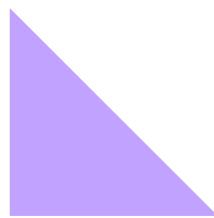
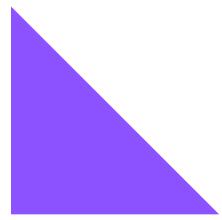


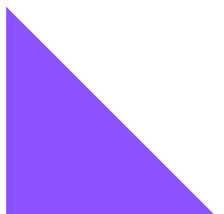
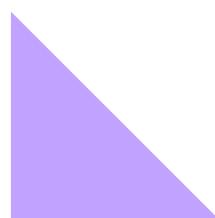


COLLATE



Digital Communication

Notes
Unit 3



Syllabus

UNIT - III

Analysis of digital receiver, Prediction Filter, Design and Property of Matched filter, Correlator Receiver, Orthogonal Signal, Gram-Schmidt Orthogonalization Procedure, Maximum likelihood receiver, Coherent receiver design, Inter Symbol Interference, Eye Pattern.

UNIT III

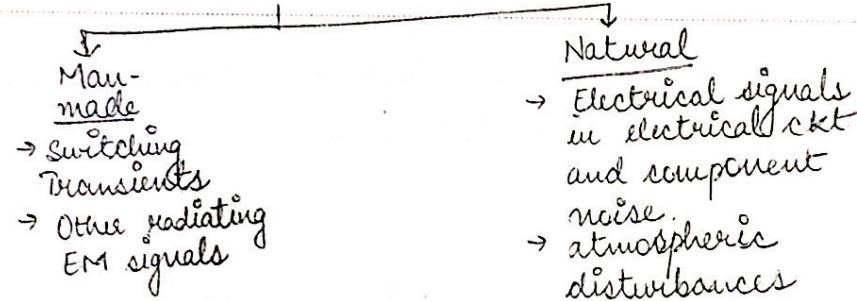
Analysis of Digital Receivers

1 → Earlier :

- (1) analog w/fis → digital data
 - (2) ← electrical w/fis or symbols
 - (2) detection of symbols and recovery of digital data from received w/f.
- II. Problem in recovering.

NOISE

2 → NOISE (unwanted electrical sig.)



can be eliminated through filtering and shielding.

3 → Noise that cannot be eliminated

→ Thermal or Johnson noise :

caused by random motion of e^os in all components eg: resistors, wires, etc

Thermal noise = zero mean Gaussian random process

→ in a digital tx^o sys, one of M possible wfs are txed.

$$s_i(t) \in \{s_1(t), s_2(t), \dots, s_M(t)\} \text{ for } 0 \leq t \leq T$$

At the Rx:

$$x(t) = s_i(t) + \frac{n(t)}{\sqrt{2}} \text{ causes a corruption in the}$$

$x(t)$ → received sig. original signal

$s_i(t)$ → original txed sig

$n(t)$ → noise (AWGN)

A → Additive (adds 'noise' onto the signal)

W → White (the noise has a flat spectrum
 $\rightarrow \text{PSD.} = \frac{N_0}{2} (\text{W/Hz}) \text{ from } -\infty \text{ to } \infty \text{ Hz}$)

G → Gaussian (amplitude of the noise voltage fluctuations follows a Gaussian distribution)

→ Now, Noise ⇒ Random process

∴ it can be described in terms of :-

→ Mean (or avg.) value

→ Variance (or std. deviation)

→ pdf (prob. density fun^o)

→ Now, noise is assumed to be AWGN, so its pdf is :

$$p(n) = \frac{1}{\sqrt{2\pi}} e^{-\frac{1}{2}\left(\frac{n}{\sigma}\right)^2}$$

$n(t)$ has mean = 0

& variance = σ^2

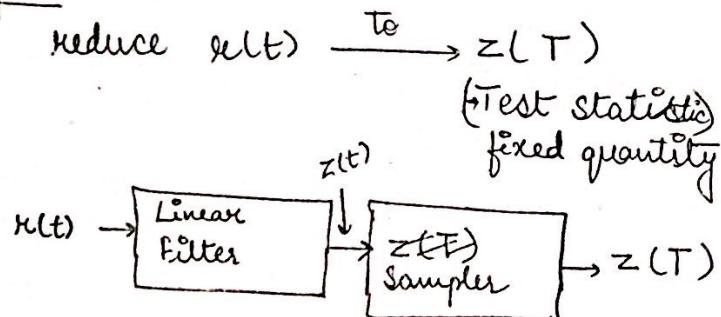
(2)

- The expression for pdf shows that the noise voltage amplitudes are distributed according to Gaussian distribution.
- The most probable amplitudes ^{of noise} are those \in small +ve or -ve values.
- Since, $n(t) \rightarrow$ randomly varying quantity,
So, $x(t) \rightarrow$ will also be randomly varying quantity.
∴ there will be uncertainty in finding the value of $x(t) \Rightarrow$ lead to loss in recovery of digital data. ^{and not "what"}
- The funⁱ of a digital Rx is to find which of the signalling wfs (or symbols) was txed in any given signalling interval.
 - * Analog Rx \rightarrow what signal was txed
 - * Digital Rx \rightarrow which , "
- In a digital Rx,
set of txed signals $\{s_1(t), s_2(t), \dots, s_M(t)\}$ is known as pilot.
- The Rx knows what to expect & it has to find out \in sig. was is being txed in any given time interval.
- Digi. Rx has superior noise performance over analog Rx.

→ How to detect a Digital Signal?

2 basic steps :

Step 1 :



Step 2 : Compare $z(T)$ \in a reference value or threshold γ

(This will find out which sig. was txed)

$$z(T) = \begin{cases} s_1 & \text{if } z(T) > \gamma \\ s_2 & \text{if } z(T) < \gamma \end{cases}$$

if $z(T) > \gamma$, $\Rightarrow s_1(t)$ is considered to have been txed

2 if $z(T) < \gamma$, $\Rightarrow s_2(t)$

* Sometimes, due to noise, we can have error i.e wrong estimation of signal.

Since, $n(t)$ = rand. var $\Rightarrow r(t)$ & hence $z(t)$ or $z(T)$ is also random.

∴ $z(T)$ can be considered by its mean or variance or pdf's.

→ The analysis of digit Rx begins \in the concept of matched filter.

⇒ MATCHED FILTER (Design and Property)

- It is a linear time-invariant (LTI) filter that leads to the optimum detection of a signal wff that is immersed in AWGN.
- * optimum = min. prob. of an error occurring
- Here, we match the impulse response of the filter $(h(t))$ or $(h(t))$ to the sig. wff. $(s_i^o(t))$
- The filter is designed to detect the presence of signal $s_i^o(t)$ (s_i^o is buried in the noisy recd signal $x_i(t)$).
- More precisely; Matched filter is designed to maximise the SNR at the filter o/p, for a given wff $s_i^o(t)$ at sampling instant $t = T$.

→ $x_i(t)$ is filter o/p

$$x_i(t) = s_i^o(t) + n_i(t)$$

(original signal) (noise)

→ Since, the filter is linear
⇒ the o/p at $t = T$ is :-

$$z(T) = a_i + n_o$$

↓ ↓
(signal component) (noise component)
Deterministic —

→ Now, the o/p Noise power (variance or avg. power)
 $= \sigma^2$

→ SNR of matched filter at sampling time instant $t=T$
is :-

$$\left(\frac{S}{N}\right)_{t=T} = \frac{\alpha^2}{\sigma^2}$$

Aim: we have to find the optimum filter transfer funⁿ $H_0(f)$ that maximises $(S/N)_{t=T}$. This will minimise the probability of making an incorrect decision i-e to minimise the prob. of error.

→ Expⁿ for SNR can be re-written as :-

$$\left(\frac{S}{N}\right)_{t=T} \leq \frac{\alpha E}{N_0}$$

$(\frac{S}{N})$ dep. on → sig. energy 'E'
and

→ Noise power spectral Density
($N_0/2$)

→ Now, max. value of $\alpha(S/N)$ is at :

$$\boxed{\frac{S}{N} = \frac{\alpha E}{N_0}}$$

Correspondingly, the max. value of $H(f)$, denoted $H_{opt}(f)$ occurs when :

impulse
response
per sec
domain

$$H_{opt}(f) = k S_i^*(f) e^{-j2\pi f T}$$

$k \rightarrow$ an arbitrary const.

