

# 6.003 (Fall 2011)

## Final Examination

*December 19, 2011*

**Name:**

**Kerberos Username:**

**Please circle your section number:**

<i>Section</i>	<i>Time</i>
2	11 am
3	1 pm
4	2 pm

**Grades will be determined by the correctness of your answers (explanations are not required).**

**Partial credit will be given for ANSWERS that demonstrate some but not all of the important conceptual issues.**

You have **three hours**.

Please put your initials on all subsequent sheets.

Enter your answers in the boxes.

This quiz is closed book, but you may use four  $8.5 \times 11$  sheets of paper (eight sides total).

No calculators, computers, cell phones, music players, or other aids.

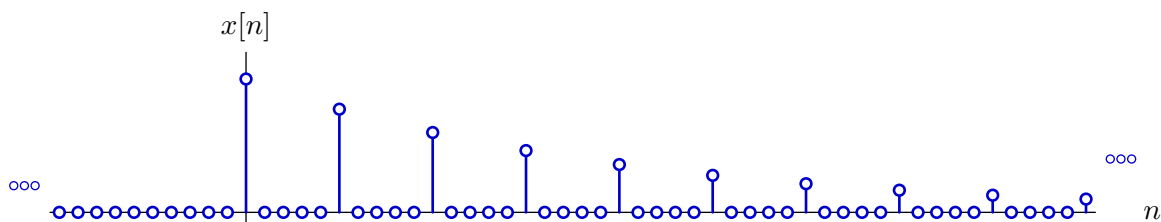
1	/14
2	/14
3	/12
4	/16
5	/16
6	/14
7	/14
Total	/100

**1. Z Transform** [14 points]

Determine  $X(z)$ , the Z transform of  $x[n]$ , where

$$x[n] = \sum_{k=0}^{\infty} a^k \delta[n - 5k] = \delta[n] + a\delta[n - 5] + a^2\delta[n - 10] + a^3\delta[n - 15] + \dots$$

is plotted below.



Enter a closed-form expression for  $X(z)$  in the box below.

$$X(z) = \boxed{\frac{z^5}{z^5 - a}}$$

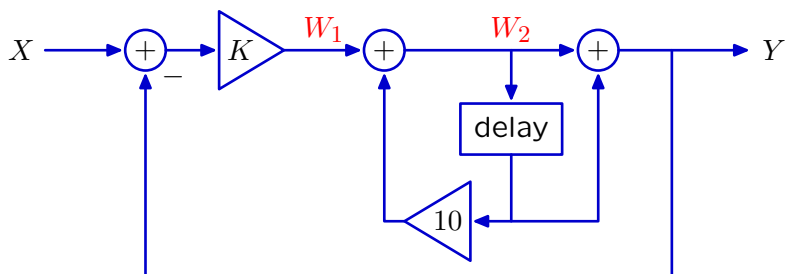
$$X(z) = \sum_{n=-\infty}^{\infty} x[n]z^{-n} = \sum_{n=-\infty}^{\infty} \sum_{k=0}^{\infty} a^k \delta[n - 5k]$$

The  $\delta[n - 5k]$  function is 1 if  $n = 5k$  and zero otherwise.

$$X(z) = \sum_{k=0}^{\infty} a^k z^{-5k} = \frac{1}{1 - az^{-5}} = \frac{z^5}{z^5 - a}$$

## 2. DT Stability [14 points]

Determine the range of  $K$  for which the following discrete-time system is stable (and causal).



range of  $K$ :

$$K > 4.5$$

$$W_2 = W_1 + 10\mathcal{R}W_2$$

$$Y = W_2 + \mathcal{R}W_2$$

$$\frac{Y}{W_1} = \frac{1 + \mathcal{R}}{1 - 10\mathcal{R}}$$

$$\frac{Y}{X} = \frac{K \frac{1+\mathcal{R}}{1-10\mathcal{R}}}{1 + K \frac{1+\mathcal{R}}{1-10\mathcal{R}}} = \frac{K(1+\mathcal{R})}{1 - 10\mathcal{R} + K + K\mathcal{R}} = \frac{K(z+1)}{z - 10 + Kz + K}$$

