Paper Number(s): E3.08

ISE3.17

IMPERIAL COLLEGE OF SCIENCE, TECHNOLOGY AND MEDICINE UNIVERSITY OF LONDON

DEPARTMENT OF ELECTRICAL AND ELECTRONIC ENGINEERING **EXAMINATIONS 2002**

EEE/ISE PART III/IV: M.Eng., B.Eng. and ACGI

ADVANCED SIGNAL PROCESSING

Wednesday, 1 May 10:00 am

There are FIVE questions on this paper.

Answer TWO of the questions 1, 2, 3 and ONE of the questions 4, 5.

Corrected Copy

4.1.

Time allowed: 3:00 hours

Examiners responsible:

First Marker(s):

Mandic, D.P.

Second Marker(s): Ward, D.B.

Special instructions for invigilators:	None	
Information for candidates:	None	

- 1. Consider a random process $Z(t) = X(t) \pm Y(t)$, where X(t) and Y(t) are also random processes. The \pm sign means that X(t) and Y(t) can be either added or subtracted.
 - a) Define random variables Z_1 and Z_2 by $Z_1 = X(t_1) \pm Y(t_1)$ and $Z_2 =$ $X(t_1+\tau)\pm Y(t_1+\tau)$. Find the autocorrelation function $R_Z(\tau)=E\left[Z_1Z_2\right]$.

b) If the two random processes Z_1 and Z_2 are statistically independent, and zero mean, what is the resulting autocorrelation function $R_Z(\tau)$? An important consequence of this result arises in the extraction of periodic signals from random noise. Suppose the autocorrelation function of the desired signal X(t) is $R_X(t) = \frac{1}{2}A^2\cos\omega\tau$. Suppose also that there is a zero-mean random noise signal Y(t) that is statistically independent of the signal and has $R_Y(\tau) = B^2 e^{-\alpha|\tau|}$. Find the autocorrelation function of X(t) + Y(t).

-Rx(2). c) A radar system is transmitting a signal X(t). The signal is then returned from the target attenuated and delayed in time, and can be represented as Y(t) = aX(t-T) + N(t), where a < 1, T is the time delay and N(t) is the noise which is statistically independent of the signal. Both the signal and noise have zero mean. Find the crosscorrelation function between the transmitted signal X(t) and received signal Y(t).

[8]

2.	a)	Define the likelihood function for a random signal x. State the likelihood
		function for the random variable $x(0) = A + w(0)$, where A is the DC level,
		and $w \sim \mathcal{N}(0, \sigma^2)$.

[4]

b) Define the curvature of the log-likelihood function. What is the curvature for the random process x(0) of part a)? What does the curvature give information about?

[4]

c) Define the Cramer-Rao Lower Bound (CRLB) (scalar parameter).

[8]

d) What is the minimum attainable variance for the random process x(0) of part a)?

3.	a)	State the equation of the second order autoregressive process $AR(2)$.	
			[2]
		Derive the autocorrelation function of this process.	
			[4]
		What is the stability condition for this process (stability triangle)?	
			[6]
	b)	Consider the process $z_t = 0.75z_{t-1} - 0.5z_{t-2} + a_t$, where a_t is white no Is the process z_t stable?	ise.
			[2]
		For z_t , state	
		i) the Yule-Walker equations.	
			[3]
		ii) the spectrum.	[3]

4. The least mean square (LMS) algorithm for a linear finite impulse response (FIR) adaptive filter is given by

$$e(k) = d(k) - y(k)$$

$$y(k) = \mathbf{x}^{T}(k)\mathbf{w}(k)$$

$$\mathbf{w}(k+1) = \mathbf{w}(k) + \eta e(k)\mathbf{x}(k)$$

where e(k) is the output error of the filter, d(k) is the desired signal, y(k) is the output of the filter, $\mathbf{x}(k)$ is the tap input vector to the filter, $\mathbf{w}(k)$ is the weight vector, η is the learning rate, and $(\cdot)^T$ denotes transposition.

a) Derive the learning rate of the normalised least mean square (NLMS) algorithm by expanding the error e(k+1) using Taylor series around e(k) and setting e(k+1) = 0.

[12]

b) Explain the difference between the learning rate η and the learning rate of the NLMS $\eta_{NLMS} = 1/\parallel \mathbf{x}(k) \parallel_2^2$.

[4]

c) Why can we say that the NLMS algorithm minimises the so-called *a poste-* riori output error?

mere C(10)= 1(10)- >(10)

- 5. A simple nonlinear finite impulse response (FIR) adaptive filter is shown in Figure 5.1. Φ is a saturation nonlinear function. The cost function for this filter is given by $E(k) = \frac{1}{2}e^2(k)$.
 - a) Give the reasons for the structure of Figure 5.1 also being called a dynamical neuron.

