## SIGNAL PROCESSING

## SCHOOL OF ELECTRICAL ENGINEERING

## Digital Signal Processing EQ2300 / 2E1340

Final Examination 2012–12–13, 14.00–19.00 Examples of Solutions

Answer: i).

b) As can be seen from the phase resonse, the filter is linear phase. A stable, causal IIR system cannot be linear phase, so i) is excluded. All Type 3 linear phase FIR systems have a zero both at f=0 and f=1/2, which excludes iii). All Type 2 linear phase FIR systems have a zero at f=1/2, which agrees with the plotted magnitude response. As an alternative, you could check the value of the phase response at f=0, which according to Table 4.1 in the book is 0 for Type 2 and  $\pi/2$  for Type 3.

Answer: ii)

c) The Matlab command produces the periodogram estimate. An M point DFT evaluates the TDFT at the frequency points f = k/M, so the peak  $k_0 = 88$  corresponds to a peak at  $f_0 = 88/M$ .

Answer: iv)

d) Two different features of the periodogram can be used to solve the problem. The first zero of  $W_R(f)$  (using the notations of Tsakonas&Bengtsson "Some Notes on Non-Parametric Spectral Estimation") is at f=1/N, so the main lobe extends from f=-1/N to f=1/N. From the figure, we see that the main lobe extends 8/M to the left and right around  $f=f_0$ , so we can conclude that 1/N=8/M, i.e., N=M/8=32. Alternatively, we can exploit that the height of the peak for a sinusoidal signal with amplitude A is  $NA^2/4$ , which gives  $4^2N/4=128$ , again resulting in N=32.

Answer: ii)

2. The result of the multiplication by c has to be rounded off, resulting in an additive quantization noise e(n) added after the multiplication. The standard approximations state that e(n) is white, uncorrelated with x(n) and has power  $\sigma_e^2 = 2^{-2B}/12$ , where in this example B = 7 (number of bits **excluding** the sign bit). Since the system is linear, superposition applies meaning that the output signal is the sum of the contribution from x(n) and e(n). Denote the output stemming from x(n) by  $y_x(n)$  and the output stemming from x(n) by  $y_x(n)$  and the output stemming from x(n) by  $y_x(n)$  and  $y_x(n)$  are uncorrelated and consequently, the total power spectral density of the output is  $y_x(n) = y_x(n) + y_y(n) + y_y(n) + y_y(n) = y_y(n) + y$ 

Contribution from x(n): The transfer function from x(n) to y(n) is given by

$$H(z) = \frac{1 + cz^{-1}}{1 - cz^{-1}}$$

so

$$P_{y_x}(f) = P_x(f) |H(f)|^2 = 0.1 \frac{|1 + ce^{-j2\pi f}|^2}{|1 - ce^{-j2\pi f}|^2} = 0.1 \frac{1 + c^2 + 2c\cos(2\pi f)}{1 + c^2 - 2c\cos(2\pi f)}$$

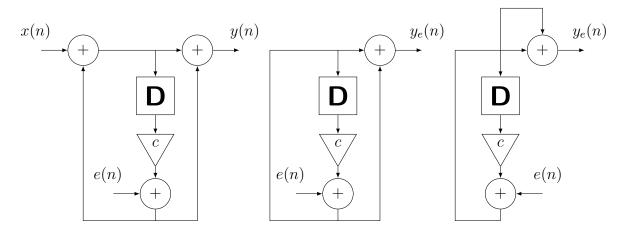


Figure 1: Approximate model for the quantization noise (left). Equivalent system considering only e(n) as the input (middle and right).

Contributions from e(n): The middle graph in Fig. 1 shows what the system looks like when only e(n) is the input. Noting that the same system can be drawn as in the right hand graph, if follows that the transfer function from e(n) to  $y_e(n)$  is

$$G(z) = \frac{2}{1 - cz^{-1}}$$

Consequently,

$$P_{y_e}(f) = P_e(f) |G(f)|^2 = \sigma_e^2 \frac{4}{|1 - ce^{-j2\pi f}|^2} = \frac{2^{-14}/3}{1 + c^2 - 2c\cos(2\pi f)}$$

The total power spectral density at the output is therefore

$$P_y(f) = P_{y_x}(f) + P_{y_e}(f) = \frac{2^{-14}/3 + 0.1(1+c^2) + 0.2c\cos(2\pi f)}{1 + c^2 - 2c\cos(2\pi f)}$$

**3.** The DTFT of x(n) is given by

$$X(f) = \sum_{m=-\infty}^{\infty} \frac{A_1}{2} \left( \delta(f-m-1/16) + \delta(f-m+1/16) \right) + \frac{A_2}{2} \left( \delta(f-m-1/8) + \delta(f-m+1/8) \right)$$

