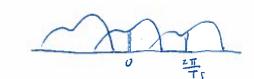
SIGSYS PSO8 - JENNIFER WEL

03262015

1. Consider signal x(t) band limited to WM



If 2 Wm > Ws =

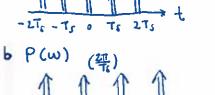
 $X(\omega)$

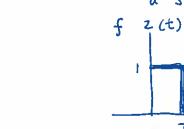
$$p(t) = \sum_{k=-\infty}^{\infty} S(t-kT_s)$$

HOW TO RECOVER X(t) from Xp(t):

xp(t) = x(t)p(t) a Sketch of $\chi_p(t)$ Limpuls &J (x(0))

To recover x(t) from xp(t), multiply the frequency by a box (a brownshippediens filter with frequency [-wm wm]) to isolate the middle frequency. In the time domain, this would mean convolving with a sinc function.

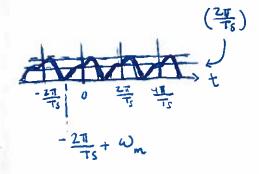


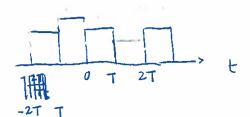




C Xp(w)

(see Q1a) g xz(t) = xp + z(t) zero-order hold reconstruction of xp(t)





d Relationship between Ts & Wm 2 Wm < 2T

$$h X_{z}(\omega) = X_{p}(\omega) Z(\omega)$$

Tr - Wm > Wm

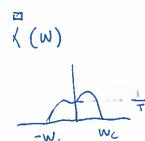
:. 2 wm < ws granntees that Xp (W) contains all the information present in $\mathbf{X}(\omega)$

$$\overline{X}(\omega) = X_{\epsilon}^{(\omega)} H(\omega)$$

$$\hat{X}(\omega) = X_{\epsilon}(\omega) H(\omega)$$

H(w) (with $w_c = \frac{\pi}{T_s}$)

χ(w)



-Wc Wc

Since We happens to encapsulate one frequency component, this is the demonstrating 10 to the whole isolate the middle frequency, idea

X(w) & X(w)

are different since

X(w) has a sinc function
applied & decays more
quickly (and is attenuated)
as a result.

(when $W_m = \overline{T}_s$, the ratio of $\overline{X}(W_m)$ & $\hat{X}(W_m)$ is M $\stackrel{\circ}{\circ}$ since both equal O at \overline{T}_s

SIGSYS PS 08 - JENNIFER WEI 2 y(t) = x,(t) cos (w,t) + $X_2(t) cos(w_2t)$ Suppose X1(W) =0, X2(W) =0 if |w| > wm ASSUME WI>> HE WM W2 >> Wm $X_1(w)$ $W_1 + 2W_m < W_2$ $\chi_{\ell}(\omega)$ FT {coswit} NOTE: The gap between WI & WZ should be larger since W, +2Wm < 100 W2 (sorry I have poor drawing SKILLS) bFT of ?y(t)cos(wit)} FT = Y(W) · FJ S (OS WY } +w, wiw with withwa -wz-w, -2w; -wz+w, w, w2m w2m w, FT of { y(t) cos (w2t)} (-w2-w1) (W2+W1) 0 $(-2W_2)$ $(-W_2)$ (W_2-W_1) (W_2+W_1) (WL+Wz)

from y(t), where y(t) is
the received AM signal
from two different AM
transmitters using frequency
W, & W2

To recover X,(t) & X2(t), multiply y(t) by cos (w,t) or cos(w2t) (respectively).

From there, like in the process described in 1e, filter with a cutoff @ [-Wm] and [Wm].

Since cos attenuates the signal, multiply by 2 to revert back.

ANOTHER WAY TO SOLVE WOULD BE TO WORK OFF OF THE FTS of 26.

TAKING THE FTS, APPLY A FILTER WITH FREQS [EWM, WM].

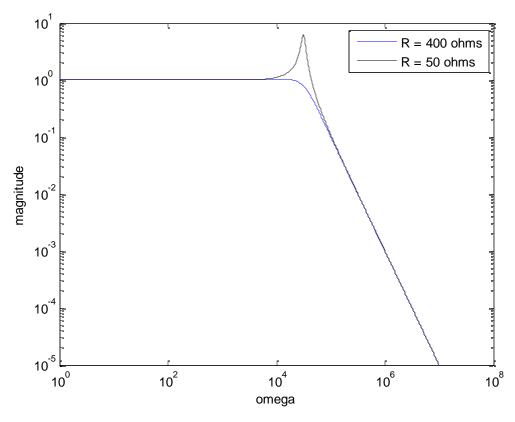
UNDO THE ATTEMENT OF THE SECOND OF THE SECON

UNDO THE ATTENVATION & use the inverse fourier to find X_1 (t) X_2 (t)

$$\begin{split} & = \frac{1}{\sqrt{1 + C_{W}}} \frac{1}{$$

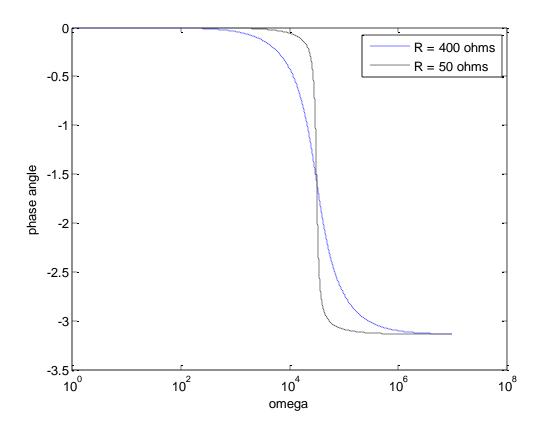
d used Wolfram Alpha

- \(\frac{1}{2}C(R+2|LW)\)\(\frac{(++Cw(Lw-\R))^2}{}



```
C = 10^-7; %F
L = 10^-2; %H
R1 = 400; %ohms
R2 = 50; %ohms

w = 1:1:10000000;
Mag1 = (1.0./( (w*R1*C).^2 + (1-(w.^2)*L*C).^2 ).^(0.5));
Mag2 = (1.0./( (w*R2*C).^2 + (1-(w.^2)*L*C).^2 ).^(0.5));
loglog(w,Mag1,'b',w,Mag2,'k');
legend('R = 400 ohms', 'R = 50 ohms');
xlabel('omega');
ylabel('magnitude');
```



```
figure;
Freq1 = 1.0./(1+(R1.*C.*j.*w)-((w.^2).*C.*L));
Freq2 = 1.0./(1+(R2.*C.*j.*w)-((w.^2).*C.*L));
phase1= angle(Freq1);
phase2 = angle(Freq2);
%phase1 = atan((j.*w.*R1.*C)./(-(w.^2).*L.*C).^2).*(360/2*pi);
%phase2 = atan((j.*w.*R2.*C)./(-(w.^2).*L.*C).^2).*(360/2*pi);
semilogx(w,phase1,'b',w,phase2,'k');
legend('R = 400 ohms', 'R = 50 ohms');
xlabel('omega');
ylabel('phase angle');
```