[4]

b) Derive the weight update equation $\Delta \mathbf{w}(k) = -\eta \nabla_{\mathbf{w}} E(k)$ for the filter above.

[8]

c) What is the difference between this filter and the standard adaptive FIR filter? Why does this structure perform generally better on filtering of nonlinear signals?

[4]

d) If the nonlinear function Φ is the arctan function, explain the effect of saturation in the output. What is the effect of saturation—type nonlinearity on the output magnitude range?

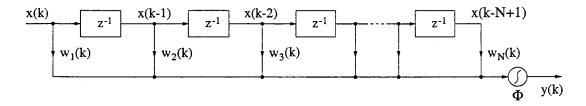


Figure 5.1. A nonlinear FIR filter

So lutions

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a) 7(6) = X16) ty(6)

Z1= X161) = Y161) = 72= / (k1+2) = (k1+2)

R=(T) = [= [= 1 = = [(x, t) (x, t)]

=モレベルナイルナメルをナイル

= Rx(T)+ Ry(T) ± Rxy(T) ± Rxx(T)

b) In the case of statistically independent and zero-mean vandom processes, both the cross-correlation functions Rxy (5) and Rxx(6) banish and

Raci) = Rxci) + Ryci)

=> P == R (XI HAYLA) (T) = { 1/2 A 2 COSLUT + B 2 E OLIV)

e) YIt) = 2x(+-T)+11(t)

PXY(T)= E(XIE) TIE+TO)=

= E(0x(t)X(++-+)+X(t) H(++-)]

= a Rx(T-T) +Rxw(T)

or me d'qual and avise are statistically

Pxy (T)=alx(T-T)

$$p(x(0); A) = \frac{1}{\sqrt{2\pi62}} \exp(-\frac{1}{262}(x(0) - A)^2)$$

The curvature gives information about the " sharpness" of the hibelihood function, rie information a on how accurately we con estimate the unknown parameter.

$$\frac{\partial \ln(b(x(0); y))}{\partial \mu(b(x(0); y))} = \frac{1}{2} (x(0) - 4)^{2}$$

$$= \frac{1}{2} \ln(b(x(0); y)) = \frac{1}{2} (x(0) - 4)^{2}$$

$$= \frac{1}{2} \ln(b(x(0); y)) = \frac{1}{2} \ln(x(0) - 4)^{2}$$

It is assumed that the PDF p(se; B) satisfies the

Furthermore, an unbiated estimater may be found that attains the bound for to it

2) This is the MUV estimator, $\theta = g(x)$ and the minimum variance is 1/10).

d) From the CPLB var (Â) > 62, +A. 3)~ ~ 2t = \$126-1+ \$226-2+ at On fracoeticients at -> white woill the stability triangle is defined by \$1+\$2<1 φ₁- φ₁ < 1</p> -1 < 22 <1 Process Zt= 0.7576-1-0.576-2+96 1's stable. For the AR(S) Provess Sn = 0184-1+ 4284-2 , k>0 gh= 0.757+= 0.5 Hene gu= 0.75 gu-1 - 0.5 gh-2 Y-wegnations for the MR (2) probis are $g_1 = \phi_1 + \phi_2.g_1$ => $g_1 = 0.75 = 0.5g_1$ $g_2 = \phi_1g_1 + \phi_2$ $g_2 = 0.75g_1 - 0.5$ P(f)= 11-41-824+ 42-8, 44+15 2 62 1+ \$12+\$22-2\$1(1-\$2)cos 2uf-2\$2 cos(4 uf)

5 2 13 c

e(hti) = elh) + Z gelh) awith) + Z Z Z Z Dwilldowjeh

· Dw; (h) Dw; (h)

The se could and higher-order + h.o.t. terms (h-o-t) vanish due to the linearity of the filter.

From the LMS we have

and hence

ell) = e(h) [1- M = a(h)]

e(h+1)=0 for $M = \frac{1}{\| x(h) \|_{2}}$

- b) Muchs adopts dynamically according to the "top input power" 11 X (W)1/2, whereas of for the LMS is static
- c) we wint wife e(k+1) using the variables from the time instant le. Therefore, ue minimité the "a posteriori" error.

a) It is has nevery, worlinearity and represents a "dynamical synapse"

6/6

12(1) = \$ (xr (1) m (11)) m (11-1) = m (11) - 2 m (11)

 $\frac{1}{2} \sum_{n=1}^{\infty} \frac{1}{2} \sum_{n=1}^{\infty} \frac{1}$

- c) this algorithm is a variant of the LMS for the case of the woulinear from at the output. Hence it is suitable for filtering of woulinear imputs.
 - d) the imput tignals with the mognitude range greater of equal to the range of the mactan second "dipped". Hence the distortion of each signals