For stability, the pole must be inside the unit circle.

$$-1 < z = \frac{10 - K}{1 + K} < 1$$

$$K > 4.5$$

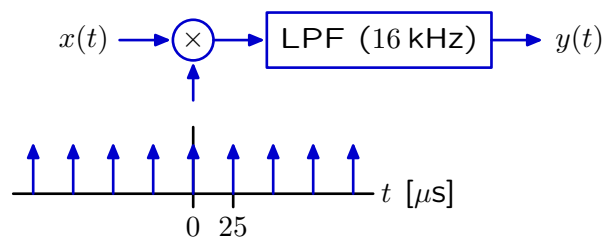
### 3. Harmonic Aliasing [12 points]

Let  $x(t)$  represent a periodic signal with the following harmonics:

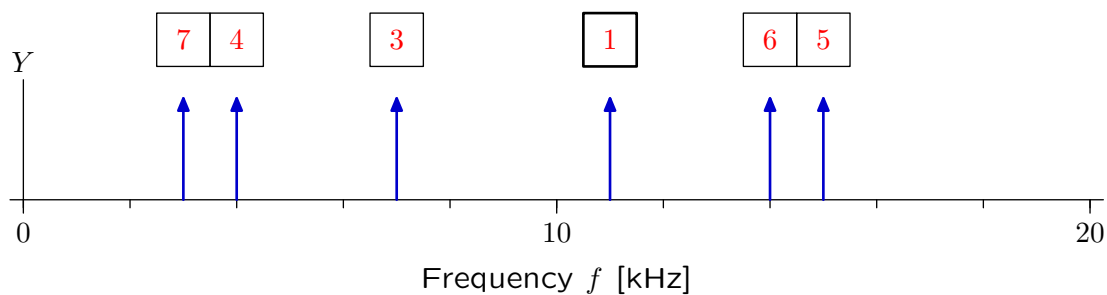
harmonic number	frequency [kHz]
1	11
2	22
3	33
4	44
5	55
6	66
7	77

Throughout this problem, frequencies ( $f$ ) are expressed in cycles per second (Hz), which are related to corresponding radian frequencies ( $\omega$ ) by  $f = \frac{\omega}{2\pi}$ .

The signal  $x(t)$  is multiplied by an infinite train of impulses separated by  $25 \times 10^{-6}$  seconds, and the result is passed through an ideal lowpass filter with a cutoff frequency of 16 kHz.

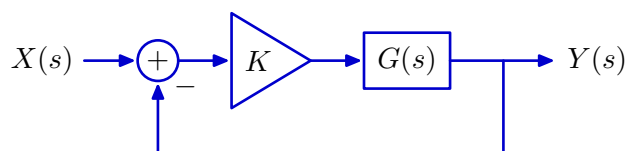


The plot below shows the Fourier transform  $Y$  of the output signal, for frequencies between 0 and 20 kHz. Write the number of the harmonic of  $x(t)$  that produced each component of  $Y$  in the box above that component. If none of 1-7 could have produced this frequency, enter **X**.

