

ENGINEERING TRIPOS PART IB

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Thursday 4 June 2009 2 to 4

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Paper 6

INFORMATION ENGINEERING

*Answer not more than **four** questions.*

*Answer not more than **two** questions from each section.*

*All questions carry the same number of marks.*

*The **approximate** number of marks allocated to each part of a question is indicated in the right margin.*

*Answers to questions in each section should be tied together and handed in separately.*

*Attachments: Additional copy of Fig. 4.*

STATIONERY REQUIREMENTS

Single-sided script paper

SPECIAL REQUIREMENTS

Engineering Data Book

CUED approved calculator allowed

**You may not start to read the questions printed on the subsequent pages of this question paper until instructed that you may do so by the Invigilator**

## SECTION A

*Answer not more than two questions from this section.*

1 (a) For a linear system characterised by a proper rational transfer function, explain how the locations of the poles determine the stability and response. Illustrate your answer using a diagram showing the typical impulse response for various pole locations. [6]

(b) For the system shown in Fig. 1, show that the closed loop transfer function relating  $\bar{y}(s)$  to  $\bar{x}(s)$  is

$$\frac{4(s-1)}{(s+1)^2} . \quad [3]$$

(c) Determine the open-loop and closed-loop poles for the system in Fig. 1. Comment on the open-loop and closed-loop stability of the system. [3]

(d) Find the response  $y(t)$  to a unit step input on  $x(t)$ , paying particular attention to the initial slope, any turning points and the final value of the response. Sketch the response  $y(t)$ . [8]

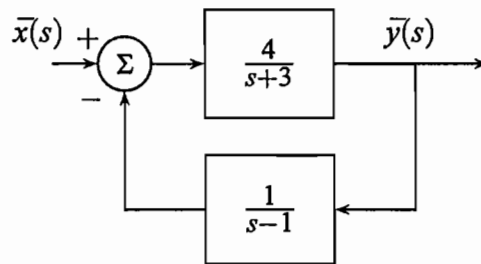


Fig. 1

2 (a) How can a *Nyquist diagram* for a linear system be constructed experimentally? [3]

(b) For the feedback system shown in Fig. 2, the controller  $K(s) = K$ , and the plant  $G(s)$  is given by

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

(i) Sketch the Nyquist diagram for the plant, showing any asymptotes. [5]

(ii) Describe how the magnitude of the closed loop frequency response can be determined from the Nyquist diagram. [2]

(iii) What is the gain margin of the feedback system if  $K = 1$  ? [2]

(iv) For a particular value of  $K$ , the phase margin is  $60^\circ$ . The frequency at which the phase margin is defined is  $\omega_c$ . Find the value of  $\omega_c$ , the corresponding value of  $K$ , and the magnitude of the closed loop frequency response at  $\omega_c$ . Comment on how these values would change for a phase margin of less than  $60^\circ$ . [8]

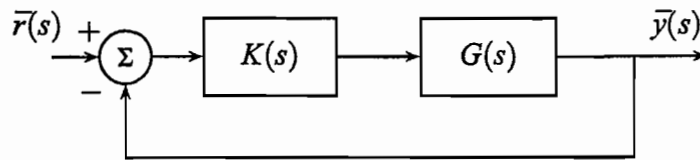


Fig. 2

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3 (a) Explain the meaning of the terms *gain margin* and *phase margin* of a feedback system. [3]

(b) For the feedback system shown in Fig. 3,  $G(s)$  is asymptotically stable and of the form

$$G(s) = \frac{12}{(1+sT_1)(1+sT_2)(1+sT_3)}$$

where  $T_1$ ,  $T_2$  and  $T_3$  are constants. The corresponding Bode diagram for  $G(s)$  is given in Fig. 4. For the case  $K(s) = 1$ , find the gain margin and phase margin of the closed loop system. [4]

(c)  $K(s)$  is now set to be a compensator with transfer function

$$K(s) = \frac{0.5(1+0.1s)}{(1+0.025s)}$$

Draw the Bode diagram of the compensated loop on the additional copy of Fig. 4 attached at the end of the paper. Estimate the phase margin and gain margin for the compensated loop. [9]

(d) Describe the differences in response to a unit step for the uncompensated and compensated loops. What is the steady-state error for both loops? [4]

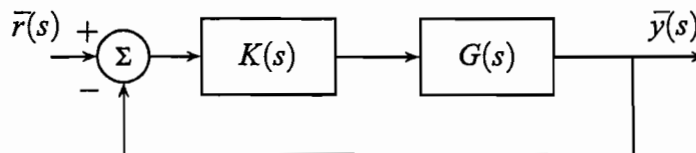


Fig. 3

(cont.)