$*$ → denotes complex conjugate

$S_i(f) \rightarrow$ Fourier Transform of sig. w/f $s_i(t)$.

(4)

- In time domain, the impulse response of the optimum filter $h_{opt}(t)$ is given by inverse FT of $H_{opt}(f)$ i.e.

$$\begin{aligned} h_{opt}(t) &= \stackrel{\text{IFT}}{\text{or}} \left\{ H(f) \right\} \\ &= \stackrel{\text{IFT}}{\text{or}} \left\{ k S_i^*(f) e^{-j2\pi f T} \right\} \\ &= k s_i(T-t) \end{aligned}$$

- So, the impulse response of the optimum filter is a time-reversed and delayed version of the i/p sig $s_i(t)$.

- In other words, $h_{opt}(t)$ is "matched" to the i/p signal $s_i(t)$.

- An LTI sys defined in this way is termed as a matched filter.

- Use of this filter will result in the reception and detection of digital signals \rightarrow smallest prob. of error.

Property / Summary:

- 1) A filter matched to an i/p sig $s_i(t)$ of duration T is char. by an impulse resp that is time reversed and time delayed

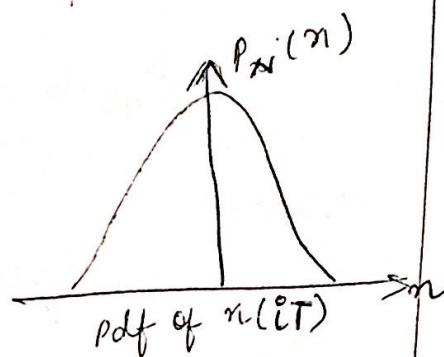
$$h_{opt}(t) = k s_i(T-t)$$

- 2) In freq domain, the matched filt. is char. by a freq resp! i.e a complex conjugate of the FT of i/p sig $s_i(t)$

$$H_{opt}(f) = k S_i^*(f) e^{-j2\pi f T}$$

3 → Max. SNR dep. upon → E [sig. energy]
 → $\left(\frac{N_0}{2}\right)$ [PSD of white noise]

$$\left(\frac{S}{N}\right)_{t=T} = \frac{\alpha E}{N_0}$$



→ Derivation for Prob. of error (P_e) for the Matched filter

P_e for optimum filter is :

$$P_e = \frac{1}{2} \operatorname{erfc} \left[\frac{x_{o1}(T) - x_{o2}(T)}{2\sqrt{2} \sigma} \right] \quad \text{--- (1)}$$

$x_{o1}(T)$ } off of the Rx in the absence of
 $x_{o2}(T)$ } noise $n(t)$.

Here, we have

$$\left[\frac{x_{o1}(T) - x_{o2}(T)}{\sigma} \right]^2_{\max} = \int_{-\infty}^{\infty} \frac{|X(f)|^2}{S_{ni}(f)} df \quad \text{--- (2)}$$

$S_{ni}(f) \rightarrow$ i/p PSD (of noise)

Here, ($S_{ni}(f)$) is considered as AWGN

$$\therefore \text{psd of this noise} = S_{ni}(f) = \frac{N_0}{2} \quad \text{--- (3)}$$

(5)

Hence,

$$\left[\frac{x_{01}(T) - x_{02}(T)}{\sqrt{N_0}} \right]^2_{\max} = \int_{-\infty}^{\infty} \frac{|X(f)|^2}{\left(\frac{N_0}{2}\right)} df.$$

$$= \frac{2}{N_0} \int_{-\infty}^{\infty} |X(f)|^2 df \quad \text{--- (4)}$$

Also, according to Parseval's theorem:

$$\int_{-\infty}^{\infty} |X(f)|^2 df = \int_{-\infty}^{\infty} x^2(t) dt = \int_0^T x^2(t) dt \quad \text{--- (5)}$$

limits 0 to T ($\because x(t)$ exists from 0 to T only)

Now,

we know that:
 $x(t) = x_1(t) - x_2(t)$

\therefore the above eqⁿ (5) becomes:

$$\begin{aligned} \int_{-\infty}^{\infty} |X(f)|^2 df &= \int_0^T [x_1(t) - x_2(t)]^2 dt \\ &= \int_0^T [x_1^2(t) + x_2^2(t) - 2x_1(t)x_2(t)] dt \\ &= \int_0^T x_1^2(t) dt + \int_0^T x_2^2(t) dt - 2 \int_0^T x_1(t)x_2(t) dt \end{aligned} \quad \text{--- (6)}$$

Now,
 $\int_0^T x_1^2(t) dt = E_1$, i.e. energy of $x_1(t)$

only,
 $\int_0^T x_2^2(t) dt = E_2$

and $\int_0^T x_1(t) x_2(t) dt = E_{12}$ (energy due to autocorrelation b/w $x_1(t)$ & $x_2(t)$)

Now, if we choose $x_1(t) = -x_2(t)$, then:

$$E_1 = E_2 = -E_{12} = E$$

Put in eq ⑥

$$\therefore \int_{-\infty}^{\infty} |x(f)|^2 df = [E + E - 2(-E)] = 4E \quad \text{--- (7)}$$

Put ⑦ in ④

$$\Rightarrow \left[\frac{x_{01}(T) - x_{02}(T)}{\tau} \right]_{\max}^2 = \frac{2}{N_0} \cdot 4E = \frac{8E}{N_0}$$

Taking $\sqrt{}$ on b/s

$$\left[\frac{x_{01}(T) - x_{02}(T)}{\tau} \right]_{\max} = 2\sqrt{2} \sqrt{\frac{E}{N_0}} \quad \text{--- (8)}$$

Put eq ⑧ in ①

\therefore prob. of error P_e for matched filter is :-

$$P_e = \frac{1}{2} \operatorname{erfc} \left[\sqrt{\frac{E}{N_0}} \right]$$

\Rightarrow Theorem \rightarrow Orthog. sets are linearly independent.
 (if $S = \{v_1, v_2, \dots, v_n\}$ is an orthog. set of non-zero vectors in an inner prod. space V , then S is linearly indep.)

Corollary \rightarrow If V is an inner prod. space of dimension n , then any orthog. set of n non-zero vectors is a basis for V .

S is a basis of V of dimension n

↑
 It means that we have to check that all the pair wise are orthog. or not.

Eg:

(2) Show that the following set is a basis for \mathbb{R}^4

$$S = \{(2, 3, 2, -2), (1, 0, 0, 1), (-1, 0, 2, 1), (-1, 2, -1, 1)\}$$

dimension
 $= 4$
 2 sets for 4 vectors

(2)

$$v_1 \cdot v_2 = 0$$

$$v_1 \cdot v_3 = 0$$

$$v_1 \cdot v_4 = 0$$

$$v_2 \cdot v_3 = 0$$

$$v_2 \cdot v_4 = 0$$

$$v_3 \cdot v_4 = 0$$

1st you have to verify this
 (check pair-wise orthogonality)

So,

So, S is a basis for \mathbb{R}^4 .

Co-ordinates relative to an orthonormal basis.

If $B = \{v_1, v_2, \dots, v_n\}$ is an orthonorm. basis for V , then the coord. rep of a vector 'w' relative to B is:-

$$w = \underbrace{\langle w, v_1 \rangle}_{\text{dot prod. of } w \text{ & } v_1} v_1 + \langle w, v_2 \rangle v_2 + \dots + \langle w, v_n \rangle v_n \quad (1)$$

① Find the coord. matrix of $w = (5, -5, 2)$ relative to the following orthon. basis for \mathbb{R}^3

$$B = \left\{ \underbrace{\left(\frac{3}{5}, \frac{4}{5}, 0 \right)}_{v_1}, \underbrace{\left(-\frac{4}{5}, \frac{3}{5}, 0 \right)}_{v_2}, \underbrace{\left(0, 0, 1 \right)}_{v_3} \right\}$$

in short: find matrix 'w' for orthon. basis B.

These coordinates are called Fourier coeff. of 'w' relative to B.