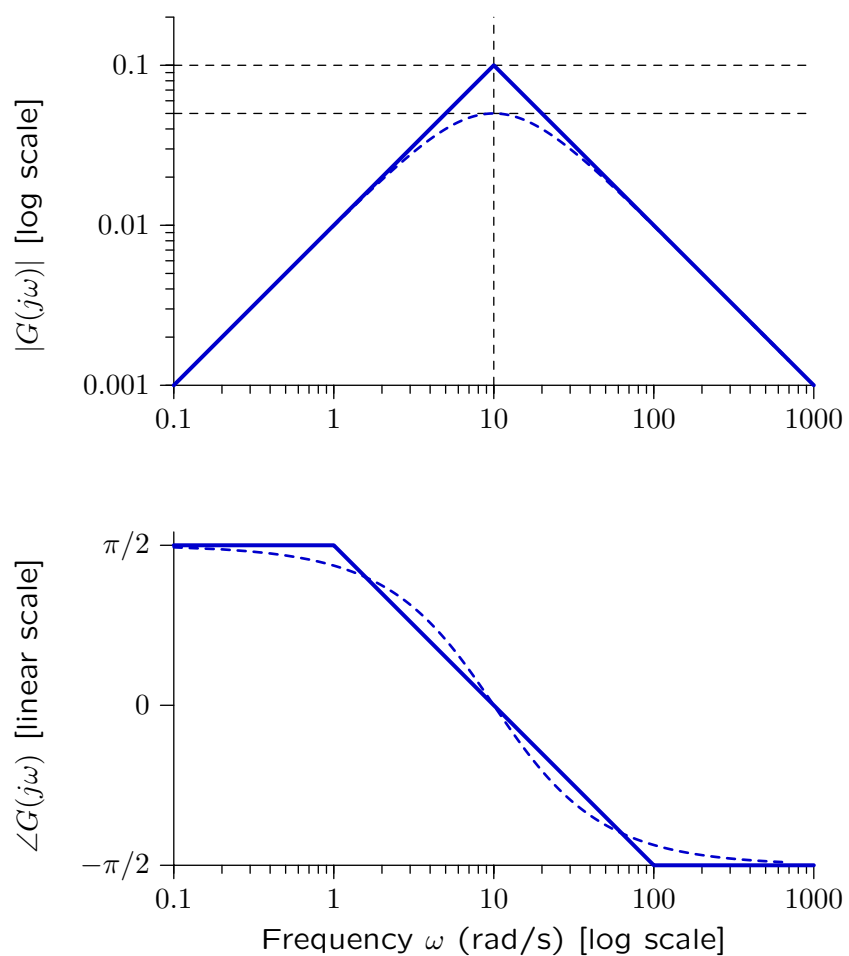


#### 4. Feedback [16 points]

Let  $H(s) = \frac{Y(s)}{X(s)}$  represent the system function of the following feedback system



where  $G(s)$  represents a linear, time-invariant system. The frequency response of  $G(s)$  is given by the following Bode plots (magnitude and frequency plotted on log scales).



