

# 东南大学考试卷 (A 卷)

课程名称 通信原理 考试学期 04-05-3 得分 \_\_\_\_\_  
适用专业 \_\_\_\_\_ 考试形式 闭卷 考试时间长度 150 分钟

## Section A(30%): True or False (Give your reason if False, 2% for each question)

1. 1. A typical mobile radio channel is a free propagation, linear, and time invariant channel. (X)
2. 2. The power spectral density of a stationary process is always nonnegative. (✓)
3. 3. In a communication system, noise is unwanted and over which we have incomplete control. (X)
4. 4. If a random process is stationary, it is ergodic; if a Gaussian random process is stationary, then it is also strictly stationary. (X)
5. 5. Double Sideband-Suppressed Carrier (DSB-SC), Single Sideband (SSB), and Frequency Modulation (FM) are all linear modulation schemes. (X)
6. 6. Figure of merit (defined as  $(\text{SNR})_o/(\text{SNR})_c$ ) of AM of DSB-SC is 1/3, and figure of merit of Amplitude Modulation (AM) is less than or equal to 1/3. (X)
7. 7.  $\mu$ -law is a nonlinear compression law and A-law is a linear compression law. (X)
8. 8. The matched filter at the receiver maximizes the peak pulse signal-to-noise ratio, thus is optimal in a baseband data transmission system with Inter-Symbol Interference (ISI). (X)
9. 9. Correlative-level coding (also known as partial-response signaling) schemes are used to avoid ISI. (X)
10. 10. Time-Division Multiplexing (TDM) is used in Asymmetric Digital Subscriber Lines (ADSL) to separate voice signals and data transmission. (X)
11. 11. If coefficients of an equalizer is adjusted using the Least-Mean-Square (LMS) algorithm adaptively, then the matched filter in front of the equalizer is not necessary. (✓)
12. 12. In an M-ary Phase-Shift Keying (M-PSK) system, if the average probability of symbol error is  $P_e$ , then the average Bit Error Rate (BER) of the system is  $P_e/\log_2 M$ . (X)
13. 13. With the same Signal-to-Noise Ratio (SNR), 16-ary Quadrature Amplitude Modulation (16-QAM) has better performance than 16-ary Phase-Shift Keying (16-PSK). The reason is that 16-QAM has constant envelop. (X)
14. 14. With the same SNR, Minimum Shift Keying (MSK) has better performance than

Sunde's Frequency-Shift Keying (FSK). They are both Continuous-Phase Frequency-Shift Keying (CPFSK).

15. 15. If the largest frequency component of an band-limited signal  $X(t)$  is at 100 Hz, then the corresponding Nyquist rate is 200 Hz.

**Section B(30%): Fill in the Blanks (3% for each question)**

1. 1. The power spectral density of a stationary process  $X(t)$  is  $S_X(f)$ , then the autocorrelation function of  $X(t)$  is  $R_X(t) = \int_{-\infty}^{\infty} S_X(f) \exp(j2\pi f t) df$ .
2. 2. A random process  $Y(t)$  is defined as  $Y(t) = X(t) \cos(2\pi f_c t + \Theta)$ , where  $X(t)$  is a stationary process,  $f_c$  is a constant frequency, and the phase  $\Theta$  is randomly distributed over the interval  $[0, 2\pi]$ . Suppose the power spectral density of  $X(t)$  is  $S_X(f)$ , the power spectral density of  $Y(t)$  is  $S_Y(f) = \frac{1}{4} [S_X(f - f_c) + S_X(f + f_c)]$ .
3. 3. In a Frequency Modulation (FM) system, the modulating signal is  $m(t) = 2 \cos(6\pi t)$  V, the frequency sensitivity is  $k_f = 0.3$  Hz/V. Using Carson's rule, bandwidth of the FM signal is approximately 72 Hz.
4. 4. An analog signal is first encoded into a binary Pulse-Code Modulation (PCM) wave. Sampling rate of the PCM system is 8 kHz, number of representation levels is 64. The binary PCM wave is transmitted over a baseband channel using a 4-ary Pulse-Amplitude Modulation (PAM) (that is, a PAM with 4 amplitude levels). The minimum bandwidth requirement for transmitting the PAM wave is 12 kHz.
5. 5. Basic operations performed in the transmitter of a PCM system include Sampling, Quantization, and Encoding.
6. 6. Bandwidth efficiency of 4-ary Quadrature Amplitude Modulation (QAM) is 1; bandwidth efficiency of 8-ary Phase-Shift Keying (8PSK) is 1.5.
7. 7. In a Delta Modulation (DM) system, sampling rate is  $f_s = 8$  kHz and step size is  $\Delta = 0.1$  V. If the input to the DM system is a 1 kHz sinusoidal signal, then to avoid slope overload, the maximum amplitude of this input signal is  $0.4/\pi$  V.
8. 8. 12 different message signals, each with a bandwidth of 20 kHz, are to be multiplexed and transmitted. If the multiplexing and modulation methods are Frequency-Division Multiplexing (FDM) and Single-Sideband Modulation (SSB), respectively, then the minimum bandwidth required is 240 kHz; if the multiplexing and modulation methods are Time-Division Multiplexing (TDM) and PAM, then the minimum bandwidth required is 120 kHz.
9. 9. The binary data sequence  $\{b_k\} = 1010110$  is applied to a duobinary (class I partial response) system with precoding, as shown in Fig. P1-9 (see next page), where  $d_k = b_k \oplus d_{k-1}$ ,  $\oplus$  denotes modulo-two addition, and  $a_k = 2 \times d_k - 1$ . The initial value of  $d_k$  is 1. The output of the duobinary encoder is  $\{c_k\} = 0-2 \ 0 \ 2 \ 0 \ 0 \ 2$ . If at the receiving end,