Sol^{*}: $\because B$ is orthon. (as $v_1 \cdot v_2 = 0$, $v_1 \cdot v_3 = 0$, $v_2 \cdot v_3 = 0$)

So, to find coord. of 'w', we use eq "①"

$$w \cdot v_1 = (5, -5, 2) \cdot \left(\frac{3}{5}, \frac{4}{5}, 0 \right) = \\ = 3 - 4 + 0 = -1$$

$$w \cdot v_2 = (5, -5, 2) \cdot \left(-\frac{4}{5}, \frac{3}{5}, 0 \right) = -4 - 3 = -7$$

$$w \cdot v_3 = (5, -5, 2) \cdot (0, 0, 1) = 2$$

So, the coord. matrix relative to B is :-

$$[w]_B = [-1 \ -7 \ 2]^T \leftarrow \text{Transpose or } \begin{bmatrix} -1 \\ -7 \\ 2 \end{bmatrix}$$



Major ques"

How to generate an orthonormal Basis?

Sol["]: If we are given a basis $B = \{v_1, v_2, \dots, v_n\}$ for three space \mathbb{R}^n , then, how to generate an orthonormal basis B' ?

Answer:

We use Gram-Schmidt Orthogonalization process (to calc. an orthon. basis B' from B)

(12)

1. Let $B = \{v_1, v_2, \dots, v_n\}$ be a basis for an inner prod space V .

2. Let $B' = \{w_1, w_2, \dots, w_n\}$ where w_i is given by:-

$$w_1 = v_1$$

$$w_2 = v_2 - \frac{\langle v_2, w_1 \rangle}{\langle w_1, w_1 \rangle} w_1$$

$$w_3 = v_3 - \frac{\langle v_3, w_1 \rangle}{\langle w_1, w_1 \rangle} w_1 - \frac{\langle v_3, w_2 \rangle}{\langle w_2, w_2 \rangle} w_2$$

:

$$w_n = v_n - \frac{\langle v_n, w_1 \rangle}{\langle w_1, w_1 \rangle} w_1 - \frac{\langle v_n, w_2 \rangle}{\langle w_2, w_2 \rangle} w_2 - \dots - \frac{\langle v_n, w_{n-1} \rangle}{\langle w_{n-1}, w_{n-1} \rangle} w_{n-1}$$

Here, $B' \rightarrow$ Orthogonal basis for V
it is not yet orthonormal. (it is orthog. right now)

The orthonormal basis is $B'' = \{u_1, u_2, \dots, u_n\}$.
 $u_1, u_2, \dots \rightarrow$ all unit vectors (\because orthonormal)

So,
$$u_i = \frac{w_i}{\|w_i\|}$$

So, B'' is an orthonormal basis for V .

Moreover,

$$\text{Span}\{v_1, v_2, \dots, v_k\} = \text{Span}\{u_1, u_2, \dots, u_k\}$$

for $k = 1, 2, 3, \dots, n$.

Lets do an example.

① Apply Gram-Schmidt Orthog' Process for the following basis for \mathbb{R}^3

$$B = \{(\underbrace{v_1}_{(1,1,0)}, \underbrace{v_2}_{(1,2,0)}, \underbrace{v_3}_{(0,1,2)})\}$$

① Step 1: Find $B' = \{w_1, w_2, w_3\}$

$$\text{so, } w_1 = v_1 = (1, 1, 0) \quad \text{--- (1)}$$

$$w_2 = v_2 - \frac{\langle v_2, w_1 \rangle}{\langle w_1, w_1 \rangle} w_1$$

$$= (1, 2, 0) - \frac{\{(1, 2, 0) \cdot (1, 1, 0)\}}{\{(1, 1, 0) \cdot (1, 1, 0)\}} (1, 1, 0)$$

$$= (1, 2, 0) - \left[\frac{(1+2)}{(1+1)} \right] (1, 1, 0)$$

$$= (1, 2, 0) - \left(\frac{3}{2}, \frac{3}{2}, 0 \right)$$

$$= \left(-\frac{1}{2}, \frac{1}{2}, 0 \right) = w_2 \quad \text{--- (2)}$$

$$w_3 = v_3 - \frac{\langle v_3, w_1 \rangle}{\langle w_1, w_1 \rangle} w_1 - \frac{\langle v_3, w_2 \rangle}{\langle w_2, w_2 \rangle} w_2$$

$$= (0, 1, 2) - \frac{\{(0, 1, 2) \cdot (-\frac{1}{2}, \frac{1}{2}, 0)\}}{\{(-\frac{1}{2}, \frac{1}{2}, 0) \cdot (-\frac{1}{2}, \frac{1}{2}, 0)\}} (-\frac{1}{2}, \frac{1}{2}, 0)$$

$$- \frac{\{(0, 1, 2) \cdot (1, 1, 0)\}}{\{(1, 1, 0) \cdot (1, 1, 0)\}} (1, 1, 0)$$

$$= (0, 1, 2) - \frac{\left\{ \frac{1}{2} \right\}}{\left\{ \frac{1}{2} \right\}} (-\frac{1}{2}, \frac{1}{2}, 0) - \frac{\{1\}}{\{2\}} (1, 1, 0)$$

$$= (0, 1, 2) - \left(-\frac{1}{2}, \frac{1}{2}, 0 \right) - \left(\frac{1}{2}, \frac{1}{2}, 0 \right)$$

4/4

(13)

$$= (0, 1, 2) - \left\{ \left(-\frac{1}{2}, \frac{1}{2}, 0 \right) - \left(\frac{1}{2}, \frac{1}{2}, 0 \right) \right\}$$

$$= (0, 1, 2) - \left(-\frac{1}{4}, 0, 0 \right)$$

$$= \left(\frac{1}{4}, 1, 2 \right) = w_3$$

Step 2 : Calc. $B'' = \{u_1, u_2, u_3\}$

$$\text{So, } u_1 = \frac{w_1}{\|w_1\|}$$

$$= \frac{(1, 1, 0)}{\|(1, 1, 0)\|} = \frac{(1, 1, 0)}{\sqrt{1^2 + 1^2 + 0^2}} = \frac{(1, 1, 0)}{\sqrt{2}}$$

$$= \left(\frac{1}{\sqrt{2}}, \frac{1}{\sqrt{2}}, 0 \right)$$

$$u_2 = \frac{w_2}{\|w_2\|}$$

$$= \frac{\left(-\frac{1}{2}, \frac{1}{2}, 0 \right)}{\sqrt{\left(-\frac{1}{2} \right)^2 + \left(\frac{1}{2} \right)^2 + 0^2}} = \frac{\left(-\frac{1}{2}, \frac{1}{2}, 0 \right)}{\sqrt{\left(\frac{1}{4} + \frac{1}{4} + 0 \right)}} = \frac{\left(-\frac{1}{2}, \frac{1}{2}, 0 \right)}{\sqrt{1/2}}$$

$$= \left(\frac{-1}{\sqrt{2}}, \frac{1}{\sqrt{2}}, 0 \right)$$

$$u_3 = \frac{w_3}{\|w_3\|} = \frac{(1, 1, 2)}{\sqrt{1^2 + 1^2 + 4}} = \frac{(1, 1, 2)}{\sqrt{6}}$$

$$= \frac{\left(\frac{1}{4}, \frac{1}{4}, 2 \right)}{\sqrt{\left(\frac{1}{4} + \frac{1}{4} + 4 \right)}} = \frac{\left(\frac{1}{4}, \frac{1}{4}, 2 \right)}{\sqrt{17/90}} = \left(\frac{17}{360}, \frac{17}{90}, \frac{34}{90} \right)$$

$\frac{1+1+4}{90}$
 $\frac{17}{90}$

we say that B'' is an orthonormal basis for \mathbb{R}^3 .

$u_1, u_2, u_3 \rightarrow$ all are unit vectors \in length 1

w_j → prediction filter coeff.

