

EECS 361
Homework #11

1. Section 5.12 & 5.13 & 5.14 Participation Activities

- 5.12.1: AM and DSB Spectra, with envelope detection.
- 5.12.2: DSB and AM.
- 5.13.1: FDM total bandwidth.
- 5.13.2: AM radio station bandwidth.
- 5.14.2: Sampling theory.

2. (Concept: Properties of double sideband suppressed carrier (DSC-SC) signals.)

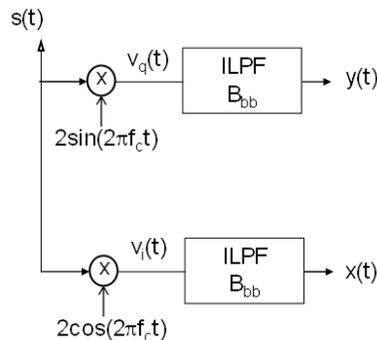
Plot the time domains signal and magnitude spectrum and find the average transmitter power, in a double sideband suppressed carrier (DSC-SC) signal, $x_{\text{DSB-SC}}(t)$ where $x_{\text{DSB-SC}}(t) = 4\cos(2\pi f_m t)\cos(2\pi f_c t)$, here $f_m = 100$ Hz and $f_c = 1$ kHz. Hint: Use trigonometry identities to expand signals into a sum of cosines

3. (Concept: Properties of double sideband large carrier (DSC-LC) signals.)

Plot the time domains signal and magnitude spectrum and find the average transmitter power, in a double sideband large carrier (DSC-LC) signal, $x_{\text{DSB-LC}}(t)$ where $x_{\text{DSB-LC}}(t) = 4(1+0.8\cos(2\pi f_m t))\cos(2\pi f_c t)$, here $f_m = 100$ Hz and $f_c = 1$ kHz. Hint: Use trigonometry identities to expand signals into a sum of cosines.

4. (Concept: Quadrature modulation)

Let $s(t) = g(t)\cos(2\pi f_c t) + z(t)\sin(2\pi f_c t)$, $g(t)$ and $z(t)$ are lowpass signals with a bandwidth of B_{bb} . The signal $s(t)$ is processed by the receiver shown below. The bandwidth of the ILPF is B_{bb} . Assume $f_c \gg B_{\text{bb}}$.



- a. Find the signal $v_i(t)$ in the time and frequency domains.

Hint: Use trigonometry identities to expand signals into a sum of sines/cosines.

- b. Find $x(t)$ and $y(t)$.

5. (Concept: The spectrum of a sampled signal is periodic)

The signal $x(t)$ has a Fourier transform of $X(f)$ and is band-limited to B_x Hz. The impulse sampling waveform is $\delta_{T_s}(t) = \sum_{k=-\infty}^{\infty} \delta(t - kT_s)$ where $T_s =$ sample spacing in time and $f_s = \frac{1}{T_s}$ is the sample rate. The sampled signal in the time domain is $x_s(t) = x(t)\delta_{T_s}(t)$. Show that the spectrum $X_s(f)$ of a impulse sampled signal $x_s(t)$ is periodic.

6. (Concept: The Nyquist sampling rate is twice the signal bandwidth)

- a. Let $y(t) = x(t)g(t)$ with $x(t) = \frac{30}{\pi}\text{sinc}(30t)$ and $g(t) = \frac{50}{\pi}\text{sinc}(50t)$, plot $y(t)$.
- b. Find the Nyquist sampling rate for $y(t)$ Hint: see Homework 9 Problem 5
- c. Make a stem plot of $y(kT_s)$ for k -5 to 5. Use the sampling interval $T_s = \frac{1}{26}$. Try to overlay the plot of $y(t)$ with stem plot of $y_s(kT_s)$

7. (Concept: Sample rate must be greater than 2B)

Let $x(t) = 24\text{sinc}(24\pi t)(1+2\cos(48\pi t))$

- a. Plot $x(t)$.
- b. Plot $X(f)$ (do the plot in f-Hz).
- c. What is the highest frequency in Hz present in $x(t)$?
- d. Sample $x(t)$ at the Nyquist rate and plot the spectrum of the sampled signal.
- e. What is the inverse Fourier Transform of the spectrum of the sampled signal?

f. The discrete time sampled signal $x(n\Delta t)=x[n]$ where $\Delta t = \frac{1}{\text{Nyquist rate}}$.

The first sample is at $t=0$.

What are the values of the first 5 samples, i.e, $x[n]$ for $n=0\dots 4$?

8. (Concept: Aliasing)

Exercise 5.16.1

9. (Concept: Aliasing)

A signal $x(t) = 500\text{sinc}^2(\pi 500t)$ is sampled at a rate of 900 samples/sec. What portion of the spectrum can be recovered from the samples? What is the Nyquist rate for $x(t)$? Hint [$B\text{sinc}^2(\pi B t) \longleftrightarrow \text{tri}(\frac{f}{B})$]

10. (Concept: Sampling a non-band-limited signal, aliasing, and the benefit of oversampling)

A high fidelity audio signal has an spectrum of $X(f) = e^{-\frac{|f|}{14430}}$.

a. Plot $X(f)$. On your plot indicate that the 6dB bandwidth of $X(f)$ which is about 10kHz

b. Assuming $X(f)$ is band-limited to its 6dB bandwidth of 10k Hz, what is the Nyquist sampling rate for this signal.

c. Using the sampling rate from part b, plot the spectrum, $X_s(f)$, of the sampled signal, that is, plot

$$\sum_{k=-\infty}^{\infty} X(f - kf_s) \quad \text{for } k = -4, -3, -2, -1, 0, 1, 2, 3, 4.$$

d. The reconstruction filter $H(f)$ is used to reconstruct the analog signal from the samples, $H(f) = \frac{1}{1 + (\frac{f}{35000})^2}$.

Plot $H(f)$.

e. Using the sampling rate from part b, the spectrum at the output of the reconstruction filter, is $Y(f) = X_s(f)H(f)$.

Define the reconstruction error as $|X(f) - Y(f)|$. Using the sampling rate from part b, plot the reconstruction error for $k = -4, -3, -2, -1, 0, 1, 2, 3, 4$

f. Using an oversampling ratio of 10, that is, use a sampling rate of ten times the sampling rate from part b, plot the reconstruction error.

g. Comment on the value of oversampling.