the first received digit is 1.25 due to noise, then the receiver output is  $\{\hat{b}_k\} = 0010110$ .

$b_k$  1010110  
 $d_k$  10011011  
 $a_k$  1-11111  
 $c_k$  0-202002  
 1st 0010110

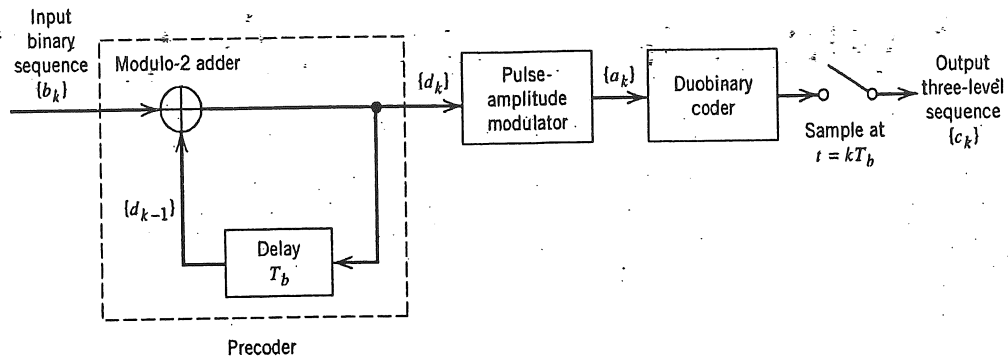


Fig. P1-9

10. 10. A communication system has the signal-space diagram shown in Fig. P1-10, where message points have equal probability of transmitting. Assume communication channel in this system is Additive White Gaussian Noise (AWGN) channel, and  $E/N_0$  is 12, where  $N_0$  is single-sideband power spectral density of the AWGN. Using union bound, the average probability of symbol error is overbounded as  $P_e \leq \frac{1}{2} \left( 1 - \frac{1}{\sqrt{E/N_0}} \right)$ . (Express using complementary error function  $\text{erfc}(\cdot)$ .)

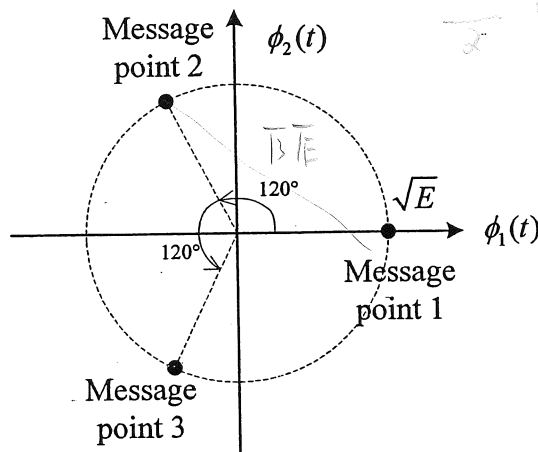


Fig. P1-10

**Section C(40%): Calculations (8% for each question)**

1. 1. In a coherent Binary Phase-Shift Keying (BPSK) system, symbols 1 and 0 are represented by signals  $s_1(t)$  and  $s_2(t)$ , respectively. The signals are defined by

$$\begin{cases} s_1(t) = \sqrt{\frac{2E_b}{T_b}} \cos(2\pi f_c t) \\ s_2(t) = \sqrt{\frac{2E_b}{T_b}} \cos(2\pi f_c t + \pi) \end{cases} \quad 0 \leq t \leq T_b$$

where  $E_b$  is the transmitted signal energy per bit, and  $T_b$  is one bit duration.