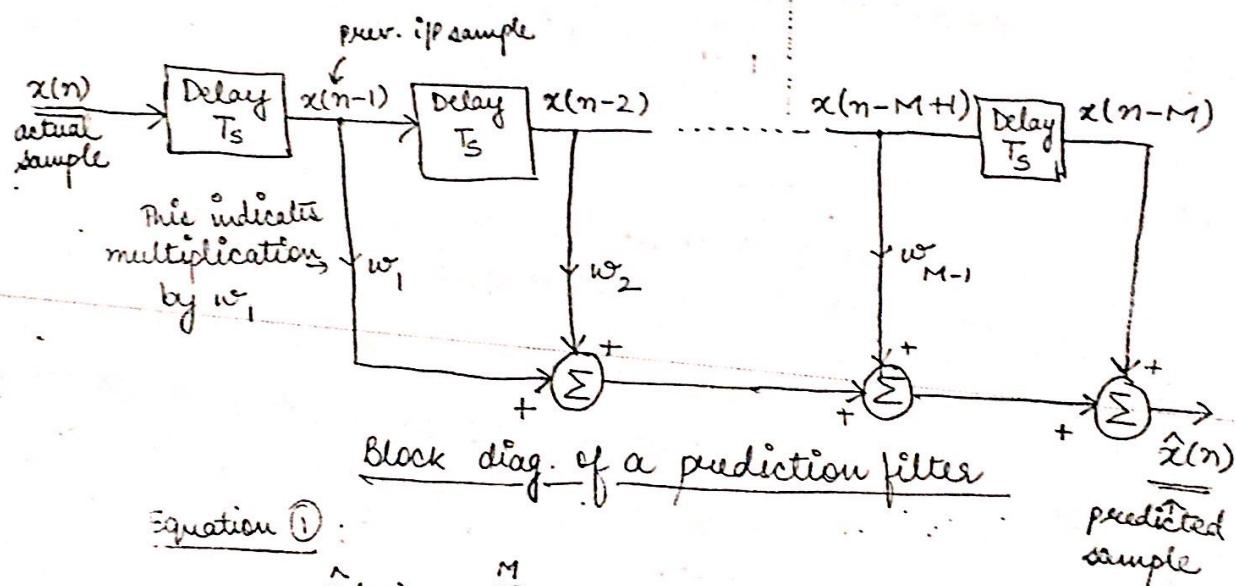
$R(k-j) = \overline{x(n-k) \cdot x(n-j)}$, is autocorrelation of $x(n)$.

and

$$R(k) = \overline{x(n) \cdot x(n-k)}, \quad \text{for } k=1, 2, \dots, M-1$$

⇒ Prediction Filter :-

⇒ implementation of eq. ① is called a prediction filter.



Equation ① :

$$\hat{x}(n) = \sum_{k=1}^M w_k x(n-k)$$

→ eq. ① is called 'filter' ∵ it represents a linear convolution.

* Convolution: it is a mathematical way of combining 2 signals to form a third signal.

Linear convolution is the basic operation to calculate the off for any linear time invariant sys given its impulse resp.

convolution calculates the resp. of an LTI sys.

cross-correlation is used for pattern matching.

Hence, $x(n-1), x(n-2), \dots, x(n-M) \rightarrow$ past i/p's
 $x(n) \rightarrow$ present i/p.

(2.)

- These i/p samples are used for filtering with the help of coeff. w_1, w_2, \dots, w_M . Such filter is of the type of finite impulse response diff digital filters.
- Filtering operation can be modified by changing the value of filter coeff.
- Prediction filter is used earlier in DPCM.
- Prediction error can also be calculated using prediction filter.

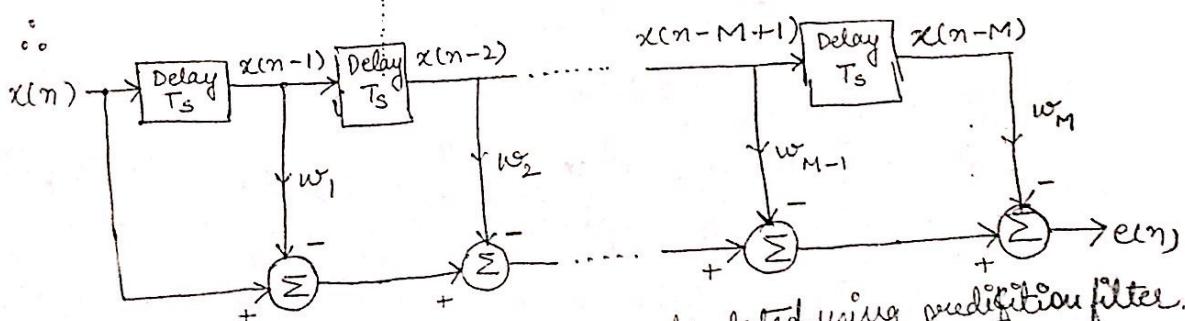
$$e(n) = x(n) - \hat{x}(n)$$

Put ① in above eq'

$$e(n) = x(n) - \sum_{k=1}^M w_k x(n-k)$$

$$= x(n) - [w_1 x(n-1) + w_2 x(n-2) + \dots + w_M x(n-M)]$$

$$= x(n) - w_1 x(n-1) - w_2 x(n-2) - \dots - w_M x(n-M)$$

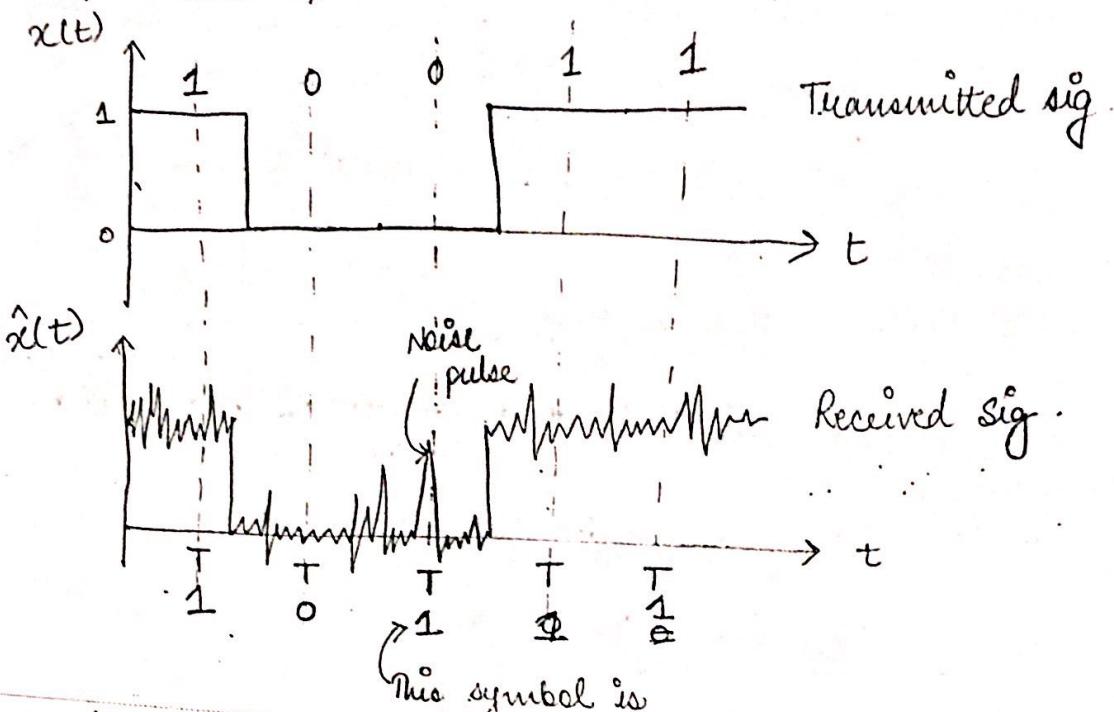


Prediction error calculated using prediction filter.

- in the above diag., $w_1 x(n-1), w_2 x(n-2), \dots, w_M x(n-M)$ are subtracted from $x(n)$. Hence o/p of prediction filter is error $e(n)$.

$\Rightarrow \underline{\text{MATCHED FILTER}}$

- used for detection of signals in baseband and passband tx.



→ The above fig. shows txed sig. digital sig. and received noisy sig. The txed sig sequence is 1 0 0 1 1. The pulse is checked at the point "T" of every bit period. Because of noise pulse present in the third bit at the instant 'T' of checking, it is detected in error.

Requirements of detection error

- SNR of Rx must be improved.
- Signal must be checked at the instant in bit period, when SNR is max.
- The error probability should be min.

