

Signal Processing and Speech Communication Laboratory—Prof. G. Kubin  
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## Exam *Adaptive Systems* on 2008/12/15

Name

MatrNr.

StudKennz.

Exam duration: 180 minutes

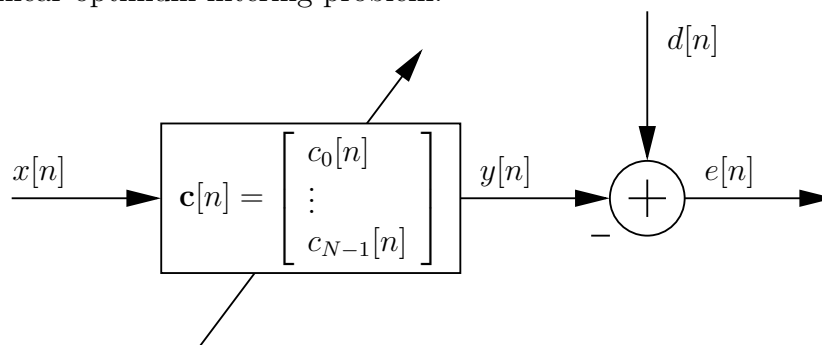
Maximum points: 100

Allowed material: mathematical formulary, calculator

**You have to return this document after the exam. Good luck!**

### Problem 1 (32 Points)

Consider a linear optimum filtering problem:



The *Leaky Gradient Search Method*

$$\mathbf{c}[n+1] = \beta \mathbf{c}[n] + \mu (\mathbf{p} - \mathbf{R}_x \mathbf{c}[n])$$

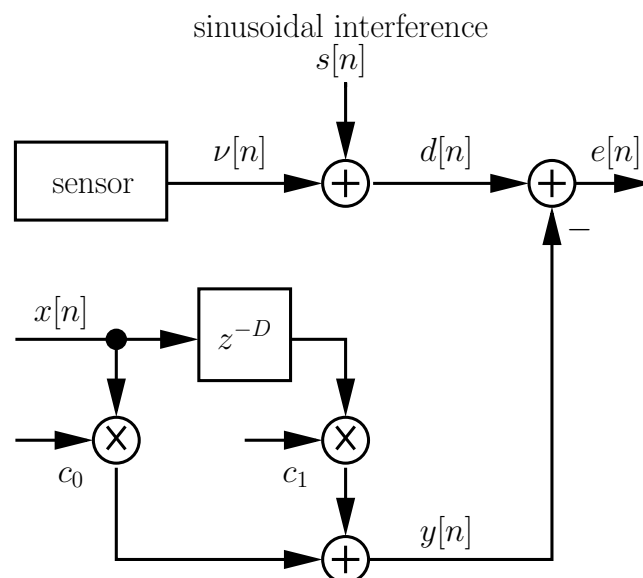
is used to adapt the coefficients of the adaptive filter.  $\mathbf{R}_x$  is the auto-correlation matrix of the tap-input vector  $\mathbf{x}[n] = [x[n], \dots, x[n-N+1]]^T$ , and  $\mathbf{p}$  is the cross-correlation vector between the desired signal  $d[n]$  and the tap-input vector.

- (a) Assume convergence of the algorithm. Where does the algorithm converge to?
- (b) Derive a condition on the step size  $\mu$  and on the leakage parameter  $\beta$  that ensures convergence, and specify the possible range for  $\mu$  given  $\beta$ .
- (c) The *Principle of Orthogonality* says:
- “The estimate  $y[n]$  of the desired signal  $d[n]$  is optimal in the sense of a minimum mean squared error if, and only if, the error  $e[n]$  is orthogonal to the input  $x[n-m]$  for  $m = 0 \dots N-1$ ”.

(i) Prove this principle. (ii) Show whether this principle applies for the solution found by the leaky gradient search method, or not.