- Determine the basis function(s) of coherent BPSK signal constellation;
- Plot the signal-space diagram of coherent BPSK system;
- Determine the error probability of BPSK;
- If there is a phase error  $\phi$  between the phase references of the transmitter and receiver, determine the error probability of BPSK in this condition again.

2. 2. Spectrum of a message signal  $m(t)$  is shown in Fig. P3-2. This message signal is Double Sideband-Suppressed Carrier (DSB-SC) modulated with a carrier wave  $A_c \cos(2\pi f_c t)$ .

- If  $f_c = 2$  kHz, plot the spectrum of the modulated signal  $s(t)$ ;
- What is the baseband bandwidth  $W$  of the message signal  $m(t)$ ?
- What is the transmission bandwidth  $B_T$  of the DSB-SC modulated signal?
- What is the lowest value of  $f_c$  that keeps the DSB-SC modulation from sideband overlap?

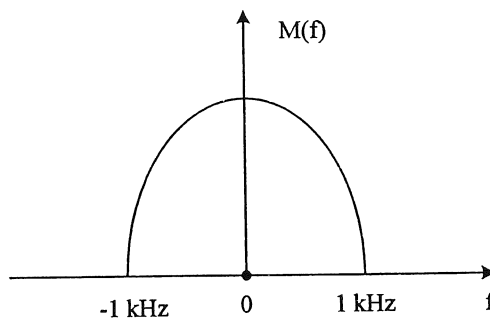


Fig. P3-2

3. 3. Consider a Quadriphase-Shift Keying (QPSK) system. The transmitted signal set is defined as:

$$s_i(t) = \begin{cases} \cos[2\pi f_c t + (2i-1)\pi/4], & 0 \leq t \leq T \\ 0, & \text{otherwise} \end{cases}$$

where  $i = 1, 2, 3, 4$ . Every two input bits select one of the signals in transmitted signal set to transmit. The rule of mapping is  $10 \rightarrow s_1(t)$ ,  $00 \rightarrow s_2(t)$ ,  $01 \rightarrow s_3(t)$ ,  $11 \rightarrow s_4(t)$ .

$$\phi_1(t) = \frac{P}{T} \cos(2\pi f_c t) \quad \text{and} \quad \phi_2(t) = \frac{P}{T} \sin(2\pi f_c t)$$

$$s(t) = \begin{bmatrix} \sqrt{E_b} \cos(2\pi f_c t) \\ -\sqrt{E_b} \sin(2\pi f_c t) \end{bmatrix}$$

- a) Determine the basis function(s) of QPSK signal constellation;
- b) Express  $s_i(t)$  using the basis function(s);
- c) If the input binary sequence is 01101000, and suppose  $f_c = 2/T$ , plot the QPSK waveform;
- d) Is the QPSK waveform continuous phase?

4. Consider the signal  $s(t)$  shown in Fig. P3-4,

- a) Assuming  $h(t)$  is the matched filter of  $s(t)$ , plot the impulse response of  $h(t)$ ;
- b) When  $s(t)$  is applied to  $h(t)$ , plot the matched filter output in the time domain.