After downsampling by D=6, we obtain

$$W(f) = \frac{1}{6} \sum_{k=0}^{5} X\left(\frac{f-k}{6}\right)$$

The peaks of X(f) at  $f = \pm 1/16$  will end up as peaks of W(f) at  $f/6 = \pm 1/16$ , i.e. at  $f = \pm 3/8$ , plus periodic repetitions. The peaks of X(f) at  $f = \pm 1/8$  will end up as peaks of W(f) at  $f/6 = \pm 1/8$ , i.e. at  $f = \pm 3/4$ , plus periodic repetitions. The only of the latter that end up in the interval  $f \le 1/2$  are at  $f = \pm 1/4$ . What happens to the amplitudes? Consider first what happens in the time domain. Clearly, downsampling a single sinusoidal signal will result in a sinusoidal signal with the same amplitude, so we can directly conclude that the amplitudes of the two sinusoids will remain  $A_1$  and  $A_2$ , respectively. To see this mathematically, consider for example the term  $A_1/2\delta(f-1/16)$  in X(f). The corresponding term with k=0 in W(f) will be

$$\frac{1}{6} \frac{A_1}{2} \delta \left( \frac{f}{6} - \frac{1}{16} \right) = \frac{1}{6} \frac{A_1}{2} \delta \left( \frac{f - 6/16}{6} \right) = \frac{A_1}{2} \delta \left( f - \frac{3}{8} \right)$$

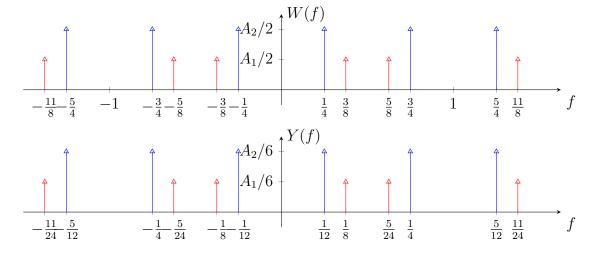


Figure 2: W(f) and Y(f)

where the last equality follows from  $\delta(f/C) = C\delta(f)$ . To summarize,

$$W(f) = \frac{A_1}{2} \left( \delta(f - 3/8) + \delta(f - m + 3/8) \right) + \frac{A_2}{2} \left( \delta(f - m - 1/4) + \delta(f - m + 1/4) \right) + \text{periodic repetitions},$$

see Fig. 2. An even simpler approach to reach the same result is to look at the time domain expressions,

$$w(k) = x(6k) = A_1 \cos\left(2\pi \frac{6}{16}k\right) + A_2 \cos\left(2\pi \frac{6}{8}k\right)$$

and note that  $\cos(2\pi \frac{3}{4}k) = \cos(2\pi k - 2\pi \frac{1}{4}k) = \cos(2\pi \frac{1}{4}k)$ .

After upsampling by I = 3 we obtain

$$Y(f) = W(3f)$$

giving the scaling of the frequency axis shown in Fig. 2. Note that we have to consider all peaks that end up in  $|f| \leq 1/2$  in Y(f), i.e., all peaks of W(f) in the interval  $|f| \leq 3/2$ . The above mentioned relationship  $\delta(f/C) = C\delta(f)$  means that the "height" of each Dirac pulse in Y(f) should be 1/3 of the corresponding Dirac pulse in W(f). In the time domain, the result displayed in Fig. 2 correspond to

$$y(n) = \frac{A_1}{3} \left( \cos\left(2\pi \frac{1}{8}k\right) + \cos\left(2\pi \frac{5}{24}k\right) + \cos\left(2\pi \frac{11}{24}k\right) \right) + \frac{A_2}{3} \left( \cos\left(2\pi \frac{1}{12}k\right) + \cos\left(2\pi \frac{1}{4}k\right) + \cos\left(2\pi \frac{5}{12}k\right) \right)$$

The amplitude by 1/3 makes sense if you for example consider that y(0) = w(0) = x(0) must hold.

Answer: See the lower plot of Fig. 2.

**4.** a) The transfer function from e(n) to x(n) is  $H(z) = 1/(1 - 0.2z^{-1})$  so the spectral density of x(n) is

$$P_x(f) = \sigma_e^2 |H(f)|^2 = \frac{1}{1 + \alpha^2 - 2\alpha \cos(2\pi f)}$$

where we used the notation  $\alpha = 0.2$ . Table lookup shows that the inverse DTFT is

$$r_{xx}(k) = \frac{1}{1 - \alpha^2} \alpha^{|k|} = \frac{1}{0.96} (0.2)^{|k|}$$

Since e(n) and w(n) are uncorrelated, also x(n) and w(n), so that

$$r_{yy}(k) = r_{xx}(k) + r_{ww}(k) = \frac{1}{0.96}(0.2)^{|k|} + 0.1\delta(k)$$

b) Use Yule-Walker to find the AR coefficients,

$$\begin{bmatrix} r_{yy}(0) & r_{yy}(1) \\ r_{yy}(1) & r_{yy}(0) \end{bmatrix} \begin{bmatrix} \hat{a}_1 \\ \hat{a}_2 \end{bmatrix} = - \begin{bmatrix} r_{yy}(1) \\ r_{yy}(2) \end{bmatrix}$$