## Problem 2 (35 Points)

Consider the following adaptive *sinusoidal-interference canceler*:

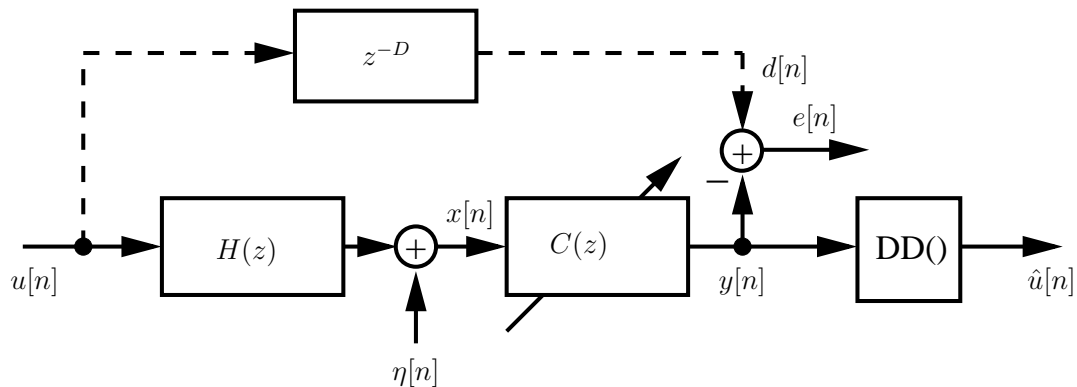


The input signal is given by  $x[n] = \cos(2\pi \frac{f_{ac}}{f_s} n + \varphi)$ . The frequency  $f_{ac} = 50$  Hz, and the sampling frequency  $f_s = 12000$  Hz. The phase offset  $\varphi$  is considered as a random variable uniformly distributed over  $(-\pi, \pi)$ . The sensor signal  $\nu[n]$  is unknown, but it is uncorrelated with  $x[n]$ . The sinusoidal interference is  $s[n] = \frac{1}{2} \cos(2\pi \frac{f_{ac}}{f_s} n + \varphi - \frac{\pi}{4})$ . The delay in the adaptive filter is  $D \in \mathbb{N}$ .

- (a) Determine the auto-correlation matrix  $\mathbf{R}_x$  of the tap-input vector  $\mathbf{x}[n] = [x[n], x[n-D]]^T$ . (Do not just give the result, but show how to find the result by integration.)
- (b) Determine the eigenvalues  $\lambda_i$  of  $\mathbf{R}_x$ , and give the condition number  $\kappa = \frac{\lambda_{max}}{\lambda_{min}}$ . Find the (smallest) delay  $D \in \mathbb{N}$  that minimizes  $\kappa$ . Explain, what theoretical/practical advantage such a selection of  $D$  may bring.
- (c) Determine the cross-correlation vector  $\mathbf{p}$  between the signal  $d[n]$  and the tap-input vector  $\mathbf{x}[n] = [x[n], x[n-D]]^T$ .
- (d) Determine the Wiener-Hopf solution for the two coefficients  $c_0$  and  $c_1$ . (Take the  $D$  found before or use a general  $D$  if you have not found one before.)
- (e) Show whether your found solution for  $c_0$  and  $c_1$  achieves perfect cancelation  $e[n] \stackrel{?}{=} \nu[n]$ , or not.

### Problem 3 (33 Points)

Consider the following channel-equalization problem:



The symbols to be transmitted are  $u[n] \in \{-1, +1\}$ , originate from a stationary process, have zero mean, and are uncorrelated with each other  $E\{u[n]u[n-m]\} = 0, \forall m \neq 0$ . The channel noise  $\eta[n]$  is white, stationary, and uncorrelated with the data  $E\{\eta[n]u[n-m]\} = 0, \forall m$ .  $H(z)$  is the discrete-time FIR model of the communication channel,  $C(z)$  is an FIR equalizer, and the decision device (DD) is

$$\text{DD}(y[n]) = \begin{cases} 1, & y[n] \geq 0 \\ -1, & y[n] < 0 \end{cases}.$$

(a) Using vector/matrix notation, derive the equation to obtain a *minimum mean squared error (MinMSE) equalizer* with  $N$  coefficients for a general FIR channel model and a given data variance  $\sigma_u^2$  and a given channel noise variance  $\sigma_\eta^2$ . Define all used vectors and matrices.

(b) Let the discrete-time FIR model of the communication channel be

$$H(z) = h_0 + h_1 z^{-1} + h_2 z^{-2} + h_3 z^{-3} = 1 + 1/2 z^{-1} + 1/3 z^{-2} + 1/4 z^{-3}.$$

Without an equalizer, what delay  $D$  should be used to minimize the intersymbol interference (ISI)? Derive whether the channel's eye is open or closed.

(c) For the above given channel, a delay of  $D = 0$ , and a noise variance of  $\sigma_\eta^2 = 1$ , compute the coefficients of a MinMSE equalizer with  $N = 2$  coefficients.

(d) Derive whether the eye of the cascade of the given channel and the obtained 2-coefficient MinMSE equalizer is open or closed.