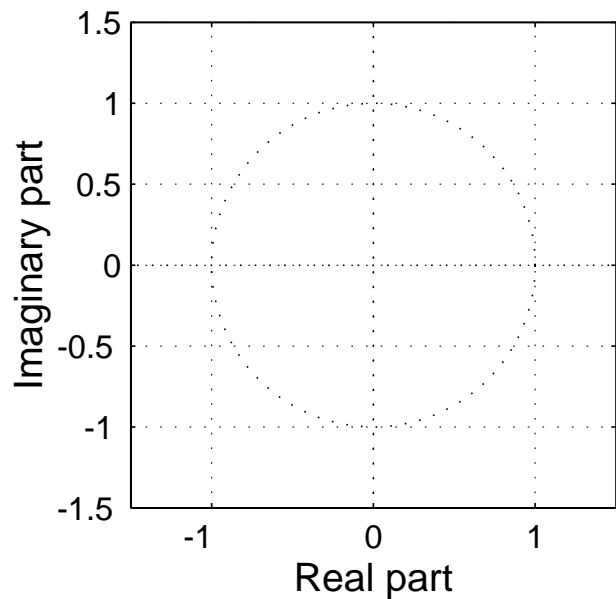
- (a) Write down the difference equation that is satisfied by the input $x[n]$ and output $y[n]$ of the system.

- (b) Fill in numbers for the vectors **bb** and **aa** in the following MATLAB computation of the frequency response of the system:

```
bb=[          ]; aa=[          ];  
yy=filter(bb,aa,xx)
```

where **xx** is the input signal to be filtered.

- (c) Determine *all* the poles and zeros of $H(z)$ and plot them in the z -plane.



- (d) Make a sketch of the magnitude of the frequency response of the system over the range $-\pi < \hat{\omega} \leq \pi$. Indicate where the peaks and valleys are located, and also determine the height of the peaks and the valleys.