

Sample Solution to Problem Set 1

1. (10 points) Applying low-pass and bandpass filters to a digital signal

A square periodic signal is represented as the following sum of sinusoids:

$$s(t) = \frac{2}{\pi} \sum_{k=0}^{\infty} \frac{(-1)^k}{2k+1} \cos((2k+1)\pi t).$$

(Note that this is just a rewriting of the formula we discussed in class.)

- (a) Suppose that the signal is applied to an ideal low-pass filter with bandwidth 15 Hz. Plot the output from the low-pass filter and compare to the original signal. Repeat for 5 Hz; for 3 Hz. What happens as the bandwidth increases.
- (b) Suppose that the signal is applied to a bandpass filter that passes the frequencies from 5 to 9 Hz. Plot the output from the filter and compare to the original signal.

For your plots, use an appropriate plotting tool. One such tool is gnuplot, available in Unix.

Answer:

- (a) The non-zero frequency components of the signal $s(t)$ correspond to the frequencies 1/2 Hz, 3/2 Hz, 5/2 Hz, If the signal is passed through an ideal low-pass filter with bandwidth 15Hz, we obtain the output signal as:

$$\frac{2}{\pi} \sum_{k=0}^{14} \frac{(-1)^k}{2k+1} \cos((2k+1)\pi t).$$

Signals can be obtained for the other cases similarly. On plotting, we obtain Figures 1, 2, and 3, respectively.

- (b) See Figure 4.

2. (4 points) Bandwidth, signal element, and SNR

A digital signaling system is required to operate at 38.4 Kbps. If a signal element encodes a 8-bit word, what is the minimum required bandwidth of the channel. What signal-to-noise ratio is required to achieve the desired capacity on the bandwidth that you have computed?

Answer: We need capacity $C \geq 38400$ bps. We have $M = 2^8$. Nyquist's sampling theorem says that $C = 2B \log_2 M$. Therefore, $B \geq 38400/(16) = 2400$ Hz.

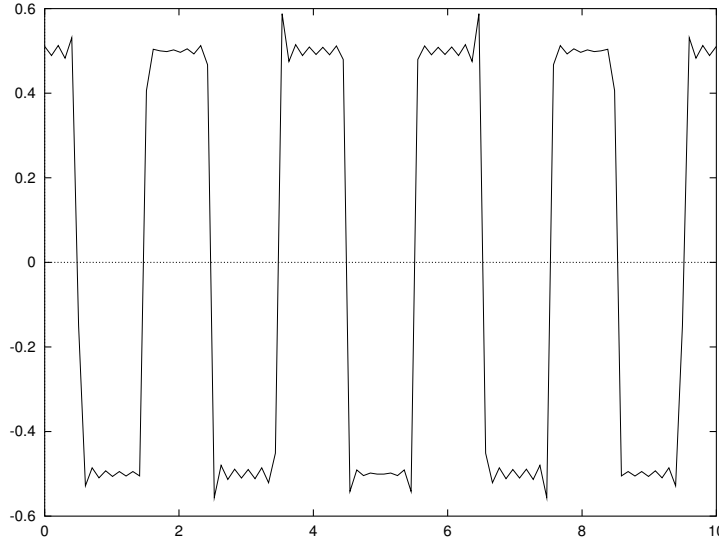


Figure 1: Passing through a low-pass filter of 15Hz bandwidth.

Using Shannon's capacity theorem, we have $C = B \log_2(1 + \text{SNR})$. Therefore $\text{SNR} \geq 2^{C/B} - 1 = 65535$. In decibels, we obtain $\text{SNR}_{dB} = 10 \cdot \log_{10}(65535) = 48.16$ dB.

3. (6 points) Effect of transmission frequency and distance on attenuation

Under the free-space path-loss model, find the transmit power required to obtain a received power of 1 dBm for a wireless system with isotropic antennas (gain = 1) and a carrier frequency $f = 5$ GHz, assuming a distance $d = 10$ m. Repeat for $d = 100$ m, keeping $f = 5$ GHz. Repeat for $f = 50$ GHz, keeping $d = 10$ m.