which has the solution

$$\begin{bmatrix} \hat{a}_1 \\ \hat{a}_2 \end{bmatrix} = -\frac{1}{r_{yy}^2(0) - r_{yy}^2(1)} \begin{bmatrix} r_{yy}(0) & -r_{yy}(1) \\ -r_{yy}(1) & r_{yy}(0) \end{bmatrix} \begin{bmatrix} r_{yy}(1) \\ r_{yy}(2) \end{bmatrix}$$
$$= \frac{1}{r_{yy}^2(0) - r_{yy}^2(1)} \begin{bmatrix} r_{yy}(1) \left( r_{yy}(2) - r_{yy}(0) \right) \\ r_{yy}(0) r_{yy}(2) - r_{yy}^2(1) \end{bmatrix} \approx \begin{bmatrix} -0.182 \\ -0.003 \end{bmatrix}$$

The resulting AR(2) model is given by

$$\hat{y}(n) + \hat{a}_1 \hat{y}(n-1) + \hat{a}_2 \hat{y}(n-2) = v(n)$$

i.e.,  $\hat{y}(n) = g(n) * v(n)$ , where v(n) is white noise with power  $\hat{\sigma}_v^2 = r_{yy}(0) + \hat{a}_1 r_{yy}(1) + \hat{a}_2 r_{yy}(2) \approx 1.18$  and the filter g(n) has transfer function

$$G(z) = \frac{1}{1 + \hat{a}_1 z^{-1} + \hat{a}_2 z^{-2}}$$

c) The model based spectral estimate is given by the power spectral density of the estimated model:

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$$\hat{P}_y(f) = \hat{\sigma}_v^2 |G(z)|^2 = \frac{\hat{\sigma}_v^2}{|1 + \hat{a}_1 e^{-j2\pi f} + \hat{a}_2 e^{-j4\pi f}|^2}$$

- **5.** a) Since the length of x(n) is N = 50, and the length of h(n) is K = 10, the length of y(n) = x(n) \* h(n) is N + K 1 = 59 samples (more precisely, y(n) is in general non-zero for  $0 \le n \le N + K 2 = 58$ .
  - b) Consider first the linear convolution sum. Since both x(n) and h(n) are zero for n < 0, we obtain

$$y(n) = \sum_{k=\infty}^{\infty} x(k)h(n-k) = \sum_{k=0}^{n} x(k)h(n-k)$$

Assume that  $0 \le n \le N-1$  and split the sum defining the circular convolution into two parts, one where  $(n-k)_{\text{mod }N} = n-k$  and one where  $(n-k)_{\text{mod }N} = n-k+N$ .

$$y_c(n) = x(n) \textcircled{4} h(n) = \sum_{k=0}^{N-1} x(k) h((n-k)_{\text{mod } N})$$

$$= \sum_{k=0}^{n} x(k) h(n-k) + \sum_{k=n+1}^{N-1} x(k) h(n-k+N) \quad (1)$$

The first sum coincides with the linear convolution y(n). Compare the second sum with the linear convolution sum for y(n+N),

$$y(n+N) = \sum_{k=0}^{n+N} x(k)h(n+N-k) = \sum_{k=0}^{N-1} x(k)h(n+N-k)$$

where the second inequality holds since x(k) = 0 when k > N - 1 and similarly if k < n, then  $n - k + N > N \ge K$ , so that h(n - k + N) = 0. This means that the non-zero terms of the second sum in (1) coincide with y(n + N). This proves that  $y_c(n) = y(n) + y(n + 50)$  holds when  $0 \le n \le 49$ .

It is recommended to visualize the above result graphically.

c) Using the above result and the given values, we have

$$n = 0 \Rightarrow y(0) + y(50) = 10 \Rightarrow y(50) = 5$$

$$\vdots$$

$$n = 4 \Rightarrow y(4) + y(54) = 10 \Rightarrow y(54) = 5$$

$$n = 5 \Rightarrow y(5) + y(55) = 10 \Rightarrow y(55) = 10 - y(5) \Rightarrow y(5), \ y(55) \text{ cannot be determined}$$

$$\vdots$$

$$n = 8 \Rightarrow y(8) + y(58) = 10 \Rightarrow y(58) = 10 - y(8) \Rightarrow y(8), \ y(58) \text{ cannot be determined}$$

$$n = 9 \Rightarrow y(9) + y(59) = 10 \Rightarrow y(9) + 0 = 10 \Rightarrow y(9) = 10$$

$$\vdots$$

$$n = 49 \Rightarrow y(49) + y(99) = 10 \Rightarrow y(49) + 0 = 10 \Rightarrow y(49) = 10$$

To summarize,

$$y(n) = \begin{cases} 5, & 0 \le n \le 4 \\ \text{unknown}, & 5 \le n \le 8 \\ 10, & 9 \le n \le 49 \\ 5, & 50 \le n \le 54 \\ \text{unknown}, & 55 \le n \le 58 \end{cases}$$