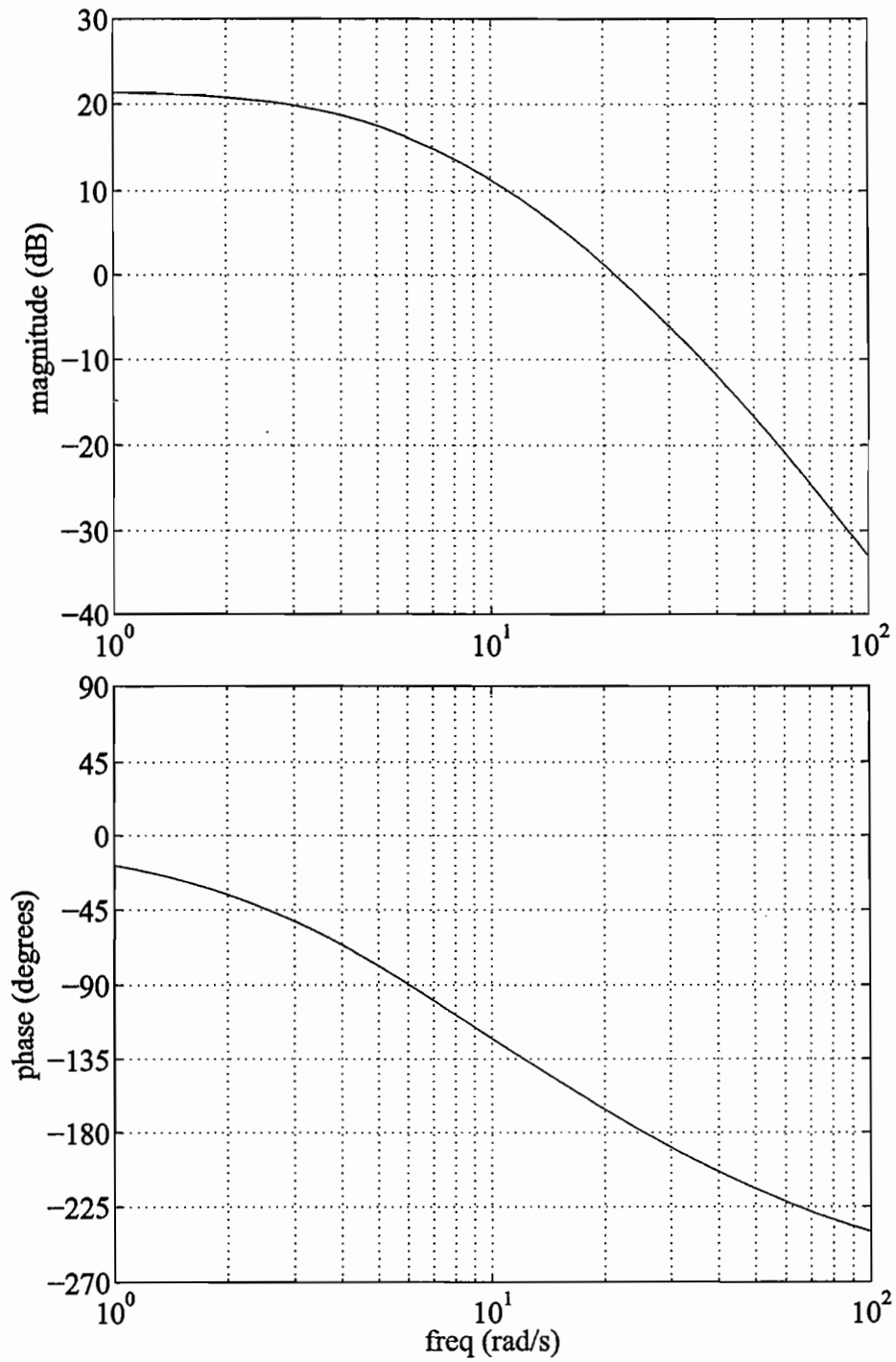


Fig. 4

**Note: an additional copy of Fig. 4 is attached at the end of this paper. This should be annotated with your constructions and handed in with your answer to this question.**