Answer: In free space, we have the path-loss model as:

$$\frac{P_t}{P_r} = \frac{(4\pi f d)^2}{(G_r G_t c^2)}.$$

Expressing the same in decibels and noting that the gains are 1, we obtain:

$$10 \log P_t = 10 \log[(4\pi f d)^2/c^2] + 10 \log P_r$$

For $d = 10$ m, $f = 5$ GHz:

$$\begin{aligned} 10 \log P_t &= 10 \log[(4\pi \cdot 5 \cdot 10^9 \cdot 10)^2/c^2] + 1 \text{ dBm} \\ &= 66.42 + 1 \text{ dBm} \\ &= 67.42 \text{ dBm} \end{aligned}$$

For $d = 100$ m $f = 5$ GHz:

$$\begin{aligned} 10 \log P_t &= 10 \log[(4\pi \cdot 5 \cdot 10^9 \cdot 100)^2/c^2] + 1 \text{ dBm} \\ &= 86.42 + 1 \text{ dBm} \\ &= 87.42 \text{ dBm.} \end{aligned}$$

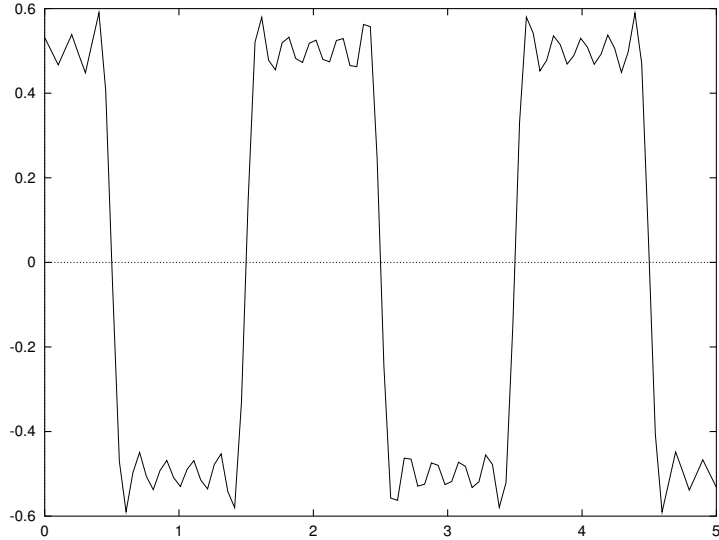


Figure 2: Passing through a low-pass filter of 5Hz bandwidth.

For $d = 10 \text{ m}$ $f = 50 \text{ GHz}$:

$$\begin{aligned}
 10 \log P_t &= 10 \log[(4\pi \cdot 50 \cdot 10^9 \cdot 10)^2 / c^2] + 1 \text{ dBm} \\
 &= 86.42 + 1 \text{ dBm} \\
 &= 87.42 \text{ dBm}
 \end{aligned}$$

4. (5 points) Received power using half-wave dipoles

Problem 5.9.

Answer: The relationship between transmitted power P_t and received power P_r is:

$$\frac{P_r}{P_t} = \frac{16\pi^2 d^2}{G_r G_t \lambda^2},$$

where d is the distance and G_r and G_t are the gains of the receiving and transmitting antennas, respectively. The two gains are given to be 3 dB, which is a factor of $10^{3/10} = 2$. The wavelength λ is given to be $3 \times 10^8 / 10^8 = 3 \text{ m}$. The distance is 10^4 m and $P_t = 1 \text{ W}$. Substituting these numbers we get:

$$P_r = \frac{2 \cdot 2 \cdot 3^2 \cdot 1}{16\pi^2 \cdot 10^8} = 2.28 \times 10^{-9} \text{ W}.$$

5. (7 points) Modulation schemes and E_b/N_0

Problem 6.2.

Answer: Let T_s be the signal element period, T_b be the bit period, and A be the amplitude of the signal.

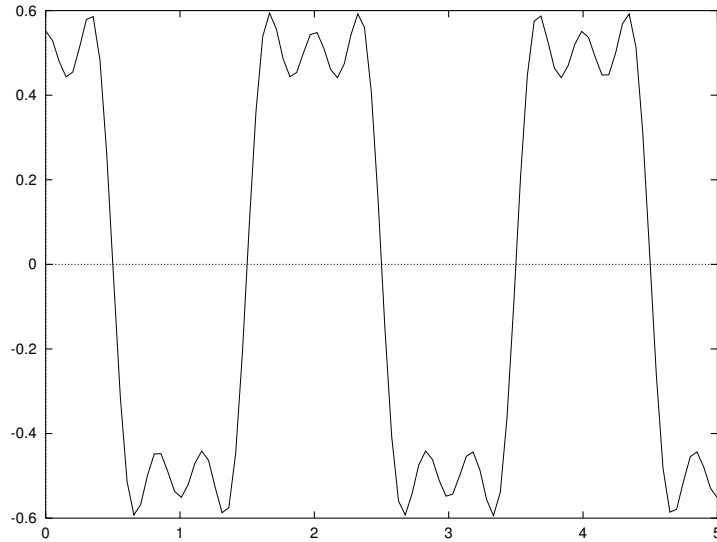


Figure 3: Passing through a low-pass filter of 3Hz bandwidth.

