Sample Final Exam (finals03) Covering Chapters 10-17 of *Fundamentals of Signals & Systems*

Problem 1 (25 marks)

Consider the discrete-time system shown below, where $\downarrow N$ represents decimation by N. This system transmits a signal x[n] coming in at 1000 samples/s over two low-bit-rate channels.



Numerical values:

lowpass filters' cutoff frequencies $\omega_{clp1} = \omega_{clp2} = \frac{\pi}{2}$, highpass filter's cutoff frequency $\omega_{chp} = \frac{\pi}{2}$, signal's spectrum over $[-\pi, \pi]$: $X(e^{j\omega}) = \begin{cases} 1 - \frac{4}{3\pi} |\omega|, \ |\omega| \le \frac{3\pi}{4} \\ 0, \ \frac{3\pi}{4} < |\omega| < \pi \end{cases}$.

(a) [7 marks] Sketch the spectra $X(e^{j\omega})$, $X_1(e^{j\omega})$, $X_2(e^{j\omega})$, $W_2(e^{j\omega})$, indicating the important frequencies and magnitudes.





(b) [10 marks] Let the decimation factors be $N_1 = 2$ and $N_2 = 4$. Sketch the corresponding spectra $V_1(e^{j\omega})$, $X_2(e^{j\omega})$, $V_2(e^{j\omega})$, indicating the important frequencies and magnitudes. Assuming for the moment that $v_1[n]$, $v_2[n]$ are quantized using 16-bit quantizers, find the bit rate of each channel, and the total bit rate. How would this compare to the bit rate for of a direct transmission of x[n] using a 16-bit quantizer?

Answer:



With $N_1 = 2$, the first channel transmits at a bit rate of:

 $\frac{1000}{2}$ samples/s×16bits/sample = 8000 bits/s

And with $N_2 = 4$, the second channel transmits at a bit rate of:

$$\frac{1000}{4}$$
 samples/s×16bits/sample = 4000 bits/s

Thus, the total bit rate is 12000 bits/s.

A direct transmission of the signal would require a bit rate of

1000samples/s×16bits/sample = 16000 bits/s

(c) [8 marks] Design the receiver system (draw its block diagram) such that y[n] = x[n] (assume that there is no quantization of the signals.) You can use upsamplers (symbol $\{\uparrow N\}_{lp}$, with embedded ideal lowpass filters of cutoff frequency $\frac{\pi}{N}$ and gain N), synchronous demodulators, ideal filters and summing junctions. Sketch the spectra of all signals in your receiver.

Answer:





Problem 2 (20 marks)

You are the engineer in charge of the design of a rocket's guidance control system so that the rocket can track a desired pitch angle trajectory $\alpha_{des}(t)$ in a vertical plane during the take-off phase. The transfer function from the rocket's thrust vector angle command with respect to its longitudinal axis, call it $\theta(t)$, to the angle between the rocket's pitch angle $\alpha(t)$ (angle between the longitudinal axis)

and the inertial vertical axis), is given by $G(s) := \frac{\hat{\alpha}(s)}{\hat{\theta}(s)} = \frac{1}{s(\frac{1}{9}s^2 + \frac{2}{3}s + 1)}$.



- (a) [8 marks] Suppose that you decide to use a pure gain $K \in \mathbb{R}$ as a feedback controller. Sketch the Bode plot of the loop gain L(s) and its Nyquist plot for K = 1, and use the Nyquist criterion to determine the range of K > 0 for which the rocket will be stable.
- (b) [8 marks] Compute the phase margin of the closed-loop system for K = 1. Assuming that the controller would be implemented on earth, what would be the longest communication delay that would not destabilize the control system?

Answer:

Loop gain:
$$L(s) = \frac{1}{s(\frac{1}{9}s^2 + \frac{2}{3}s + 1)}$$





Assuming that the Nyquist contour is indented so as to include the pole at 0, then the Nyquist criterion states that the number of counterclockwise encirclements of the critical point -1/K should be equal to 1. Note that there is already one counterclockwise encirclement at infinity due to the indentation around the pole s=0, so the "visible" part of the Nyquist plot of $L(j\omega)$ should leave the critical point to its left. On the Bode plot, this means that the gain should be smaller than 0dB at frequencies where the phase crosses the -180deg line. We can see on the Bode plot above that the gain is approximately -15dB at that point, therefore the closed-loop system

Phase margin:

The crossover frequency is approximately $\omega_{co} = 1$ rd/s. The phase margin is given by:

