科目: 通訊系統(5007)

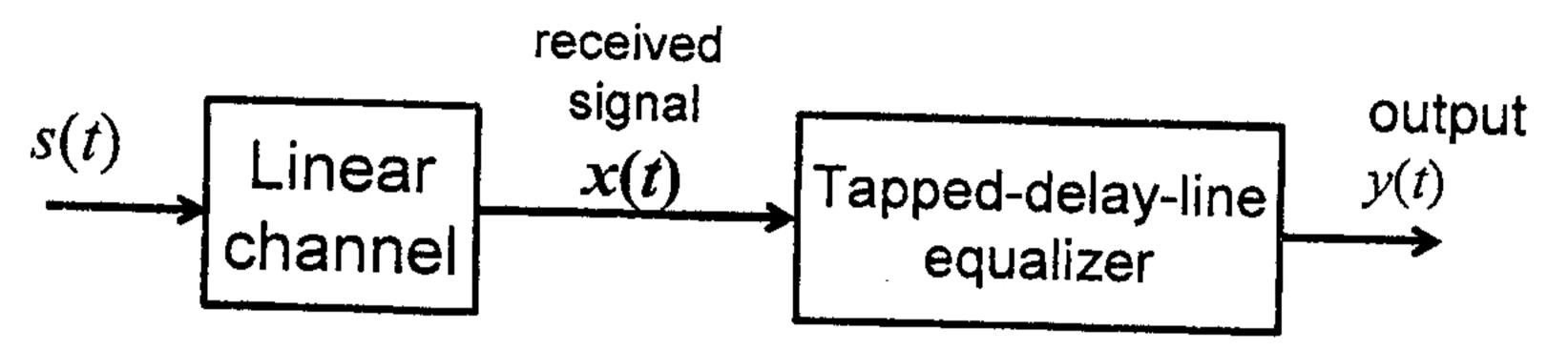
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1. (20%) Consider the communication system in response to a signal s(t) as in the following figure. The linear channel suffers from multipath distortion and the channel output is defined by

$$x(t) = a_1 s(t - \tau_1) + a_2 s(t - \tau_2)$$

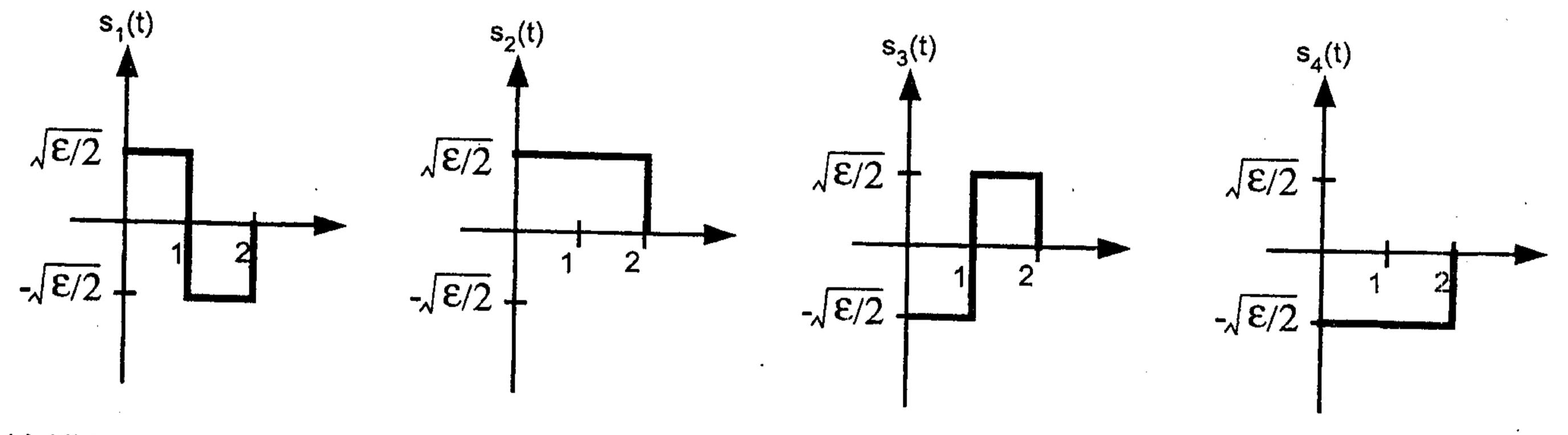
where  $a_1$  and  $a_2$  are constant and  $\tau_1$  and  $\tau_2$  represent the corresponding delays of the propagation paths.