Matched filter

- It satisfies all the above requirements.
- It is called matched filter since its impulse response is matched to the shape of ip signal.

and

■ Communication Systems ■

where we have used equations (13.15), (13.13) and (13.9). Recognizing that $|\cos \theta| \leq 1$, the Schwarz inequality of equation (13.16) immediately follows from equation (13.17). Moreover, from the first line of equation (13.17), it may be noted that $|\cos \theta| = 1$ if and only if $s_2 = cs_1$, i.e., $s_2(t) = cs_1(t)$, where c is an arbitrary constant.

The proof of the Schwarz inequality applies to real-valued signals. It may be readily extended to complex-valued signals, in which case equation (13.16) is reformulated as under:

$$\left| \int_{-\infty}^{\infty} s_1(t) s_2^*(t) dt \right|^2 \leq \left(\int_{-\infty}^{\infty} |s_1(t)|^2 dt \right) \left(\int_{-\infty}^{\infty} |s_2(t)|^2 dt \right) \quad \dots(13.18)$$

where the equality holds if and only if $s_2(t) = cs_1(t)$, where c is a constant.

13.6 GRAM-SCHMIDT ORTHOGONALIZATION PROCEDURE

(Expected)

After discussing geometric representation of energy signals, let us now discuss the Gram-Schmidt orthogonalization procedure for which we require a complete orthonormal set of basis functions. To start, let us assume that we have a set of M energy signals represented by $s_1(t), s_2(t), \dots, s_M(t)$.

If, we start with $s_1(t)$, which chosen from this set arbitrarily, the first basis function is defined as

$$\phi_1(t) = \frac{s_1(t)}{\sqrt{E_1}} \quad \dots(13.19)$$

where E_1 = energy of the signal $s_1(t)$

From equation (13.19), we write

$$s_1(t) = \sqrt{E_1} \cdot \phi_1(t) \quad \dots(13.20)$$

or $s_1(t) = s_{11} \phi_1(t)$

where coefficient $s_{11} = \sqrt{E_1}$, and

$\phi_1(t)$ has unit energy, as required.

Next, using the signal $s_2(t)$, we can define the coefficient s_{21} as

$$s_{21} = \int_0^T s_2(t) \phi_1(t) dt \quad \dots(13.21)$$

Therefore, we can introduce a new intermediate function

$$g_2(t) = s_2(t) - s_{21} \phi_1(t) \quad \dots(13.22)$$

which is orthogonal to $\phi_1(t)$ over the interval $0 \leq t \leq T$ by equation (13.21) and by the fact that the basis function $\phi_1(t)$ has unit energy.

Now, we can define the second basis function as

$$\phi_2(t) = \frac{g_2(t)}{\sqrt{\int_0^T g_2^2(t) dt}} \quad \dots(13.23)$$

Now, substituting equation (13.22) into equation (13.23) and simplifying, we get the desired result as under:

$$\phi_2(t) = \frac{s_2(t) - s_{21} \phi_1(t)}{\sqrt{E_2 \cdot s_{21}}} \quad \dots(13.24)$$

where E_2 is the energy of the signal $s_2(t)$.

■ Introduction to Signal Space Analysis ■

From equation (13.23), it is obvious that

$$\int_0^T \phi_2^2(t) dt = 1$$

and, from equation (13.24), it is obvious that

$$\int_0^T \phi_1(t) \phi_2(t) dt = 0$$

This means that $\phi_1(t)$ and $\phi_2(t)$ form an orthonormal pair, as required.

Depending upon the above discussion, we may write the general form as under:

$$g_i(t) = s_i(t) - \sum_{j=1}^{i-1} s_{ij} \phi_j(t) \quad \dots(13.25)$$

where the coefficients s_{ij} are defined as

$$s_{ij} = \int_0^T s_i(t) \phi_j(t) dt, \quad j = 1, 2, \dots, i-1 \quad \dots(13.26)$$

It may be noted that equation (13.22) is a special case of equation (13.25) with $i = 2$. Further for $i = 1$, the function $g_i(t)$ reduces to $s_i(t)$.

Given the $g_i(t)$, now, we can define the set of basis functions as under:

$$\phi_i(t) = \frac{g_i(t)}{\sqrt{\int_0^T g_i^2(t) dt}}, \quad i = 1, 2, \dots, N \quad \dots(13.27)$$

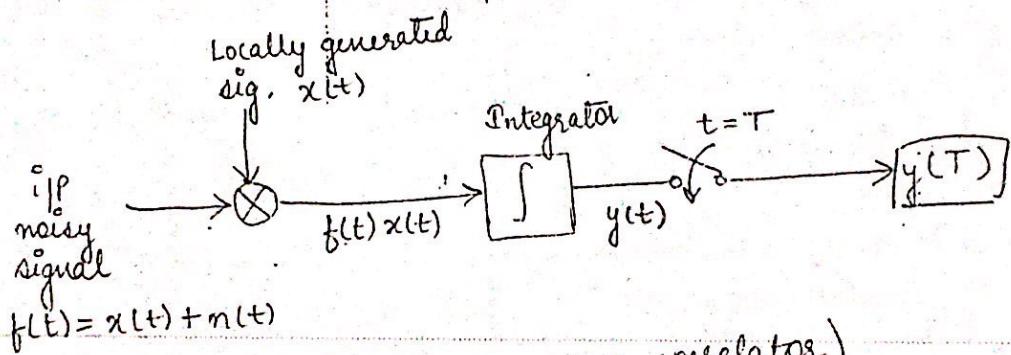
which forms an orthonormal set.

Further, the dimension N is less than or equal to the number of given signals, M depending on one of two possibilities:

- (i) The signals $s_1(t), s_2(t), \dots, s_M(t)$ form a linearly independent set, in which case $N = M$.
- (ii) The signals $s_1(t), s_2(t), \dots, s_M(t)$ are not linearly independent, in which case, $N < M$, and the intermediate function $g_i(t)$ is zero for $i > N$.

CORRELATOR RECEIVER

- Correlation is a pattern matching process where we find out similarities b/w 2 signals.
- In practice, a linear matched filter is more often implemented as a correlator. This allows a cheaper implementation of filters.



(Block diag. of a correlator)

Steps:-

i/p noisy sig. is multiplied to a replica of i/p sig $x(t)$

↓

$f(t)$

Result $(f(t)x(t))$ is fed to an integrator

↓

{ working of correlator

↓

o/p of integrator is sampled at $t = T$

→ Why it is called a correlator?

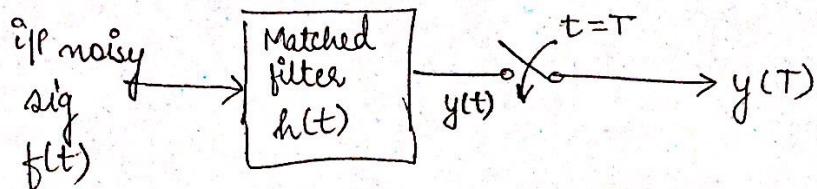
→ ∵ it correlates received sig. $f(t)$ & i/p signal $x(t)$.

$$\text{So, o/p } y(t) = \int_0^T f(t)x(t) dt \quad \text{--- (1)}$$

Now, at $t = T$

O/P of correlator is :

$$y(T) = \int_0^T f(t) x(t) dt \quad \text{--- (2)}$$



Now, consider the matched filter ⁱⁿ acc to the above diagram.

→ The O/P $y(t)$ is obtained by convolution of i/p. $f(t)$ & its impulse response $h(t)$.

* here, we don't need a locally generated replica of the i/p signal $x(t)$.

$$\begin{aligned} y(t) &= f(t) \otimes h(t) \\ &= \int_{-\infty}^{\infty} f(z) \cdot h(t-z) dz \quad \text{--- (3)} \end{aligned}$$

Now, we know that :

Impulse resp. of matched filter, $h(t)$ is :-

$$h(t) = \frac{2K}{N_0} x(T-t) \quad \text{--- (4)}$$

$$\therefore h(t-z) = \frac{2K}{N_0} x(T-t+z)$$

Now, integration is performed over one bit period.