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## SECTION B

*Answer not more than two questions from this section.*

- 4 (a) Show, from first principles, the convolution property of the Fourier transform, i.e.,

$$\mathcal{F}\{x(t) * y(t)\} = \mathcal{F}\{x(t)\} \mathcal{F}\{y(t)\}$$

where  $*$  and  $\mathcal{F}\{.\}$  denote the convolution and Fourier transform operators, respectively. [4]

- (b) Let the signal  $y(t)$  be given by  $y(t) = x_1(t) + x_2(t)$  where, for  $i = 1, 2$

$$x_i(t) = \begin{cases} \sin \omega_i t & -\frac{T}{2} \leq t \leq \frac{T}{2} \\ 0 & \text{otherwise} \end{cases}$$

Assume  $\omega_1 > \omega_2$ . Spectral analysis is used to estimate  $\omega_1$  and  $\omega_2$ .

- (i) Calculate the Fourier transform,  $X_i(\omega)$ , of  $x_i(t)$ . [5]  
(ii) Sketch  $|Y(\omega)|$  for large and small values of  $T$ . What trade-offs between  $\omega_1, \omega_2$  and  $T$  do you envisage in estimating  $\omega_1$  and  $\omega_2$ ? [7]  
(iii) What advantage might be gained if the signals  $x_i(t)$  are obtained as

$$x_i(t) = \Lambda(t) \sin \omega_i t$$

where  $\Lambda(t)$  is a triangular pulse of duration  $2T$  centered at  $t = 0$ ? [4]

5 A signal of bandwidth  $B$  Hz is to be digitised.

(a) The signal is first sampled at a rate of  $f_s = 1/T_s$  samples per second, where  $T_s$  is the sampling period. By taking the Fourier transform of the sampled signal, show that if  $f_s \geq 2B$ , it is possible to have perfect signal recovery from the sampled signal. Explain the steps involved. [5]

(b) The sampled signal is next quantised with a uniform quantiser of step size  $\Delta$ . Show that the RMS quantisation noise is given by  $\Delta/\sqrt{12}$ , stating the assumptions made. [5]

(c) Calculate the minimum resolution (in bits) necessary to achieve an SNR of 30 dB when a sinusoidal signal with amplitude 5 V is quantised with a uniform quantiser. [4]

(d) Calculate the resulting minimum data rate of the digitised signal. [2]

(e) Fig. 5 shows the input-output characteristic of a non-uniform quantiser. Explain the advantages of this over a uniform quantiser. For what type of signals would this non-uniform quantiser be most effective? [4]