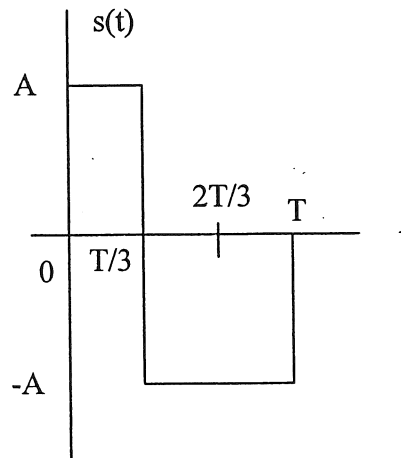


Fig. P3-4

5. Suppose  $X(t) = A \cos(2\pi f t - \Theta)$ , where  $A$  is a constant, and  $f$  and  $\Theta$  are independent.  $\Theta$  is uniformly distributed over the interval  $[0, 2\pi]$ . Determine the power spectrum density of  $X(t)$  in terms of the probability density function of the frequency  $f$ .

$$R_X(\tau) = E[X(t+\tau)X(t)]$$

$$= E[A \cos(2\pi f(t+\tau) - \Theta) A \cos(2\pi f t - \Theta)]$$

$$= A^2 E[\frac{1}{2} \cos(4\pi f t + 2\pi f \tau - \Theta) + \frac{1}{2} \cos(2\pi f \tau)]$$

$$= \frac{A^2}{2} E[\cos(2\pi f \tau)]$$

$$= \frac{A^2}{2} \int_{-\infty}^{\infty} F_f(f) \cos(2\pi f \tau) df$$

$$S_X(f) = \int_{-\infty}^{\infty} R_X(\tau) \exp(-j2\pi f \tau) d\tau$$

$$= \frac{A^2}{2} \int_{-\infty}^{\infty} \int_{-\infty}^{\infty} F_f(f) \cos(2\pi f \tau) \cos(-2\pi f \tau) df d\tau$$

$$= \frac{A^2}{2} \int_{-\infty}^{\infty} F_f(f) df$$

## 通信原理期终考试参考答案和评分标准

考试学期: 04-05-3

## Section A: (30分)

1. × The mobile radio channel is typically time variant.
2. ✓
3. ✓
4. × Stationary is not necessary ergodic.
5. × FM is not a linear modulation scheme
6. × Figure of merit of DSB-SC is 1, not 1/3.
7. × A-law is also a nonlinear compression law.
8. × The matched filter is optimal with AWGN channel.
9. × Correlative-level coding is to use ISI to achieve  $2W$  signaling rate in a bandwidth of  $W$  Hz.
10. × FDM is used in ADSL.
11. ✓
12. × The BER is usually not  $P_e/\log_2 M$ .
13. × 16-QAM is not constant envelop.
14. ✓
15. ✓

## Section B: (30分)

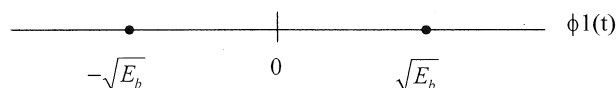
1.  $\int_{-\infty}^{\infty} S_X(f) \exp(j2\pi f\tau) df$
2.  $\frac{1}{4}[S_X(f-f_c) + S_X(f+f_c)]$
3. 7.2
4. 12
5. Sampling, Quantizing, and Encoding
6. 1, 1.5
7. 1.27
8. 240, 240
9. 0 -2 0 2 0 0 2, 0010110
10.  $\text{erfc}(3)$

## Section C: (40分)

1. (a) The basis function is:

$$\phi_1(t) = \sqrt{\frac{2}{T_b}} \cos(2\pi f_c t), \quad 0 \leq t \leq T_b \quad \text{归一化 (2分)}$$

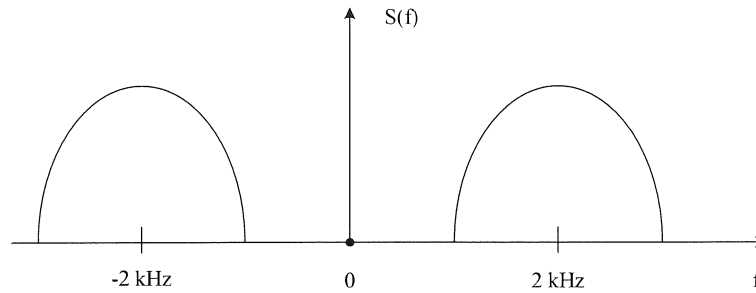
- (b) The signal-space diagram is: (2分)



(c) The error probability is  $P_e = \frac{1}{2} \operatorname{erfc} \left( \sqrt{\frac{E_b}{N_0}} \right)$ . (2分)

With phase error  $\varphi$ , the error probability becomes  $P_e = \frac{1}{2} \operatorname{erfc} \left( \sqrt{\frac{E_b}{N_0}} \cos \varphi \right)$ . (2分)

2. (a) The spectrum is: (2分)