∴ limits change to 0 to T

(7)

So,

$$y(t) = \frac{2k}{N_0} \int_0^T f(z) x(T-t+z) dz$$

at $t=T$

$$\begin{aligned} y(T) &= \frac{2k}{N_0} \int_0^T f(z) x(T-T+z) dz \\ &= \frac{2k}{N_0} \int_0^T f(z) x(z) dz \end{aligned}$$

let $\tau = t$ (only for convenience)

$$\therefore y(T) = \frac{2k}{N_0} \int_0^T f(\tau) x(\tau) d\tau \quad \text{--- (5)}$$

from eq⁴ (2) and (5)eq⁴ (5) gives the off of matched filtereq⁴ (2) " " " " " correlator.we may say that both are identical
as eq⁴ (2) = (5).The constt. in eq⁴ (5) $(\frac{2k}{N_0})$ can be normalized
to 1.∴ we may say that both the matched
filter & correlator provides the same off.



ORTHOGONAL SIGNALS

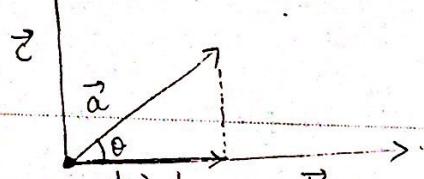
Orthogonality: It is the property that allows transmission of ~~more than one signal over a common channel & successful detection~~.

Orthogonal signals: 2 sig's are orthog. if they are mutually independent.

How to find out whether 2 signals are orthog?

→ Orthog. vectors:

\vec{a} = (light source)



$|\vec{a}_b|$ (projection of \vec{a} on \vec{b})

shadow of \vec{a} on \vec{b} .

\Rightarrow mag. of proj.

$$|\vec{a}_b| = |\vec{a}| \cos \theta$$

$$|\vec{a}_b| \cdot |\vec{b}| \neq 0$$

(mag. of proj. \propto mag. of vector \vec{b}) \Rightarrow their \times $\neq 0$

$$= |\vec{a}| \cos \theta \cdot |\vec{b}| \neq 0$$

$$|\vec{a}| |\vec{b}| \cos \theta \neq 0$$

↑ dot prod. of $\vec{a} \cdot \vec{b}$

$$\Rightarrow \vec{a} \cdot \vec{b} \neq 0$$

$\therefore \vec{a} \& \vec{b}$ are not independent
we say that these vectors are not orthog.
to each other.

(8)

Now, if we have another vector (let it be \vec{c}) and we put it under a light source. we see that, proj of \vec{c} on \vec{B} is null. i.e. \vec{c} does not have any shadow over \vec{B} .

Now, $\theta = 90^\circ$. (b/w \vec{b} & \vec{c})

$$\cos 90^\circ = 0$$

$$\therefore \vec{b} \cdot \vec{c} = 0 \quad (\because \cos 90^\circ = 0)$$

∴ we say that \vec{b} & \vec{c} are orthog. (mutually indep.)

Signal space \rightarrow inner prod. (Or dot prod.)
(Definite integral)

Let there are 2 sig's $x_1(t)$ & $x_2(t)$

\Rightarrow The 2 sig's are orthog. if

$$\boxed{\int_{-\infty}^{\infty} x_1(t) x_2(t) dt = 0} \quad [\text{for non-periodic sig's}]$$

and

$$\boxed{\int_0^T x_1(t) x_2(t) dt = 0} \quad [\text{for periodic sig's}]$$

over one time period

\Rightarrow Properties of orthog. sig's (4)

(i) 2 harmonics of different frequencies are always \rightarrow orthog.

$$\text{Let } x_1(t) = \sin(n\omega_0 t + \phi_1) \quad [n \neq m]$$

$$x_2(t) = \sin(m\omega_0 t + \phi_2) \quad [\phi_1 \neq \phi_2]$$

∴ both phase & frequencies are different

$$\Rightarrow \text{Hence, } x_1(t) x_2(t) \rightarrow 0$$

$$\int_0^T \sin(n\omega_0 t + \phi_1) \sin(n\omega_0 t + \phi_2) dt = 0$$

(ii) sine & cosine fun" \bar{c} same freq. & phase $\rightarrow 0$.
(answ)

$$\int_0^T \sin(n\omega_0 t + \phi) \cos(n\omega_0 t + \phi) dt = 0$$

↑ ↑
f. & φ are same

(iii) dc value & sine fun" $\rightarrow 0$

$$\int_0^T a \sin(n\omega_0 t + \phi) dt = 0$$

↑
dc value

Imp (iv) Effects of orthogonality on total energy (E) & avg. power (P) calculations.

if we have

$$x_1(t) \perp x_2(t) \rightarrow 0$$

$$\Rightarrow \int_{-\infty}^{\infty} x_1(t) x_2(t) dt = 0$$

& if we have another sig. $y(t) \subseteq$ is :-

$$y(t) = x_1(t) + x_2(t)$$

\rightarrow Then, the av. power $P_y = P_{x_1} + P_{x_2}$ (we can use this directly if $x_1(t)$ & $x_2(t)$ are 0)

$$\Rightarrow \text{Total energy, } E_y = E_{x_1} + E_{x_2} \quad (n)$$

① Total 'E' is ∞ in case of power sig's

we use eq "①" if $x_1(t) + x_2(t) \rightarrow$ power sig's.

② Avg. power = 0(zero) in case of energy sig's

we use eq "②" if $x_1(t) + x_2(t) \rightarrow$ E.Sig's.

Example :-

Example :-

$$\textcircled{1} \quad y(t) = \overbrace{2 \sin(3\omega_0 t + 45^\circ)}^{x_1(t)} + \overbrace{4 \sin(4\omega_0 t + 35^\circ)}^{x_2(t)}$$

Cal. avg. power & total energy.

① Solut^o:

(i) Find whether $x_1(t)$ & $x_2(t)$ are orthog. or not.

From the above eq"

$$f_1 \neq f_2 \quad (\text{i.e } 3 \neq 4)$$

$\Rightarrow \phi_1 \neq \phi_2$ (i.e. $45^\circ \neq 35^\circ$)

$\therefore x_1(t) + x_2(t)$ are $\rightarrow 0$. [from prop. (i)]

(ii) Use property (iv) to calc. Avg. P. & total energy (E)

$$P_y = P_{x_1} + P_{x_2} \quad E_y = \infty \quad (\because x_1(t) \text{ & } x_2(t) \text{ are periodic.})$$

$$= \frac{a^2}{2} + \frac{b^2}{2}$$

$$= \frac{9}{2} + \frac{16}{2}$$

$$= 2 + 8$$

$$P_y = 10 \text{ W}$$

$$\Delta \boxed{E_y = 0}$$

$$\therefore f = \frac{A_0^2}{2}$$

↓
periodic sig's are power
sig's

2- in case of power sigs
energy, $E = \infty$)

* How to find whether a sig is E.S. or Power-Sig?

$$\text{Eq: } \begin{array}{l} 1. x(t) = t u(t) \\ \quad \quad \quad \uparrow \quad \uparrow \\ \quad \quad x_1(t) \quad x_2(t) \end{array}$$

$t \rightarrow$ ramp fun"

$$x_1(t) = t = \begin{cases} 0 & t < 0 \\ t & t \geq 0 \end{cases}$$

↓
Simply calc. the avg.
power (P) & tot. Energy (E)
of that dig.

If $E = \text{finite}$ } Energy Sig
 $P = 0$

if $P = \text{finite}$ } Power sig.