Now you are supposed to design the tapped-delay-line filter to equalize the multipath distortion produced by this channel. The time response of the tapped-delay-line filter is

$$y(t) = w_0 x(t) + w_1 x(t-T) + w_2 x(t-2T).$$

- (a) (4%) Find the frequency transfer function of the linear channel.
- (b) (6%) Identify the desired frequency response at the equalizer output such that the channel multipath distortion is equalized.
- (c) (10%) Assuming that  $a_2 \ll a_1$  and  $\tau_2 > \tau_1$ , evaluate the parameters of the tapped-delay-line equalizer, i.e.  $w_0$ ,  $w_1$ ,  $w_2$ , and T, such that the channel multipath distortion is equalized.
- 2. (20%) Suppose we transmit a signal s(t) through an additive white Gaussian noise channel. Assume that s(t) is equal to  $s_1(t)$ ,  $s_2(t)$ ,  $s_3(t)$ , or  $s_4(t)$  with equal probability, where the waveforms are given as follows:



- (a) (6%) Use the Gram-Schmidt procedure in the order of  $s_1(t)$ ,  $s_2(t)$ ,  $s_3(t)$ ,  $s_4(t)$  to find the ordered set of orthonormal basis functions  $\{f_1(t), f_2(t), \dots, f_K(t)\}$ , where K is the dimension of the signals. Find K and give the vector expression
- s<sub>1</sub>, s<sub>2</sub>, s<sub>3</sub>, s<sub>4</sub> for each of the waveforms, respectively, with the ordered set of basis functions derived above.
- (b) (8%) Following from (a), suppose that the signal is passed through a vector channel so that the receiver observes  $\mathbf{r} = \mathbf{C} \cdot \mathbf{s} + \mathbf{n}$ .

where **n** is a Gaussian vector with zero mean and covariance matrix  $\Sigma = \operatorname{diag}(\sigma_1^2, \sigma_2^2, \dots, \sigma_K^2)$ . Derive the optimal detector that minimizes the error probability when C = 1, where 1 is the K-by-K identity matrix, and compute the symbol error rate assuming that C = 1 and  $\sigma^2 \triangleq \sigma_1^2 = \sigma_2^2 = \dots = \sigma_K^2$ .

(c) (6%) Following from (b), derive the symbol error probability for  $\mathbf{C} = \begin{bmatrix} \frac{1}{\sqrt{2}} & -\frac{1}{\sqrt{2}} \\ \frac{1}{\sqrt{2}} & \frac{1}{\sqrt{2}} \end{bmatrix}$  and  $\sigma^2 \triangleq \sigma_1^2 = \sigma_2^2 = \cdots = \sigma_K^2$ .

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3. (10%) Let X(t) be a baseband transmitted signal of a symbol sequence  $A_n$  given by

$$X(t) = \sum_{n=-\infty}^{\infty} A_n g(t - T_d - nT)$$

where  $A_n$  is an independent identically distributed (iid) complex random sequence with zero mean and variance  $\sigma_A^2$ ,  $T_d$  is a random variable uniformly distributed over [0, T], and g(t) is a pulse shaping function. It is known that the power spectral density of X(t) is given by

$$S_{XX}(f) = \frac{\sigma_A^2}{T} |G(f)|^2$$

where G(f) is the Fourier transform of g(t).

