EE1206 Linear Systems and Signal Processing Exam 2

Time: 120 min Maximum score: 30 points

Instructions

- This is a closed book closed notes exam. No phones, calculators, or laptops can be used during the exam. The notation used is the same as from class.
- There are 6 questions for 6 marks each. The maximum score you can obtain is 30 points (and not 36 marks)
- There are no negative marks. Use a blue or a black pen to answer the questions.
- 1. (6 pts) (a) Let X(t) be a wide sense stationary random process with autocorrelation $R_{xx}(\tau)$. Let Y(t) = aX(ct) for some constants $a, c \neq 0$. Is Y WSS (and if so, what is its autocorrelation function) ?
 - (b) Let w(t) be a (deterministic, non random) signal, and C a random variable. Identify the conditions on C and w(t) such that X(t) = Cw(t) is WSS.
- 2. (6 pts) Let X(t) be a WSS random process with autocorrelation $R_{xx}(\tau)$ and PSD $S_{xx}(s)$.
 - (a) Explain why $R_{xx}(\tau) = 2 \int_0^\infty S_{xx}(s) \cos 2\pi s \tau ds$.
 - (b) Show that

$$R_{xx}(0) - R_{xx}(\tau) \ge \frac{1}{4^n} \left(R_{xx}(0) - R_{xx}(2^n \tau) \right).$$

You may find the inequality $1 - \cos \theta \ge (1 - \cos 2\theta)/4$ helpful.

- (c) Let Y(t) be the output of an LTI system with input process X(t). Suppose it is known that the PSDs S_{xx} of the input and S_{yy} of the output satisfy $S_{xx}(s)S_{yy}(s)=0$ for all s. What can you say about the output process Y(t)?
- 3. (6 pts) (a) A randomly phased sinusoid $X(t) = \cos(2\pi f t + \Theta)$, with Θ uniformly distributed in $(0, 2\pi)$ is given as input to an LTI system with frequency response H(s) (assume the LTI system has a real impulse response, as usual). What is the autocorrelation function of the output of this filter?
 - (b) A discrete-time white noise random process is given as input to an LTI system with impulse response h. We know that h(0) = 1, and that all impulse response values h(m) are nonzero. Can the output power of the filter ever be less than the input power? Justify with a proof or counterexample.
- 4. (6 pts) A zero mean discrete time WSS random process Y(n) with PSD $S_{yy}(s) = 2 2\cos(2\pi s)$ is corrupted by zero mean white noise Z(n) with power $E(Z(0)^2) = 1$; so that we have measurements of X(n) = Y(n) + Z(n).
 - (a) What is the frequency response of the optimal Weiner filter to recover Y(t)? Plot this frequency response as a function of the frequency s, for $-1/2 \le s \le 1/2$. Is this a low-pass or a high-pass filter?
 - (b) Does the power of the estimate \hat{Y} match the power in Y, i.e do we have $E(Y(0)^2) = E(\hat{Y}(0)^2)$?
- 5. (6 pts) Let f(t) be a bandlimited signal whose Fourier transform satisfies $|\mathcal{F}f(s)|=0$ for $|s|\geq 1$. According to the sampling theorem, one has to sample f(t) with $\mathrm{III}_{1/2}(t)$ to reproduce the signal without aliasing.

You try to test this out in the lab, but there is something wrong with the ideal sampler — it can *only* sample using $\mathrm{III}_1(t)$. This will inevitably cause aliasing, but you think you can devise a scheme to somehow reconstruct f(t) without aliasing.

This problem explores how this can be done by using this faulty sampler and some filter h(t).

- (a) Let $h(t) = -1/(\pi i t)$. If g(t) = (f * h)(t), express G(s) in terms of F(s) only. Hint: Consider the cases s = 0, s > 0 and s < 0.
- (b) Suppose you sample f(t) with $\mathrm{III}_1(t)$ to yield y(t). For the case of 0 < s < 1, express Y(s) in terms of F(s) and F(s-1) only.
- (c) Suppose you sample g(t) with $\mathrm{III}_1(t)$ to yield x(t). For the case of 0 < s < 1, express X(s) in terms of F(s) and F(s-1) only.
- (d) Using the two equations you have from parts (b) and (c), show how you might reconstruct f(t) without aliasing.
- 6. (6 pts) Earlier in the course we discussed the uncertainty principle for continuous time signals, the main idea being that time width and frequency width are inversely related. A similar relationship holds for discrete signals, as we will see in this problem. Let f be an N-length vector and $F = \mathcal{F}f$ be its DFT. Let $\phi(f)$ indicate the number of nonzero coordinates in f, and similarly let $\phi(\mathcal{F}f)$ be the number of non-zero coordinates in $F = \mathcal{F}f$. Then the discrete uncertainty principle states that

$$\phi(f)\phi(\mathcal{F}f) \geq N.$$

In other words, the time width $(\phi(f))$ and frequency width $(\phi(\mathcal{F}f))$ are inversely related.

Below is an outline of the proof of this uncertainty principle. Complete the proof by briefly justifying the statements (a)-(f) in the proof.

Proof: Let the vector s denote the sign of f:

$$s[n] := \begin{cases} 1 \text{ if } f[n] > 0, \\ -1 \text{ if } f[n] < 0, \\ 0 \text{ if } f[n] = 0, \end{cases}$$

and let (f,s) denote the dot product of the vectors f and s. Assume that $f \neq 0$. Defining s[n] gives us the advantage of expressing |f[n]| as |f[n]| = f[n]s[n].

Now

$$\begin{aligned} \max_{m} |F[m]| & \stackrel{(a)}{\leq} & \sum_{n=0}^{N-1} |f[n]| \stackrel{(b)}{=} & (f,s) \\ & \stackrel{(c)}{\leq} & (f,f)^{1/2} (s,s)^{1/2} \stackrel{(d)}{=} \frac{1}{\sqrt{N}} (F,F)^{1/2} (s,s)^{1/2} \\ & \stackrel{(e)}{=} \frac{1}{\sqrt{N}} (F,F)^{1/2} \phi(f)^{1/2} \stackrel{(f)}{\leq} \frac{\max_{m} |F[m]|}{\sqrt{N}} \phi(F)^{1/2} \phi(f)^{1/2}. \end{aligned}$$

Canceling out $\max_{m} |F[m]|$ on both sides, we are left with the desired inequality

$$\phi(f)\phi(F) \geq N$$
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