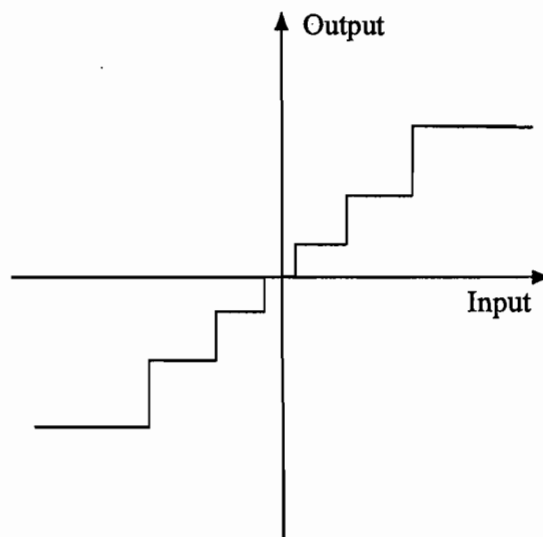


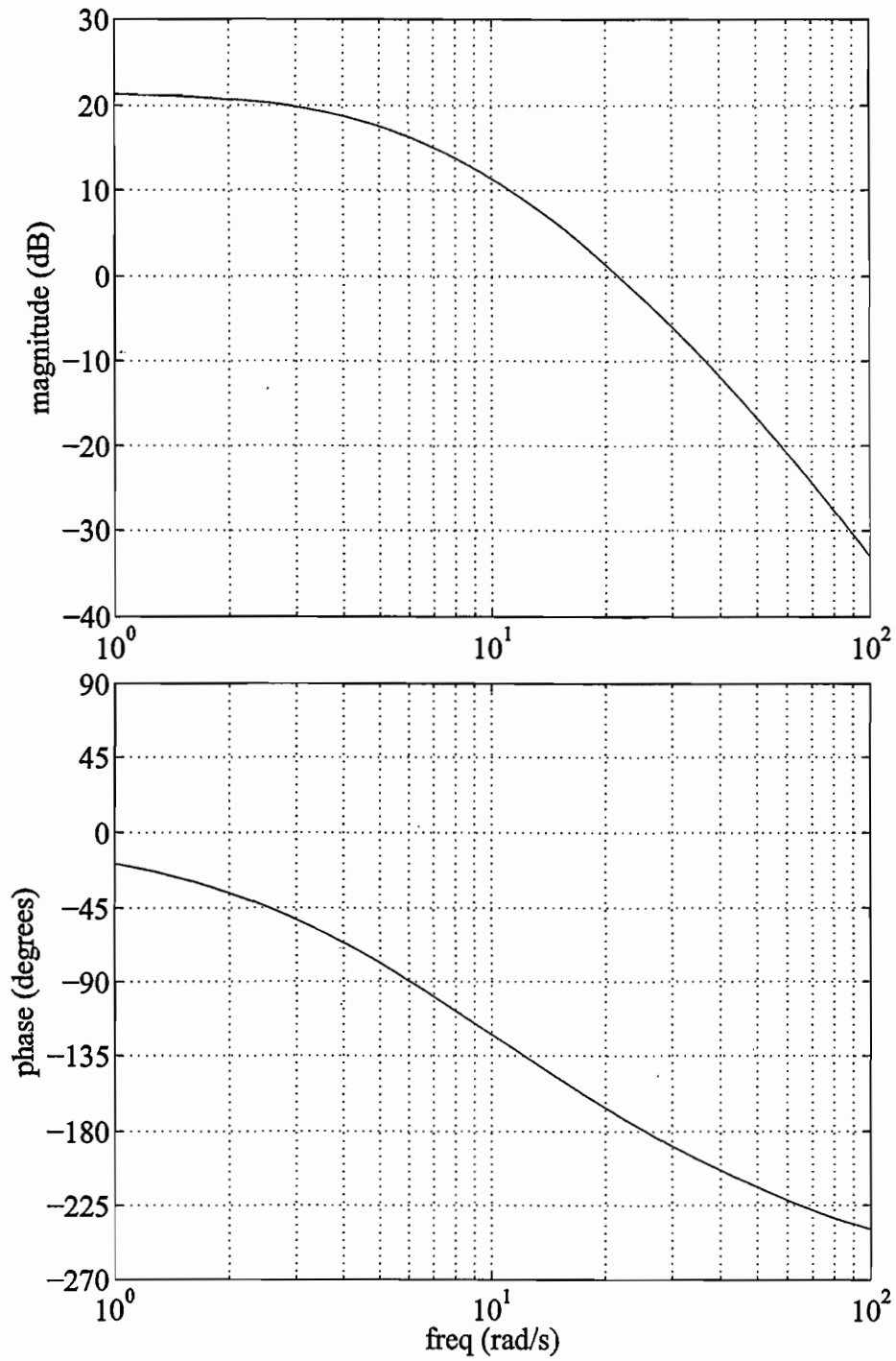
Fig. 5

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- 6 (a) Describe *modulation* and explain why it is used in practical communications systems. [2]
- (b) Explain the differences in bandwidth consumption between the following modulation schemes: AM, DSB-SC, SSB-SC, FM and BPSK. [5]
- (c) Show that the error probability of BPSK when transmitted over an additive white Gaussian noise (AWGN) channel is given by  $P_e = Q(\sqrt{2\text{SNR}})$ , where  $Q(x) = 1 - \Phi(x)$ , and  $\Phi(x)$  is the cumulative distribution function of a zero mean unit variance Gaussian random variable. [5]
- (d) Find the minimum SNR required to ensure that a BPSK modulated system has a probability of error  $P_e \leq 5 \times 10^{-6}$ . [4]
- (e) Calculate the total bandwidth required to accommodate 20 BPSK users transmitting at rate  $R = 50$  kbit/s with a rectangular pulse shape. Assume that no interference is caused beyond the second zero of the sinc function. Repeat the above calculation if now we need a guard band of 10 kHz between users. [4]

**END OF PAPER**





**Copy of Fig. 4. This should be annotated with your constructions and handed in with your answer to question 3.**