**Part a.** Determine a closed-form expression for  $g(t)$ , the impulse response of  $G(s)$ .

$g(t) =$

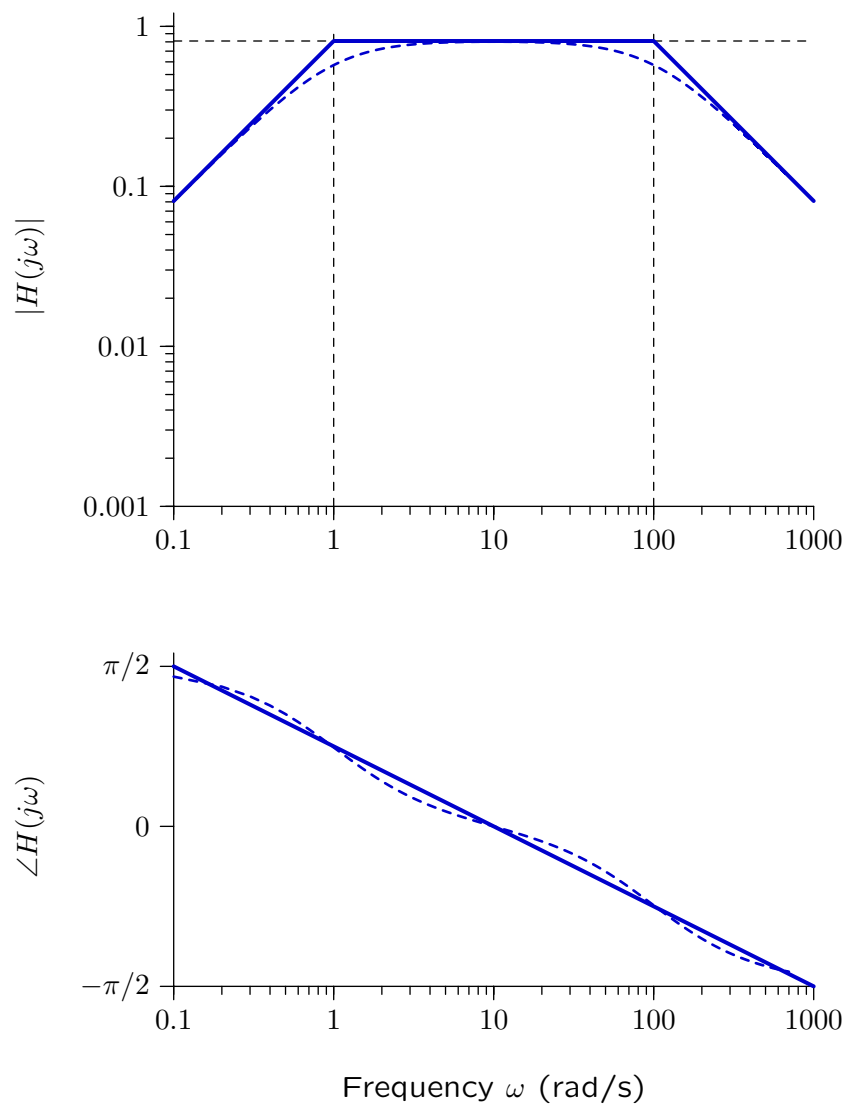
$$(1 - 10t)e^{-10t} u(t)$$

$$G(s) = \frac{s}{(s+10)^2} = \frac{1}{s+10} - \frac{10}{(s+10)^2}$$

$$g(t) = (1 - 10t)e^{-10t} u(t)$$

**Part b.** Sketch straight-line approximations (Bode plots) for the magnitude (log scale) and angle (linear scale) of  $H(j\omega)$  when  $K = 81$ .

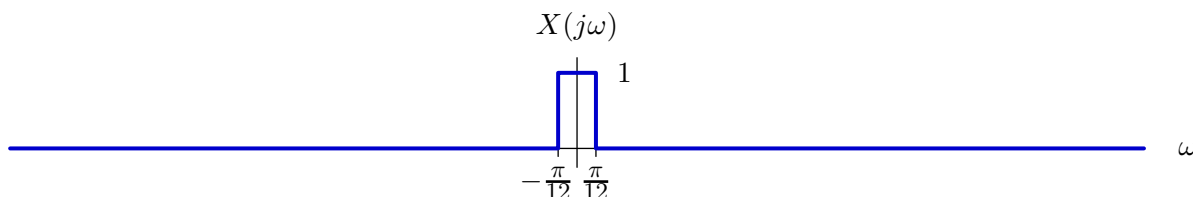
**Clearly label all important magnitudes, angles, and frequencies.**



$$\begin{aligned}
 H(s) &= \frac{KG(s)}{1 + KG(s)} = \frac{K \frac{s}{(s+10)^2}}{1 + K \frac{s}{(s+10)^2}} \\
 &= \frac{Ks}{s^2 + 20s + 100 + Ks} = \frac{81s}{s^2 + 101s + 100} = \frac{81s}{(s+1)(s+100)}
 \end{aligned}$$

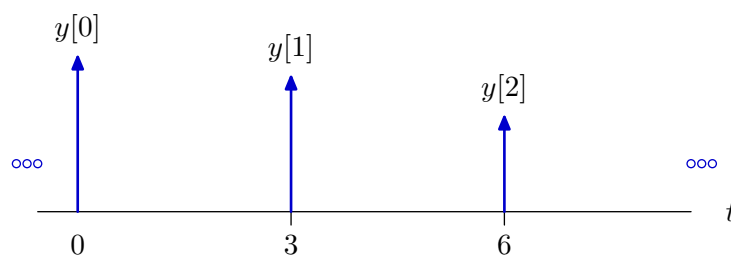
### 5. Triple reconstruction [16 points]

A CT signal  $x(t)$  is sampled to produce a DT signal  $y[n] = x(3n)$ . The Fourier transform of  $x(t)$  is given below.