 $\angle L(j\omega_{co}) \cong -130^{\circ} \Rightarrow \phi_m = 50^{\circ}$. From the broken line approximation, we find a phase margin of $\angle L(j\omega_{co}) \cong -140^{\circ} \Rightarrow \phi_m = 40^{\circ}$. Using the latter, we compute the maximum time-delay that can be tolerated:

$$\omega_{co} \tau = \frac{\pi}{180} \phi_m$$
$$\Rightarrow \tau = \frac{\pi}{180 \omega_{co}} \phi_m = \frac{40\pi}{180 \cdot 10} = 0.0698 \,\mathrm{s}$$

(c) [4 marks] Compute the sensitivity function of the system and give the steady-state pitch angle error to a desired pitch angle step of 30 degrees on the output. *Answer:*

$$S(s) = \frac{1}{1+L(s)} = \frac{1}{1+\frac{1}{s(\frac{1}{9}s^2 + \frac{2}{3}s + 1)}} = \frac{s(\frac{1}{9}s^2 + \frac{2}{3}s + 1)}{s(\frac{1}{9}s^2 + \frac{2}{3}s + 1) + 1}$$

The Laplace transform of the error signal is

 $\hat{e}(s) = \frac{1}{s}S(s)$, and from the final value theorem, we have $\lim_{t \to +\infty} e(t) = S(0) = 0$. Therefore, the steady-state error to a 30deg step is 0deg, i.e., the rocket tracks the trajectory perfectly.

Problem 3 (15 marks)

Discrete-time AM-SSB modulator

You have to design an upper SSB modulator in discrete-time as an implementation of the continuoustime modulator shown below.



The modulating signal has a Fourier transform as shown below, and the carrier frequency is $\omega_c = 2\,000\,000\pi$ rd/s.



Design the modulation system (with ideal components) for the slowest sampling rate. Find the ideal phase-shift filter $H_d(e^{j\Omega})$ and compute its impulse response $h_d[n]$. Compute the frequency Ω_c . Sketch the spectra $X(e^{j\Omega})$, $Y_1(e^{j\Omega})$, $Y_2(e^{j\Omega})$, $Y(e^{j\Omega})$. Explain how you could implement an approximation to this ideal filter, and describe modifications to the overall system so that it would work in practice.

Answer:

First, the ideal phase-shift filter should have the following frequency response:

$$H_d(e^{j\Omega}) = \begin{cases} j, & 0 < \Omega < \pi \\ -j, & -\pi < \Omega < 0 \end{cases}$$

The inverse FT yields:

$$h_{d}[n] = \frac{1}{2\pi} \int_{-\pi}^{\pi} H_{d}(e^{j\Omega}) e^{j\Omega n} d\Omega$$

$$= \frac{1}{2\pi} \int_{-\pi}^{0} -j e^{j\Omega n} d\Omega + \frac{1}{2\pi} \int_{0}^{\pi} j e^{j\Omega n} d\Omega$$

$$= \frac{-j}{2\pi} \frac{1}{jn} \left[e^{j\Omega n} \right]_{-\pi}^{0} + \frac{j}{2\pi} \frac{1}{jn} \left[e^{j\Omega n} \right]_{0}^{\pi}.$$

$$= \frac{-1}{2\pi n} \left[1 - e^{-j\pi n} \right] + \frac{1}{2\pi n} \left[e^{j\pi n} - 1 \right]$$

$$= \frac{1}{\pi n} \left[(-1)^{n} - 1 \right]$$

To compute the slowest sampling frequency, we start from the desired upper SSB modulated signal at the output, whose bandwidth is $\omega_{M} = 2\,004\,000\pi$ which should correspond to $\Omega = \pi$. Thus, we can

compute the sampling period from the relationship
$$\frac{\Omega}{T} = \omega \implies \frac{\pi}{\omega_M} = T$$

So $T = 4.99 \times 10^{-7} s$. The DT carrier frequency is then: $\Omega_c = \omega_c T \implies \omega_c \frac{\pi}{\omega_M} = 200000\pi \frac{\pi}{2004000\pi} = \frac{500\pi}{501} = 3.1353$

This system could be implemented in practice using an FIR approximation to the ideal phase shift filter. Starting from $h_d[n]$, a windowed impulse response of length M+1 time delayed by M/2 to make

it causal would work. However, the resulting delay of M/2 samples introduced in the lower path of the modulator should be balanced out by the introduction of an equivalent delay $z^{-M/2}$ in the upper path.

Problem 4 (20 marks)

The system shown below demodulates the noisy modulated continuous-time signal $y_n(t) = y(t) + n(t)$ composed of the sum of:

- signal y(t) which is an upper single-sideband, suppressed carrier amplitude modulation (SSB/SC-AM) of x(t) (assume for $Y(j\omega)$ a magnitude of one half that of $X(j\omega)$),
- a noise signal n(t).

