TSKS04 Digital Communication Continuation Course Solutions for the exam 2015-03-16

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Answer:

This is a two-dimensional situation in which the PSD is given by

$$R_s(f) = \frac{1}{T} \begin{pmatrix} \Psi_1^*(f) & \Psi_2^*(f) \end{pmatrix} \begin{pmatrix} R_{S_1,S_1}[fT] & R_{S_2,S_1}[fT] \\ R_{S_1,S_2}[fT] & R_{S_2,S_2}[fT] \end{pmatrix} \begin{pmatrix} \Psi_1(f) \\ \Psi_2(f) \end{pmatrix}.$$

Since the symbols are equally probable and independent, we get

$$R_{S_1,S_1}[fT] = R_{S_2,S_2}[fT] = \frac{P}{2}$$

 $R_{S_2,S_1}[fT] = R_{S_1,S_2}[fT] = 0.$

Notice that the total signal variance P is divided equally between the two dimensions.

Moreover, direction computation of Fourier transforms give

$$\Psi_{1}(f) = j \frac{T}{2} e^{-j\pi fT} \left(e^{-j\pi f_{c}T} \operatorname{sinc}((f+f_{c})T) - e^{+j\pi f_{c}T} \operatorname{sinc}((f-f_{c})T) \right)$$

$$\Psi_{2}(f) = \frac{T}{2} e^{-j\pi fT} \left(e^{-j\pi f_{c}T} \operatorname{sinc}((f+f_{c})T) + e^{+j\pi f_{c}T} \operatorname{sinc}((f-f_{c})T) \right).$$

(See A.3.1 in the extra course material for details).

By multiplying everything together, according to the formula above, we get

$$R_{s}(f) = \frac{1}{T} \frac{P}{2} \left(\frac{T}{2} \right)^{2} \left(\left| e^{-j\pi f_{c}T} \operatorname{sinc}((f+f_{c})T) - e^{+j\pi f_{c}T} \operatorname{sinc}((f-f_{c})T) \right|^{2} + \left| e^{-j\pi f_{c}T} \operatorname{sinc}((f+f_{c})T) + e^{+j\pi f_{c}T} \operatorname{sinc}((f-f_{c})T) \right|^{2} \right)$$

$$= \frac{PT}{4} \left(\operatorname{sinc}^{2}((f+f_{c})T) + \operatorname{sinc}^{2}((f-f_{c})T) \right).$$

2

Answer:

The different questions can be answered independently.

a. As mentioned on page 218, ZF requires that L > K - 1. This can also be expressed as $L \ge K$. Also for these cases, it can happen that ZF does not exist because $\mathbf{U}^H\mathbf{U}$ is non-invertible. A trivial example is when \mathbf{U} is an all zero matrix.

It is fine to just give an example of a U where U^HU is non-invertible.

It also does not exist when $\mathbf{u}_0 = \mathbf{0}$.

- b. Whenever $\mathbf{P}_{I}^{\perp}\mathbf{u}_{0} = \mathbf{u}_{0}$, ZF will both cancel the ISI (if there is any) and not suffer from noise amplification. This occurs, for example, when \mathbf{u}_{0} is orthogonal to the other columns of \mathbf{U} . (The desired signal in Figure 5.9 is orthogonal to the interference subspace.) This one way to motivate optimality.
 - Alternatively, if minimizing the MSE is the optimal thing, then ZF is optimal when it equals MMSE (with happens as said in (d)).
- c. The noise enhancement factor is as largest when ZF and MF are equal, which gives a factor 1.
- d. By comparing (5.42) for MMSE and (5.30) for ZF, one can see that these conditions are equal when $C_{\mathbf{w}} = \mathbf{0}$. In other words, when there is no additive noise.

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Answer:

Part (b) can be solved independently from (a) and (c).

a. Let \mathbf{s}_i for i = 1, ..., 8 denote the 8 signal points in the figure. These can be numbered in any way, but one possibility is

$$\mathbf{s}_1 = (-A \ A)^T$$

$$\mathbf{s}_2 = (0 \ A)^T$$

$$\mathbf{s}_3 = (A \ A)^T$$

$$\mathbf{s}_4 = (-A \ 0)^T$$

$$\mathbf{s}_5 = (A \ 0)^T$$

$$\mathbf{s}_6 = (-A \ -A)^T$$

$$\mathbf{s}_7 = (0 \ -A)^T$$

$$\mathbf{s}_8 = (A \ -A)^T$$

Theorem 3.4.1 defines the hypotheses in (3.29) and provides the ML decision rules in (3.30). In particular, one can obtain the log-likelihood ratios for each hypothesis:

$$H_i: rac{1}{\sigma^2} \left(<\mathbf{y}, \mathbf{s}_i > -rac{\|\mathbf{s}_i\|^2}{2}
ight).$$

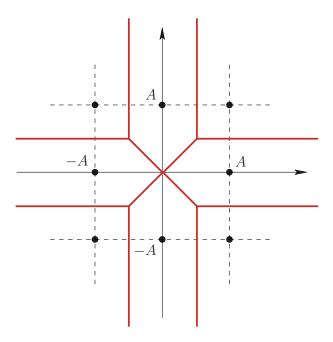
One picks the hypothesis with the highest value.