$x_2(t) \rightarrow$ unit step func'

$$x_2(t) = u(t) = \begin{cases} 0 & t < 0 \\ 1 & t \geq 0 \end{cases}$$

$$\text{av. power, } P = \lim_{T \rightarrow \infty} \frac{1}{2T} \int_{-T}^T |x(t)|^2 dt \quad \begin{matrix} \text{std. formula} \\ \text{calc. avg. P} \end{matrix}$$

$$= \lim_{T \rightarrow \infty} \frac{1}{2T} \left[\int_{-T}^0 0 dt + \int_0^T |t|_1^2 dt \right]$$

$$= \lim_{T \rightarrow \infty} \frac{1}{2T} \left[0 + \left[\frac{t^3}{3} \right]_0^T \right]$$

$$= \lim_{T \rightarrow \infty} \frac{1}{2T} \left[\frac{T^3}{3} \right]$$

$$= \lim_{T \rightarrow \infty} \frac{T^2}{6} = \infty.$$

So, av. Power = ∞ for $x(t)$

$$\text{Tot. energy, } E = \int_{-\infty}^{\infty} |x(t)|^2 dt$$

$$= \int_{-\infty}^0 0 dt + \int_0^{\infty} t^2 dt$$

$$= \left[\frac{t^3}{3} \right]_0^{\infty}$$

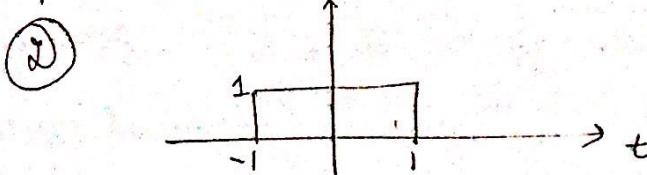
$$= \infty - 0 = \infty$$

$$\text{tot. } E = \infty$$

So, sig. $x(t)$ is neither energy sig nor p.s.

(10)

$$② y(t) = 2 \text{ rect}(t/2)$$

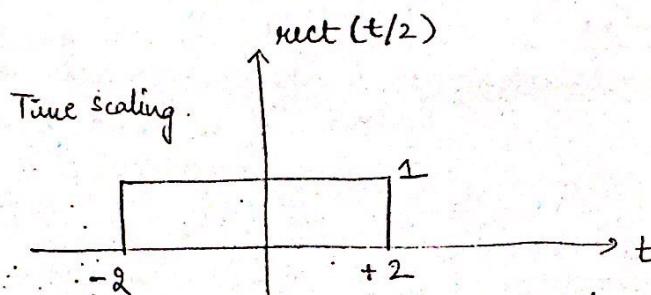


This is a rect. sig.

first, perform time scaling of rect. sig.

2nd, " amplitude" of '2' amplitude

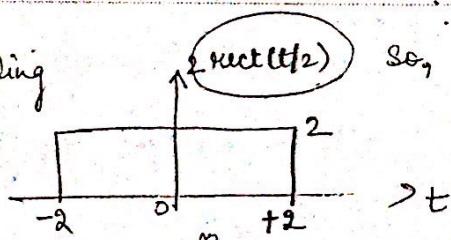
So,



(divide 1/0.5 on bds
here we have t/2)

so, time is divided by 0.5

Amp. Scaling



$$\text{(i) Cal. tot. E.} = \int_{-\infty}^{\infty} |y(t)|^2 dt$$

$$= \int_{-2}^{2} (2)^2 dt = 4 [t]_{-2}^{2}$$

2+2

$$= 16 \text{ Joules (finite)}$$

$$\text{(ii) Cal. avg. P} = \lim_{T \rightarrow \infty} \frac{1}{2T} \int_{-T}^{T} 4 dt$$

$$= \frac{1}{2T} [t]_{-T}^T = 4 \cdot \frac{T}{2T} = 2 \text{ (finite no.)}$$

$$= \frac{\infty}{\infty} = 0$$

$$\therefore \text{sig. } y(t) = \underline{\underline{E.S}}$$

⇒

GRAM-SCHMIDT ORTHOGONALIZATION PROCEDURE :-

Goal: To show ① a set of vectors is orthogonal.

↓
forms orthonormal basis

↓ +
rep. a vector relative to an
orthonormal basis.

② → apply Gram-Schmidt O[†] procedure.

Definitions:

1. Orthogonal set : A set 'S' of vectors in an inner prod. space V is called orthog. when every pair of vectors in S is orthog.
i.e for every v_i, v_j in S $\Rightarrow v_i \cdot v_j = 0$ [cond. for orthog.] $i \neq j$.

2. Orthonormal : If each vector in the set is a unit vector then, S is called orthonormal.

Eg: ① Show that set 'S' is an orthonormal basis for R^3 .

$$S = \{v_1, v_2, v_3\} = \left\{ \underbrace{\left(\frac{1}{\sqrt{2}}, \frac{1}{\sqrt{2}}, 0 \right)}_{\text{these vectors lie on diff. planes.}}, \left(-\frac{\sqrt{2}}{6}, \frac{\sqrt{2}}{6}, \frac{2\sqrt{2}}{3} \right), \left(\frac{2}{3}, -\frac{2}{3}, \frac{1}{3} \right) \right\}$$

① Solution :

(i) Verify that all vectors are pair-wise orthog.
 $\therefore \parallel \parallel$ of each vect. = 1.

$$v_1 \cdot v_2 = 0 \quad \text{and} \quad \|v_1\| = 1 = \sqrt{\left(\frac{1}{\sqrt{2}}\right)^2 + \left(\frac{1}{\sqrt{2}}\right)^2 + (0)^2} = \sqrt{\frac{1}{2} + \frac{1}{2}} = 1$$

$$v_1 \cdot v_3 = 0 \quad \|v_2\| = 1 = \sqrt{\left(-\frac{\sqrt{2}}{6}\right)^2 + \left(\frac{\sqrt{2}}{6}\right)^2 + \left(\frac{2\sqrt{2}}{3}\right)^2} = \sqrt{\frac{2}{36} + \frac{2}{36} + \frac{8}{9}} = \sqrt{\frac{10}{36}} = \sqrt{\frac{5}{18}}$$

$$v_2 \cdot v_3 = 0 \quad \|v_3\| = 1 = \sqrt{\left(\frac{2}{3}\right)^2 + \left(-\frac{2}{3}\right)^2 + \left(\frac{1}{3}\right)^2} = \sqrt{\frac{4}{9} + \frac{4}{9} + \frac{1}{9}} = \sqrt{\frac{9}{9}} = 1$$

$$\therefore \parallel \parallel = \sqrt{\left(\frac{2}{3}\right)^2 + \left(-\frac{2}{3}\right)^2 + \left(\frac{1}{3}\right)^2}$$

$$= \sqrt{\frac{4}{9} + \frac{4}{9} + \frac{1}{9}} = \sqrt{\frac{9}{9}} = 1$$

∴ we conclude that set 'S' is an orthonormal set and it is also a basis for R^3 .

MAXIMUM LIKELIHOOD DETECTOR

(1)

Concept of "Detection of signals in noise"

A source tx's M diff. signals i.e. $s_1(t), s_2(t), \dots, s_M(t)$ etc.

All these are equally likely

each one has a probability of

$$\left(\frac{1}{M}\right)$$

Now, let us consider a time slot of 'T' sec.

if one of these 'M' possible signals
be txed in one (T sec) duration

Then, for an AWGN channel

$$x(t) = s_i(t) + w(t) \quad 0 \leq t \leq T$$

where $i = 1, 2, 3, \dots, M$

$x(t) \rightarrow$ Rxed sig.

$s_i(t) \rightarrow$ i^{th} msg sig.

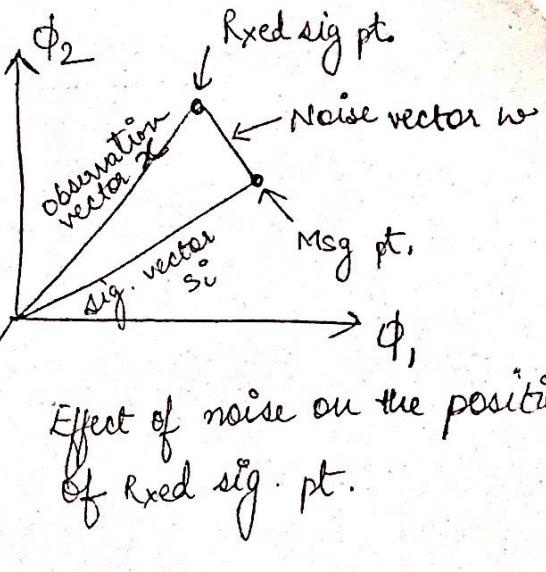
$w(t) \rightarrow$ sample fun. of white noise process

$w(t) \Rightarrow$ zero mean value

$$psd = \frac{N_0}{2}$$

* Note: After Rxing the sig. $x(t)$, the Rx has to make an estimate of the txed signal $s_i(t)$ to decide which symbol was txed.