The carrier signal is $\cos(\omega_c t)$ where $\omega_c = 98000\pi$ rd/s.

The antialiasing filter $H_a(j\omega)$ is a perfect unity-gain lowpass filter with cutoff frequency ω_a .



The modulating (or message) signal x(t) has spectrum $X(j\omega)$ as shown below. The spectrum $N(j\omega)$ of the noise signal is also shown.



(a) [6 marks] Find the minimum antialiasing filter's cutoff frequency ω_a that will avoid any unrepairable distortion of the modulated signals due to the additive noise n(t). Sketch the spectra $Y_n(j\omega)$ and $W(j\omega)$ of signals $y_n(t)$ and w(t) for the frequency ω_a that you found. Indicate the important frequencies and magnitudes on your sketch.

Answer:



period *T* that would satisfy the sampling theorem, i.e., that would allow perfect reconstruction of the modulated signal. Give the corresponding cutoff frequency Ω_1 of the perfect lowpass filter and its gain K_1 so that $\tilde{x}(t) = x(t)$, and find the demodulation frequency Ω_c . Using these frequencies, sketch the spectra $W_d(e^{j\Omega}), V_d(e^{j\Omega}), X_d(e^{j\Omega})$, and $\tilde{X}(j\omega)$.

Answer:

$$\omega_s = 2\omega_a = 200000\pi, \ T = \frac{2\pi}{\omega_s} = \frac{1}{100000} = 10^{-5}$$

Demodulation frequency: $\Omega_c = 98000\pi T = 0.98\pi$. Cutoff frequencies: $\Omega_1 = 2000\pi T = 0.02\pi$. The lowpass filter must have a gain of $K_1 = 4$:





Problem 5 (20 marks)

The causal continuous-time LTI system given by its transfer function $G(s) = \frac{s}{(s+1)(s+2)}$ is discretized with sampling period T = 0.1 s for simulation purposes. The "c2d" transformation is used. (a) [4 marks] Find a state-space realization of the system. *Answer:*

$$G(s) = \frac{s}{(s+1)(s+2)} = \frac{s}{s^2 + 3s + 2}$$

The observable canonical state-space realization is given by:

$$\begin{bmatrix} \dot{x}_1 \\ \dot{x}_2 \end{bmatrix} = \begin{bmatrix} 0 & 1 \\ -2 & -3 \end{bmatrix} \begin{bmatrix} x_1 \\ x_2 \end{bmatrix} + \begin{bmatrix} 0 \\ 1 \end{bmatrix}_B u,$$

and the output equation is

$$y = \underbrace{[0 \quad 1]}_{C} \begin{bmatrix} x_1 \\ x_2 \end{bmatrix}.$$

and, D=0 .

(b) [10 marks] Compute the discrete-time state-space system $(A_{c2d}, B_{c2d}, C_{c2d}, D_{c2d})$ for G(s) and its associated transfer function $G_{c2d}(z)$, specifying its ROC. Sketch the pole zero plot of $G_{c2d}(z)$.

$$O_{c2d}(2)$$

Answer:

The A matrix is first diagonalized:

$$A = \begin{bmatrix} 0 & 1 \\ -2 & -3 \end{bmatrix} = \begin{bmatrix} 1 & 1 \\ -1 & -2 \end{bmatrix} \begin{bmatrix} -1 & 0 \\ 0 & -2 \end{bmatrix} \begin{bmatrix} 1 & 1 \\ -1 & -2 \end{bmatrix}^{-1}_{W}$$

Discretized state-space system using "c2d":