We wish to compare two methods of using  $y[n]$  to reconstruct approximations to  $x(t)$ .

**Part a.** Let  $w_1(t)$  represent a signal in which each sample of  $y[n]$  is replaced by an impulse of area  $y[n]$  located at  $t = 3n$ . Thus  $w_1(t)$  has the following form

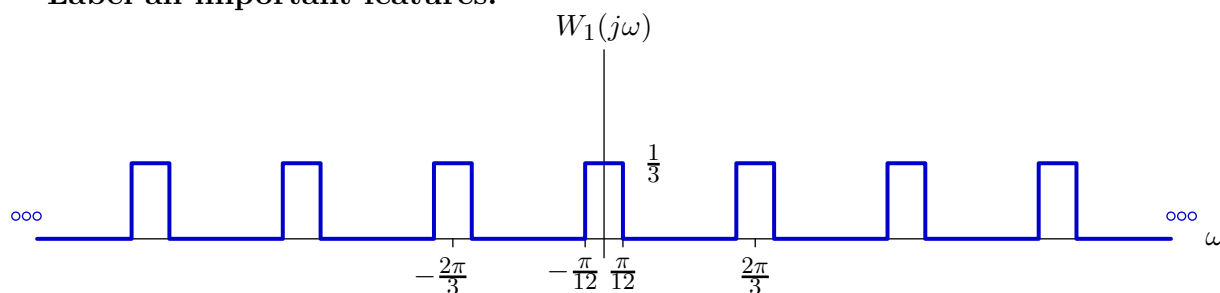


which can be represented mathematically as

$$w_1(t) = \sum_{n=-\infty}^{\infty} y[n] \delta(t - 3n).$$

Sketch the Fourier transform of  $w_1(t)$  on the axes below.

**Label all important features.**



Since  $y[n] = x(3n)$ ,  $w_1(t)$  is equal to  $x(t)$  times an infinite train of unit impulses separated by  $T = 3$  seconds:

$$w_1(t) = \sum_{n=-\infty}^{\infty} y[n] \delta(t - 3n) = \sum_{n=-\infty}^{\infty} x(3n) \delta(t - 3n) = \sum_{n=-\infty}^{\infty} x(t) \delta(t - 3n)$$

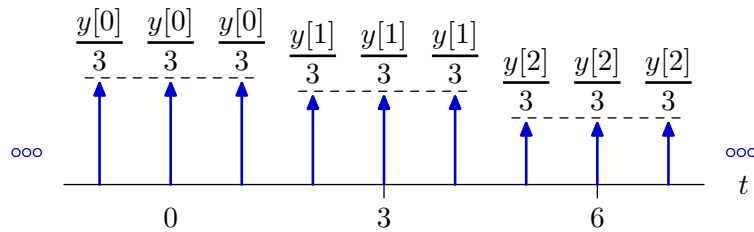
where the last step follows from the fact that  $\delta(t - 3n)$  is zero except at  $t = 3n$ . Multiplication in time by an impulse train corresponds to convolution in frequency

$$W_1(j\omega) = X(j\omega) * P(j\omega)$$

where  $P(j\omega)$  represents an infinite train of impulses, each of weight  $2\pi/3$ , and separated by  $\omega = 2\pi/3$ . (i.e., the Fourier transform of an infinite train of impulses spaced at  $T = 3$  seconds). Thus the frequency content of  $X(j\omega)$  is periodically replicated in frequency, spaced at  $\Delta\omega = 2\pi/3$ .