(b)  $W = 1$  kHz; (2分)

(c)  $B_T = 2$  kHz; (2分)

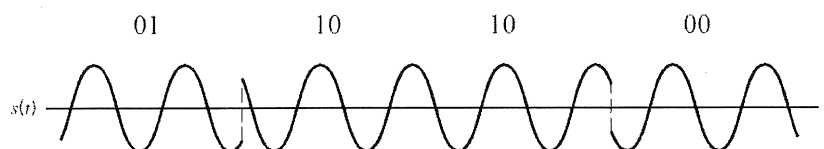
(d) To avoid sideband overlap, the lowest frequency is 1 kHz. (2分)

3. (a) The basis functions are: (2分)

$$\begin{cases} \phi_1(t) = \sqrt{\frac{2}{T}} \cos(2\pi f_c t), & 0 \leq t \leq T \\ \phi_2(t) = \sqrt{\frac{2}{T}} \sin(2\pi f_c t), & 0 \leq t \leq T \end{cases}$$

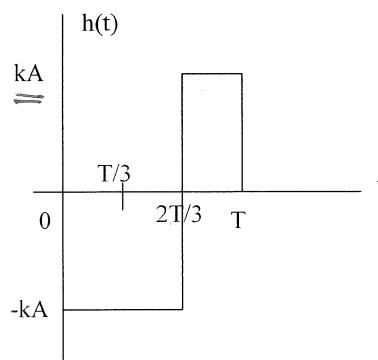
(b)  $s_i(t) = \sqrt{\frac{T}{2}} \cos \left[ (2i-1) \frac{\pi}{4} \right] \phi_1(t) - \sqrt{\frac{T}{2}} \sin \left[ (2i-1) \frac{\pi}{4} \right] \phi_2(t)$ ,  $i = 1, 2, 3, 4$  (2分)

(c) The QPSK waveform is: (2分)

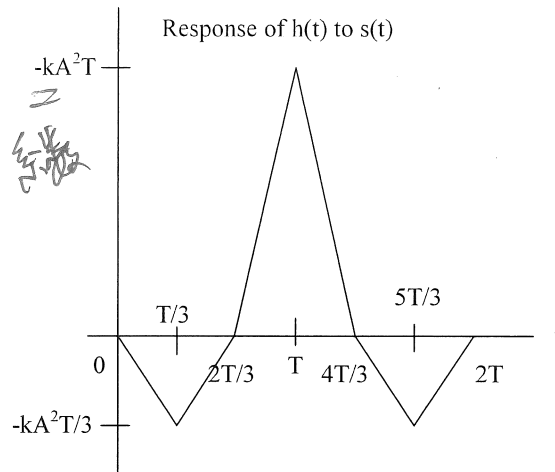


(d) The QPSK waveform is not continuous phase. (2分)

4. (a)  $h(t)$  is shown below, where  $k$  is a positive constant. (4分)



(b) The filter output is shown below. (4 分)



5. The autocorrelation function of  $X(t)$  is: (2 分)

$$\begin{aligned}
 R_X(\tau) &= E[X(t+\tau)X(t)] \\
 &= A^2 E[\cos(2\pi Ft + 2\pi F\tau - \Theta) \cos(2\pi Ft - \Theta)] \\
 &= \frac{A^2}{2} E[\cos(4\pi Ft + 2\pi F\tau - \Theta) + \cos(2\pi F\tau)]
 \end{aligned}$$

Since  $\Theta$  is uniformly distributed over the interval  $[0, 2\pi]$ , we get: (2 分)

$$\begin{aligned}
 R_X(\tau) &= \frac{A^2}{2} E[\cos(2\pi F\tau)] \\
 &= \frac{A^2}{2} \int_{-\infty}^{\infty} f_F(f) \cos(2\pi f\tau) df
 \end{aligned}$$

Since  $X(t)$  is a real-valued random process,  $S_X(f)$  is an even function of frequency, we also have: (2 分)

$$\begin{aligned}
 R_X(\tau) &= \int_{-\infty}^{\infty} S_X(f) \exp(j2\pi f\tau) df \\
 &= \int_{-\infty}^{\infty} S_X(f) \cos(j2\pi f\tau) df
 \end{aligned}$$

where  $S_X(f)$  is the power spectrum density of  $X(t)$ . Therefore, we get: (2 分)

$$S_X(f) = \frac{A^2}{2} f_F(f)$$