* Here, $\phi_1, \phi_2, \phi_3 \rightarrow$ are
orthonormal basis
functions where $N=3$



Effect of noise on the position
of Rxed sig. pt.

In short:

→ Signals the above fig. illustrates relationship b/w the observation vector x , sig. vector s_i and noise vector w for $N=3$.

MLD

→ Here, we are dealing w/ the detection problem.

→ if observat' vector x is given then :-
we have to perform a mapping from x to get
an estimate \hat{m}_i of the txed sig symbol m_i ,
in a way the avg. P_e is minimum

(av. prob. of symbol error)

This is called Max. Likelihood Detection.

Now, $m_i \rightarrow$ Original txed sig.

$\hat{m}_i \rightarrow$ estimated sig.

Assume that $x =$ Observation vector

then,

Rx will make decision of $\hat{m}_i = m_i$

Now, aq. prob. of symbol error in such a decision is: ⁽²⁾

$$P_e(m_i^c, x) = P(m_i^c \text{ not sent}/x) = P(m_i^c \text{ sent}/x), \quad (1)$$

$P(m_i^c \text{ sent}/x) \Rightarrow$ conditional prob. that
 m_i^c was sent provided x is fixed.

Now, Optimum Decision Rule

Aim \Rightarrow minimize the prob. of error in mapping
 x into a decision.

Hence, from eq["] (1), we have to deduce an
optimum ^{decision} rule.

The estimate $\hat{m}_i = m_i^c$ if

$$P(m_i^c \text{ sent}/x) \geq P(m_k^c \text{ sent}/x) \quad \text{for all } k \neq i \\ \text{where } k=1, 2, \dots, M. \quad (2)$$

This rule is called as the
maximum a posterior prob.

Now, applying Bayes rule to eq["] (2).
we restate the decision rule as:-

The estimate $\hat{m}_i = m_i^c$ if

$$\left\{ \frac{p_k f_x(x/m_k)}{f_x(x)} \right\} \text{ is max. for } k=i \quad (3)$$

$p_k \rightarrow$ prior prob. of occurrence of symbol m_k
 $f_x(x/m_k) \rightarrow$ likelihood fun["] of corresponding to x
of symbol m_k .

$f_{X|Y}(x|y) \rightarrow$ unconditional joint pdf of rand. variable X .

Observations from eq^③

- (i) denominator term $f_X(x)$ is indep. of txed sig.
- (ii) Then, a prior prob. $p_k = p_i$
 \therefore all txed sigs are equally likely.

So, simplifying the decision rule:-

The estimate $\hat{m}_i = m_i$, if

$f_{X|Y}(x|m_k)$ is maximum for $k=i$

Generally, using ^{natural} logarithm of likelihood fun⁴ is easier than the likeli. fun's itself.

\downarrow
natural logarithm of likeli. fun⁴ = metric

$f_{X|Y}(x|m_k) \geq 0$ (always non-negative)

Now, if $X > Y > 0$, then $\log_e X > \log_e Y$

\therefore decision rule can be restated as:-

The estimate $\hat{m}_i = m_i$ if

$\log_e [f_{X|Y}(x|m_k)]$ is max. for $k=i$

Eq^④ is called the max. likeli. rule &
device used for its implementat⁵ is called max. likelihood Detector.

(3)

Conclusion :-

From eq. (5) :

The MLD computes the metric of each txed sig. m_k
 $k=1, 2, \dots, m$

compares it w/ these values

then, makes a decision based on the max. of them.

Scrambling :- Pg - 788

INTERSYMBOL INTERFERENCE (ISI)

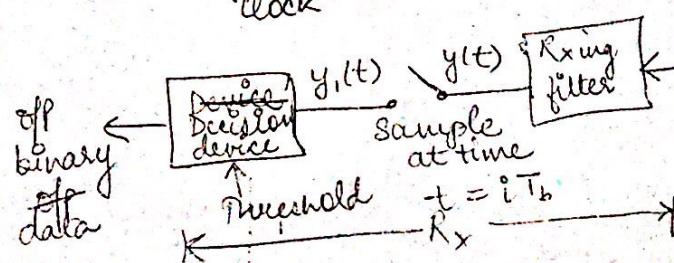
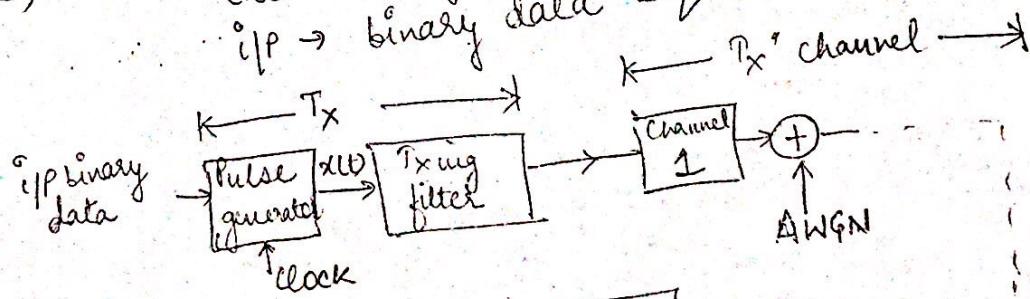
→ ISI defn'

Data is txed in form of bits or pulses, the off produced at the Rx due to other bits or symbols interferes w/ the off produced by the desired bit. This is ISI.

→ ISI introduces errors in the detected sig.

→ Elements of impulse binary PAM sys.

i/p → binary data seq, b_k in bit durat' T_b sec



i/p \rightarrow binary data seq $\{b_k\}$

The seq. is applied to a pulse generator to produce a discrete PAM signal. It is given by :-

$$x(t) = \sum_{k=-\infty}^{\infty} a_k v(t - kT_b) \quad (1)$$

O/P
of pulse
generator

$v(t) \rightarrow$ basic pulse, (normalized such that $v(0) = 1$)

\rightarrow The 1st block of the sys. i.e PAM \Rightarrow converts i/p seq $\{b_k\}$ into polar form i.e

$$\text{if } b_k = 1 \Rightarrow a_k = 1$$

$$\text{if } b_k = 0 \Rightarrow a_k = -1$$

↓

This PAM sig. $x(t)$ is then fed to tx'ing filter.

↓

its O/P is fed to the channel.
(impulse response of this channel be $h(t)$)

A random noise is then added to the txed while travelling over the tx' channel. So, the rx'd sig. is contaminated by noise.

↓
channel O/P is passed through a Rx'ing filter

↓
This filter O/P is sampled synchronously at the tx

Seq. of samples obtained at off of Rx^{ing} is used (4)
to reconstruct the original data seq. w/ the help
of a decision making device

Each sample is compared to a threshold level in
the decision making device.
if amplitude is > threshold level \Rightarrow symbol
is 1 is Rxed.

if amp. is \leq threshold level \Rightarrow it is decided
that symbol 0 is Rxed.

$$\text{Rx}^{\text{ing}} \text{ filter off } \Rightarrow y(t) = u \sum_{k=-\infty}^{\infty} a_k p(t - kT_b) + n(t)$$

$u \rightarrow$ scaling factor

$n(t) \rightarrow$ noise

$p(t - kT_b) \rightarrow$ combined impulse response of the Rx^{ing} filter

at time $t = t_i = iT_b$ ($i \rightarrow$ any integer)

$$\text{so, } y(t_i) = u \sum_{k=-\infty}^{\infty} a_k p(iT_b - kT_b) + n(t_i) \quad (2)$$

$$\text{or } y(t_i) = u a_i + u \sum_{\substack{k=-\infty \\ k \neq i}}^{\infty} a_k p(iT_b - kT_b) + n(t_i) \quad (3)$$

Eq (3) shows the Rx off $y(t)$ at instant $t = t_i$

Now, from eq (3) :-

- (i) The first term $u a_i$ is produced by i^{th} txed bit.
Only this term should be present.

(ii) 2nd term represents the residual effect of all
txed bits, obtained at the time of sampling
ith bit. This residual effect = ISI.

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- Cause of ISI
- Effect of ISI
- Remedy to Reduce ISI

⇒ EYE DIAGRAM

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