**Part b.** Let  $w_2(t)$  represent a signal in which each sample of  $y[n]$  is replaced by three impulses (one at  $t = 3n - 1$ , one at  $t = 3n$ , and one at  $t = 3n + 1$ ), each with area  $y[n]/3$ . Thus  $w_2(t)$  has the following form

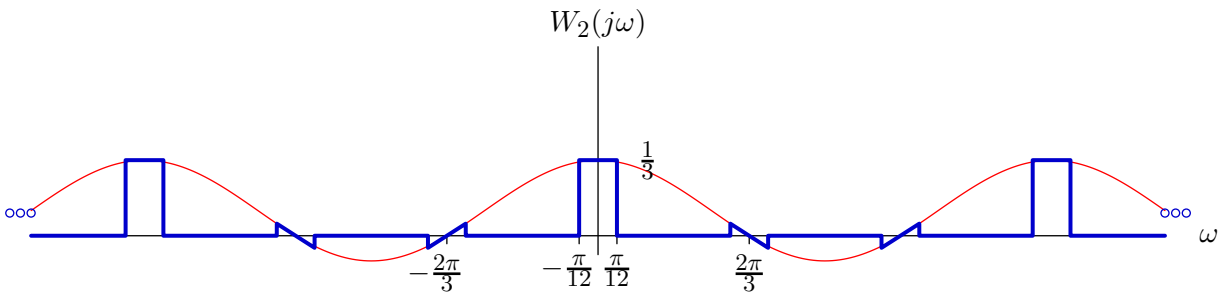


which can be represented mathematically as

$$w_2(t) = \frac{1}{3} \sum_{n=-\infty}^{\infty} y[n] \left( \delta(t-3n-1) + \delta(t-3n) + \delta(t-3n+1) \right).$$

Sketch the Fourier transform of  $w_2(t)$  on the axes below.

**Label all important frequencies as well as the value of  $W_2(j\omega)$  at  $\omega = 0$ .**



The signal  $w_2(t)$  can be derived by convolving  $w_1(t)$  with a signal

$$s(t) = \frac{1}{3}\delta(t-1) + \frac{1}{3}\delta(t) + \frac{1}{3}\delta(t+1).$$

Thus  $W_2(j\omega)$  can be determined by multiplying  $W_1(j\omega)$  by the Fourier transform of  $s(t)$

$$S(j\omega) = \frac{1}{3}e^{-j\omega} + \frac{1}{3} + \frac{1}{3}e^{j\omega} = \frac{1}{3} + \frac{2}{3}\cos\omega$$

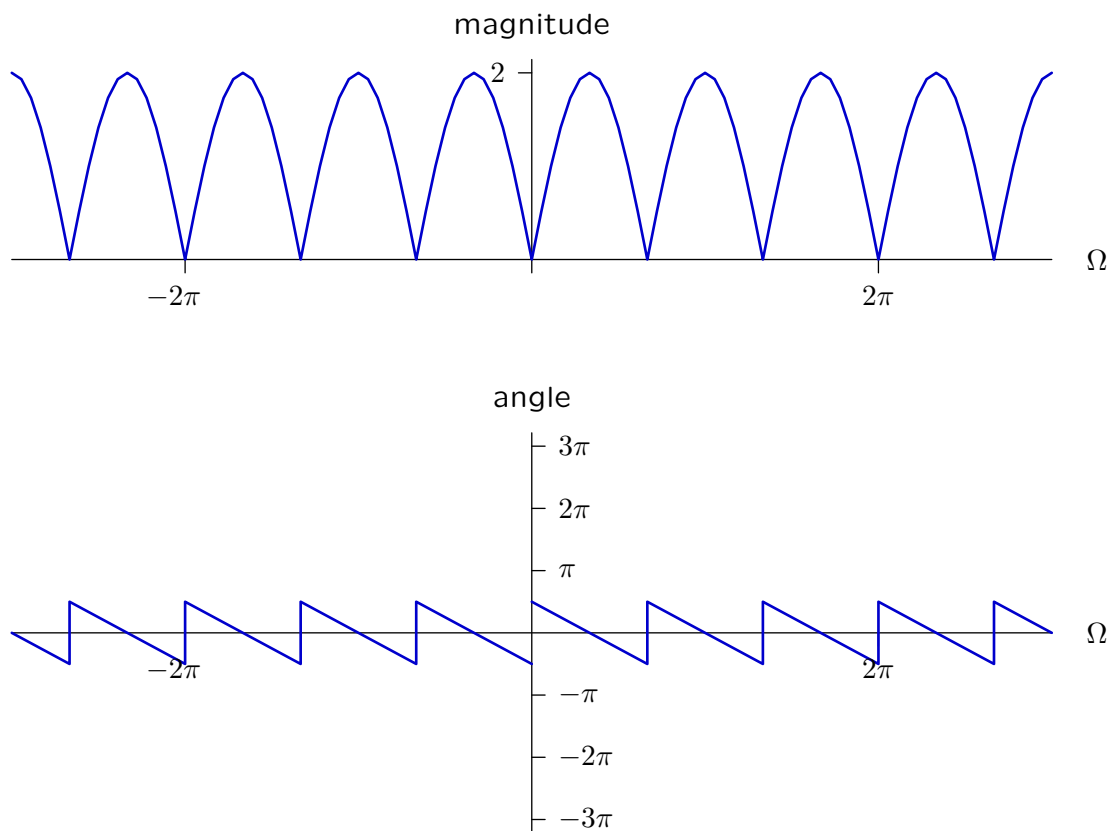
which is shown in red above.