(a) We have $A = 0.005\text{V}$, $T_s = T_b = 10^{-5}\text{s}$. We obtain the power P as follows.

$$P = \frac{1}{T_s} \int_0^{T_s} s^2(t) dt = \frac{A^2}{2}$$

$$E_b = P \times T_b = P \times T_s = \frac{A^2}{2} T_s; N_0 = 2.5 \times 10^{-8} \times T_s$$

$$\frac{E_b}{N_0} = \frac{(A^2/2) \times T_s}{2.5 \times 10^{-8} \times T_s} = 500.$$

In dB, we obtain 27dB.

(b) For this, all the values are identical except that $T_b = T_s/2$. So we obtain $E_b/N_0 = 250$. In dB, this is 24 dB.

6. (5 points) Telephone channel bandwidth

Problem 6.6.

Answer: The required bandwidth for 2400 bps QPSK is $2R/2 = 2400$ Hz. The required bandwidth for 4800 bps MPSK with 8-level multisignaling is $2R/3 = 3200\text{Hz}$. Thus, bandwidth of the telephone channel is adequate for the former but not for the latter.

7. (5 points) Sampling

Problem 6.10.

Answer: The bandwidth of the signal is 2700 Hz. And the sampling rate is 7000 samples per second.

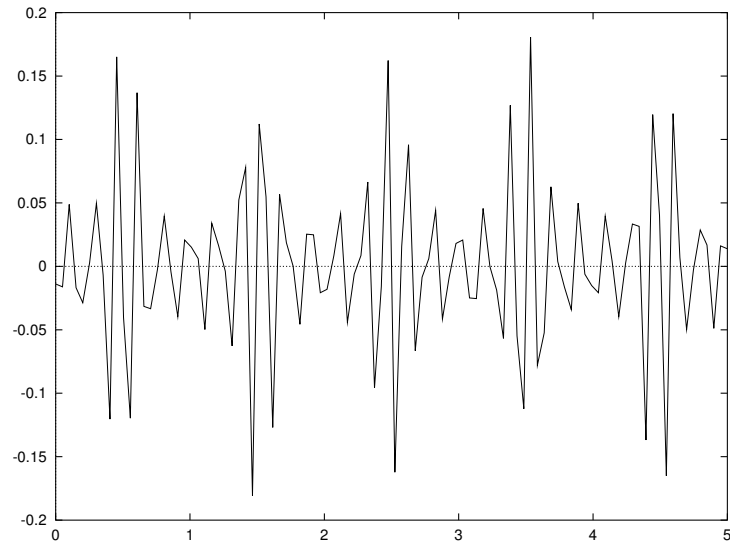


Figure 4: Passing through a 5-9Hz bandpass filter.

(a) SNR (in dB) and the number n of bits in the number of quantizing levels are related as

$$\text{SNR} = 6.02n + 1.76\text{dB}.$$

So we obtain $n \geq 4.69$. Thus, the number of quantizing levels needed is 32.

(b) With 7000 samples per second and 5 bits per sample, we obtain a data rate of 35000 bps.

8. (8 points) Delta modulation

Problem 6.13.

Answer: See Figure 5.

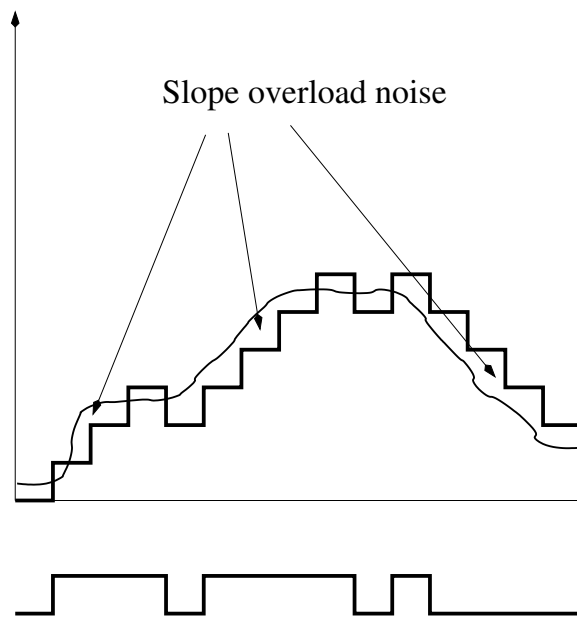


Figure 5: Delta modulation