$$\begin{split} A_{d} &= e^{AT} = We^{\Lambda T}W^{-1} = \begin{bmatrix} 1 & 1 \\ -1 & -2 \end{bmatrix} \begin{bmatrix} e^{-0.1} & 0 \\ 0 & e^{-0.2} \end{bmatrix} \begin{bmatrix} 2 & 1 \\ -1 & -1 \end{bmatrix} \\ &= \begin{bmatrix} 1 & 1 \\ -1 & -2 \end{bmatrix} \begin{bmatrix} 2e^{-0.1} & e^{-0.1} \\ -e^{-0.2} & -e^{-0.2} \end{bmatrix} = \begin{bmatrix} 2e^{-0.1} - e^{-0.2} & e^{-0.1} - e^{-0.2} \\ -2e^{-0.1} + 2e^{-0.2} & -e^{-0.1} + 2e^{-0.2} \end{bmatrix} \\ &= \begin{bmatrix} 0.990944 & 0.086107 \\ -0.17221 & 0.73262 \end{bmatrix} \\ B_{d} &\coloneqq A^{-1} \begin{bmatrix} e^{AT} - I_{n} \end{bmatrix} B = \begin{bmatrix} -3/2 & -1/2 \\ 1 & 0 \end{bmatrix} \begin{bmatrix} 2e^{-0.1} - e^{-0.2} - 1 & e^{-0.1} - e^{-0.2} \\ -2e^{-0.1} + 2e^{-0.2} & -e^{-0.1} + 2e^{-0.2} - 1 \end{bmatrix} \begin{bmatrix} 0 \\ 1 \end{bmatrix} \\ &= \begin{bmatrix} -3/2 & -1/2 \\ 1 & 0 \end{bmatrix} \begin{bmatrix} e^{-0.1} - e^{-0.2} \\ -e^{-0.1} + 2e^{-0.2} - 1 \end{bmatrix} = \begin{bmatrix} -e^{-0.1} + 1/2e^{-0.2} + 1/2 \\ e^{-0.1} - e^{-0.2} \end{bmatrix} \\ &= \begin{bmatrix} 0.0045280 \\ 0.086107 \end{bmatrix} \\ C_{d} &\coloneqq C = \begin{bmatrix} 0 & 1 \end{bmatrix} \\ D_{d} &\coloneqq D = 1 \end{split}$$

transfer function:

$$\begin{aligned} G_{c2d}(z) &= C(zI_n - A)^{-1}B + D \\ &= \begin{bmatrix} 0 & 1 \end{bmatrix} \begin{bmatrix} z - 0.990944 & -0.086107 \\ 0.17221 & z - 0.73262 \end{bmatrix}^{-1} \begin{bmatrix} 0.0045280 \\ 0.086107 \end{bmatrix} \\ &= \frac{1}{(z - 0.990944)(z - 0.73262) + 0.17221(0.086107)} \begin{bmatrix} 0 & 1 \end{bmatrix} \begin{bmatrix} z - 0.73262 & 0.086107 \\ -0.17221 & z - 0.990944 \end{bmatrix} \begin{bmatrix} 0.0045280 \\ 0.086107 \end{bmatrix} \\ &= \frac{0.086107z - 0.086107}{z^2 - 1.72356z + 0.74081} = \frac{0.086107(z - 1)}{(z - 0.90484)(z - 0.81872)}, \ |z| > 0.90484 \end{aligned}$$

(c) [6 marks] Simulate the unit step response of G(s) by computing recursively the first 6 values (for n = 0, ..., 5) of the unit step response of the difference equation corresponding to $G_{c2d}(z)$. Also find the settling value of the step response by applying the final value theorem to $G_{c2d}(z)$.

Answer:

$$\begin{aligned} G_{c2d}(z) &= \frac{0.086107z - 0.086107}{z^2 - 1.72356z + 0.74081} \\ &= \frac{0.086107z^{-1} - 0.086107z^{-2}}{1 - 1.72356z^{-1} + 0.74081z^{-2}} \end{aligned}$$

Corresponding difference equation:

$$y[n] = 1.72356y[n-1] - 0.74081y[n-2] + 0.086107x[n-1] - 0.086107x[n-2]$$
Step response:

$$y[0] = 1.72356(0) - 0.74081(0) + 0.086107(0) - 0.086107(0)$$

$$= 0$$

$$y[1] = 1.72356(0) - 0.74081(0) + 0.086107(1) - 0.086107(0)$$

$$= 0.086107$$

$$y[2] = 1.72356(0.086107) - 0.74081(0) + 0.086107(1) - 0.086107(1)$$

$$= 0.1484$$

$$y[3] = 1.72356(0.1484) - 0.74081(0.086107) + 0.086107(1) - 0.086107(1)$$

$$= 0.1920$$

$$y[4] = 1.72356(0.1920) - 0.74081(0.1484) + 0.086107(1) - 0.086107(1)$$

$$= 0.2210$$

$$y[5] = 1.72356(0.2210) - 0.74081(0.1920) + 0.086107(1) - 0.086107(1)$$

$$= 0.2386$$
Final ender of the labelies (1 - 1), $\frac{G_{c2d}(z)}{2} = C_{c2d}(z)$

Final value: $y[+\infty] = \lim_{z \to 1} (1 - z^{-1}) \frac{G_{c2d}(z)}{(1 - z^{-1})} = G_{c2d}(1) = 0$

END OF EXAMINATION