(a) (3%) Find  $S_{XX}(f)$  if g(t) = u(t) - u(t-T) where u(t) is a unit-step function.

(b) (7%) Let  $p(t) = u(t) - u(t - T_c)$  and

$$g(t) = \sum_{k=0}^{N-1} c_k p(t - kT_c)$$

where  $T_c = T/N$  and  $c_k$  are iid binary random variables of  $\{\pm 1\}$  with  $\Pr(c_k = 1) = \Pr(c_k = -1) = 1/2$ . Find  $S_{XX}(f)$ .

What are the distinctions between the results of part (a) and part (b)?

4. (10%) Consider a coherent binary frequency shift keying (BFSK) system where symbols '0' and '1' occur with equal probability. Let symbols '1' and '0' be encoded by signals  $s_1(t)$  and  $s_2(t)$ , respectively, where

$$s_i(t) = \begin{cases} \sqrt{\frac{2E_b}{T_b}}\cos(2\pi f_i t), & 0 \le t \le T_b \\ 0, & \text{otherwise} \end{cases}$$

in which  $E_b$  is the transmitted signal energy per bit,  $T_b$  is the symbol duration and  $f_i = (n_c + i)/T_b$  for some fixed integer  $n_c$ . The received signal can be expressed as

$$x(t) = s_i(t) + w(t)$$

where w(t) is a white Gaussian process with zero mean and power spectral density equal to  $\mathcal{N}_0/2$ .

- (a) (5%) Determine the optimum receiver with minimum bit error rate (BER).
- (b) (5%) Derive the BER of the optimum receiver in terms of the complementary error function or Q-function defined as follows:

$$Q(u) = \int_{u}^{\infty} \frac{1}{\sqrt{2\pi}} e^{-z^{2}/2} dz = \frac{1}{2} \operatorname{erfc}(\frac{u}{\sqrt{2}})$$

$$\operatorname{erfc}(u) = \frac{2}{\sqrt{\pi}} \int_{u}^{\infty} e^{-z^{2}} dz = 2 Q(\sqrt{2}u)$$

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5. (20%) Consider the random variables X and Y with joint probability density function

$$f_{XY}(x,y) = \frac{1}{2\pi\sigma^2}e^{-\frac{(x-A)^2+y^2}{2\sigma^2}}.$$

- (a) (5%) Please find the marginal probability density function  $f_{\gamma}(y)$ .
- (b) (15%) Please find the probability density function  $f_R(r)$  of  $R = \sqrt{X^2 + Y^2}$  in terms of the modified Bessel function of the first kind of zero order  $I_0(x) = \frac{1}{2\pi} \int_0^{2\pi} e^{x\cos\theta} d\theta$ .
- 6. (20%) We consider the maximum-ratio combining scheme. We have received a set of noisy signals  $\{x_j(t)\}_{j=1}^N$ , where  $x_j(t)$  is defined by

$$x_j(t) = s_j(t) + n_j(t), j = 1, 2, ..., N.$$

The signal components  $s_j(t)$  are locally coherent, that is,  $s_j(t) = z_j m(t)$ , j = 1, 2, ..., N where  $z_j$  are positive real numbers, and m(t) denotes a message signal with unit power. The noise signals  $n_j(t)$  have zero mean and variance  $\sigma_j^2$ , and they are statistically independent. The output of the linear combiner is defined by  $x(t) = \sum_{j=1}^{N} \alpha_j x_j(t)$  where the parameters  $\alpha_j$  are the combiner coefficients to be determined.

- (a) (10%) Show that the output signal-to-noise ratio is  $(SNR)_O = (\sum_{j=1}^N \alpha_j z_j)^2 / \sum_{j=1}^N \alpha_j^2 \sigma_j^2$ .
- (b) (10%) Please show that the optimum values of the combiner's coefficients to maximize the output signal-to-noise ratio are  $\alpha_j = z_j / \sigma_j^2$ . (Hint: Schwarz inequality)