Alternatively, one can pick the hypothesis that minimizes the distance:

$$H_i: \|\mathbf{y} - \mathbf{s}_i\|$$

These can decision rules are proved using the technique outlined in the book.

b. The ML decision regions are given by the red lines:



As pointed out in Theorem 3.4.1, these regions are based on the "minimum distance rule".

c. The log-likelihoods are computed as described above. The values become

$$H_{1} = -38/15 \frac{A^{2}}{\sigma^{2}}$$

$$H_{2} = -51/30 \frac{A^{2}}{\sigma^{2}}$$

$$H_{3} = -56/30 \frac{A^{2}}{\sigma^{2}}$$

$$H_{4} = -25/30$$

$$H_{5} = -1/6 \frac{A^{2}}{\sigma^{2}}$$

$$H_{6} = -2/15 \frac{A^{2}}{\sigma^{2}}$$

$$H_{7} = 21/30 \frac{A^{2}}{\sigma^{2}}$$

$$H_{8} = 8/15 \frac{A^{2}}{\sigma^{2}}$$

The highest value is given by H_7 , which is $(0 - A)^T$.

4

Answer:

a. The sampled autocorrelation sequence is defined in (5.6) as

$$h[m] = \int p(t)p^*(t - mT)dt.$$

From the problem formulation we see that T=1.

We now get

$$h[0] = \int p(t)p^*(t)dt = 2^2 + (1/2)^2 + (-1/2)^2 = 4.5$$

$$h[\pm 1] = \int p(t)p^*(t \mp 1)dt = 2 \times 1/2 + (1/2) \times (-1/2) = 3/4$$

$$h[\pm 2] = \int p(t)p^*(t \mp 2)dt = 2 \times (-1/2) = -1$$

$$h[\pm k] = \int p(t)p^*(t \mp 3)dt = 0$$

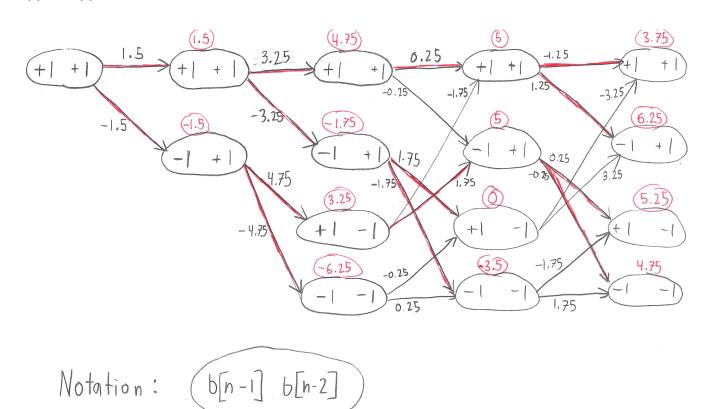
for $k \geq 3$.

- **b.** The memory is L=2 since this is the largest value on m for which $h[\pm m] \neq 0$.
- **c**. The branch metric (5.13) becomes

$$\lambda_n(b[n], (b[n-1], b[n-2])) = b[n]z[n] - 2.25 - b[n]\frac{3}{4}b[n-1] + b[n]b[n-2]$$
(1)

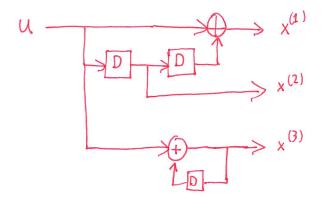
for the values on h[m] above, since everything is real-valued and |b[n]| = 1 for BPSK.

For simplicity, we neglect the constant -2.25. See the finalized Viterbi algorithm below. If we terminate the algorithm by selecting the state with the highest end probability, then the bit sequence will be b[1] = 1, b[2] = 1, b[3] = 1, b[1] = -1.



5

a. The encoder can look like this:



b. This input sequence can be expressed in D-transform as $U(D) = D + D^3$. The codeword will then be

$$U(D)G(D) = \begin{pmatrix} D + D^5 & D^2 + D^4 & \frac{D + D^3}{1 + D} \end{pmatrix} = \begin{pmatrix} D + D^5 & D^2 + D^4 & D + D^2. \end{pmatrix}$$

The second equality follows from the fact that

$$\frac{D+D^3}{1+D} = D\sum_{i=0}^{\infty} D^i + D^3 \sum_{j=0}^{\infty} D^j = D + D^2$$

since the two summations cancel out each other term of order D^3 or larger.

Hence, the first output is $x^1 = (0\,1\,0\,0\,0\,1,\,0,\,0,...)$, the second output is $x^2 = (0\,0\,1\,0\,1\,0,\,0,\,0,...)$, and the third output $x^3 = (0\,1\,1\,0\,0\,0,\,0,\,0,...)$.

c. Yes, this is possible. For example, the input u = (100, ...) will give $x^3 = (1, 1, ...)$. The input has Hamming weight 1, but the output has infinite Hamming weight since x^3 has that.