## 6. DT Filtering [14 points]

Sketch the magnitude and angle of the frequency response of a linear, time-invariant system with the following unit-sample response:

$$h[n] = \delta[n] - \delta[n - 3].$$

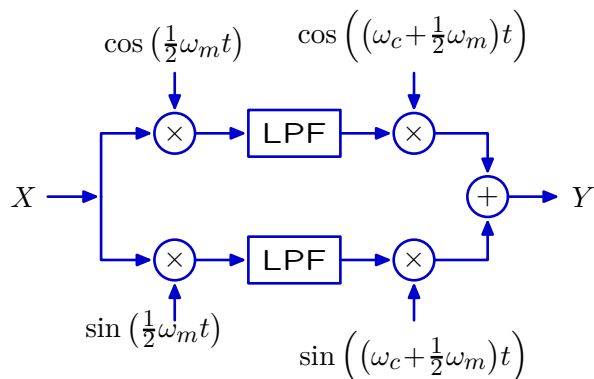
**Label all important magnitudes, angles, and frequencies.** All scales are linear.



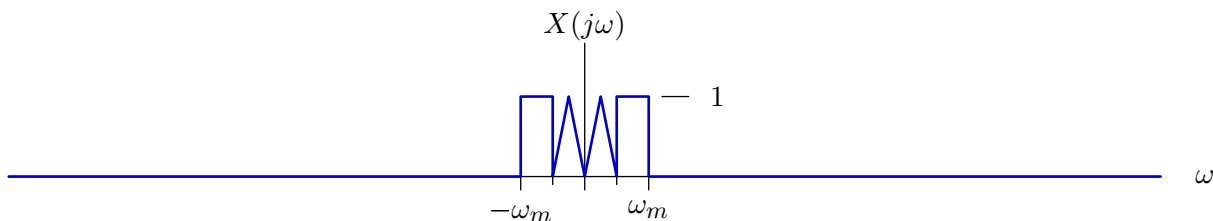
$$H(e^{j\Omega}) = 1 - e^{-j3\Omega} = e^{-j3\Omega/2} (e^{j3\Omega/2} - e^{-j3\Omega/2}) = j2e^{-j3\Omega/2} \sin \frac{3\Omega}{2}$$

### 7. Bandwidth Conservation [14 points]

Consider the following modulation scheme, where  $\omega_c \gg \omega_m$ .



Assume that each lowpass filter (LPF) is ideal, with cutoff frequency  $\omega_m/2$ . Also assume that the input signal has the following Fourier transform.



Sketch  $Y(j\omega)$  on the following axes.  
**Label all important magnitudes and frequencies.**

