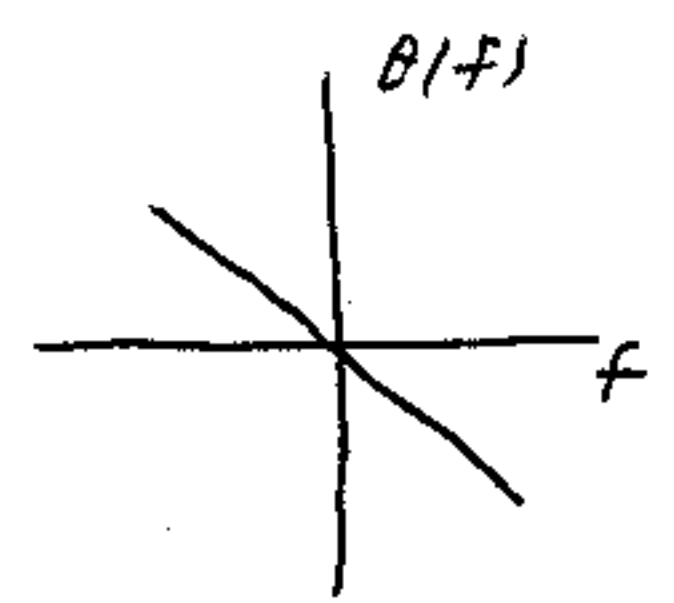
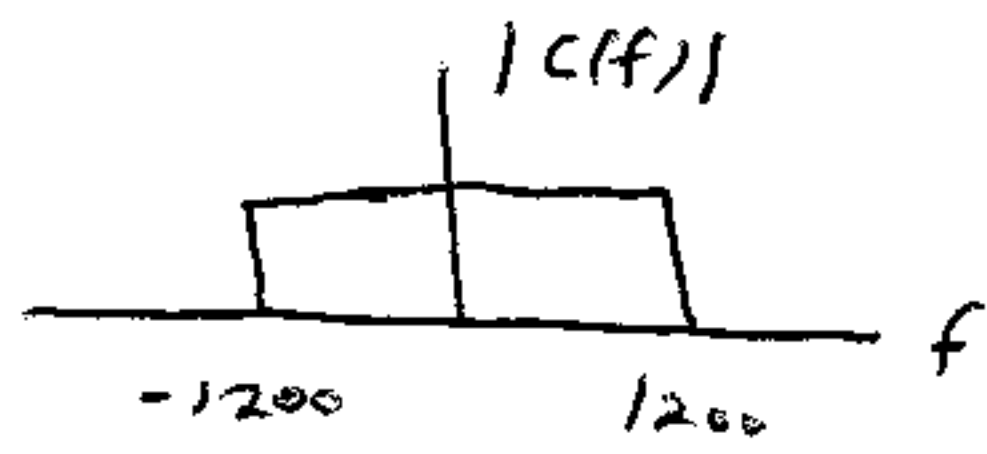


SET $X(f) = X_{RC}(f) \Rightarrow X_{RC}(f) = |G_T(f)|^2$

$\therefore G_T(f) = \sqrt{|X_{RC}(f)|} e^{-j2\pi f t_0}$

WHERE t_0 CHOSEN TO MAKE $G_T(f)$ REALIZABLE (APPROXIMATELY)

EXAMPLE :



$C(f) = |C(f)| e^{j\theta(f)}$

$\theta(f) = -2\pi f t_0$

MAXIMUM SIGNALING RATE

IS $1/T = 2W = 2400$ SYMBOLS/SEC

A MORE PRACTICAL RATE IS $\frac{1}{2T} + \frac{\alpha}{2T} = W$

WITH $\alpha = \frac{1}{2} \Rightarrow \frac{1}{T} = \frac{2W}{1+\alpha} = \frac{2W}{3/2} = \frac{4}{3}W = 1600$

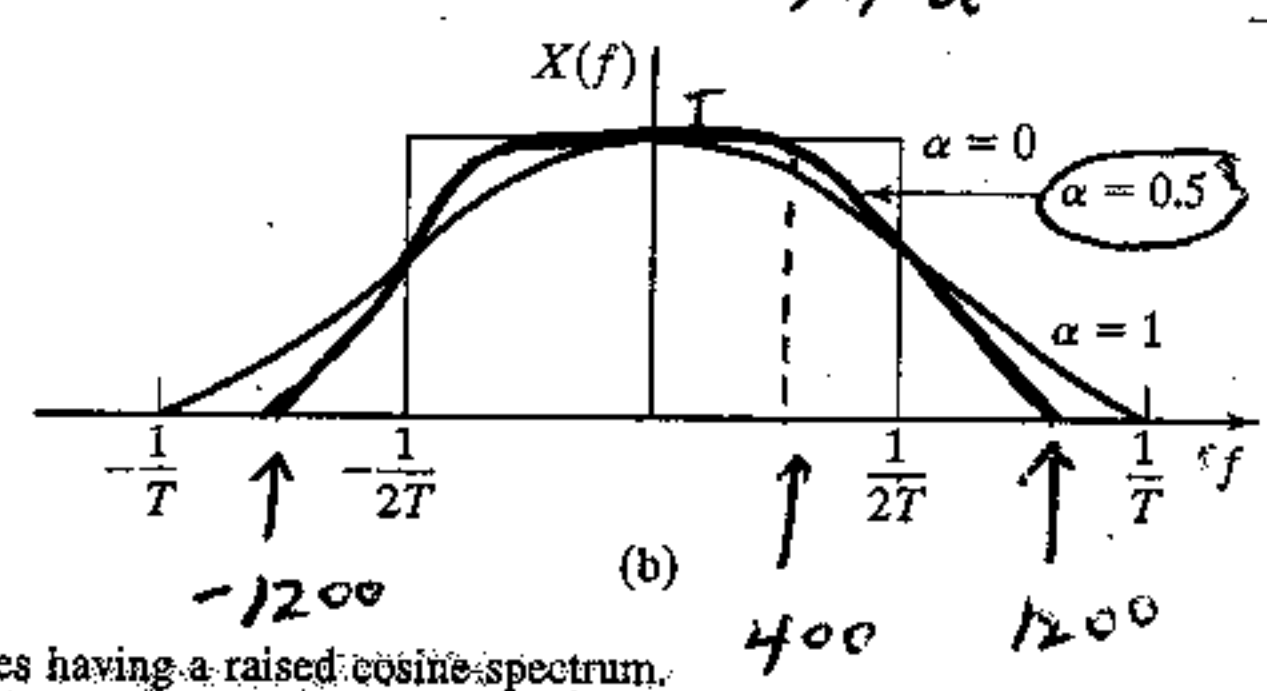


Figure 9.9 Pulses having a raised cosine spectrum.

$$X_{RC}(f) = \begin{cases} T, & 0 \leq |f| \leq 400 \\ \frac{T}{2} [1 + \cos(\frac{\pi}{800} (|f| - 400))], & 400 \leq |f| \leq 1200 \\ 0, & |f| \geq 1200 \end{cases}$$

SYSTEM DESIGN - CHANNEL DISTORTION

FOR NO ISI $G_T(f)C(f)G_R(f) = X_{pc}(f)$,
 BUT NOW ASSUME CHANNEL IS NOT IDEAL.

$$C(f) = |C(f)| e^{j\theta_c(f)}$$

\uparrow NOT A CONSTANT WITH FREQ. \uparrow NOT LINEAR ($\neq -2\pi ft_0$)

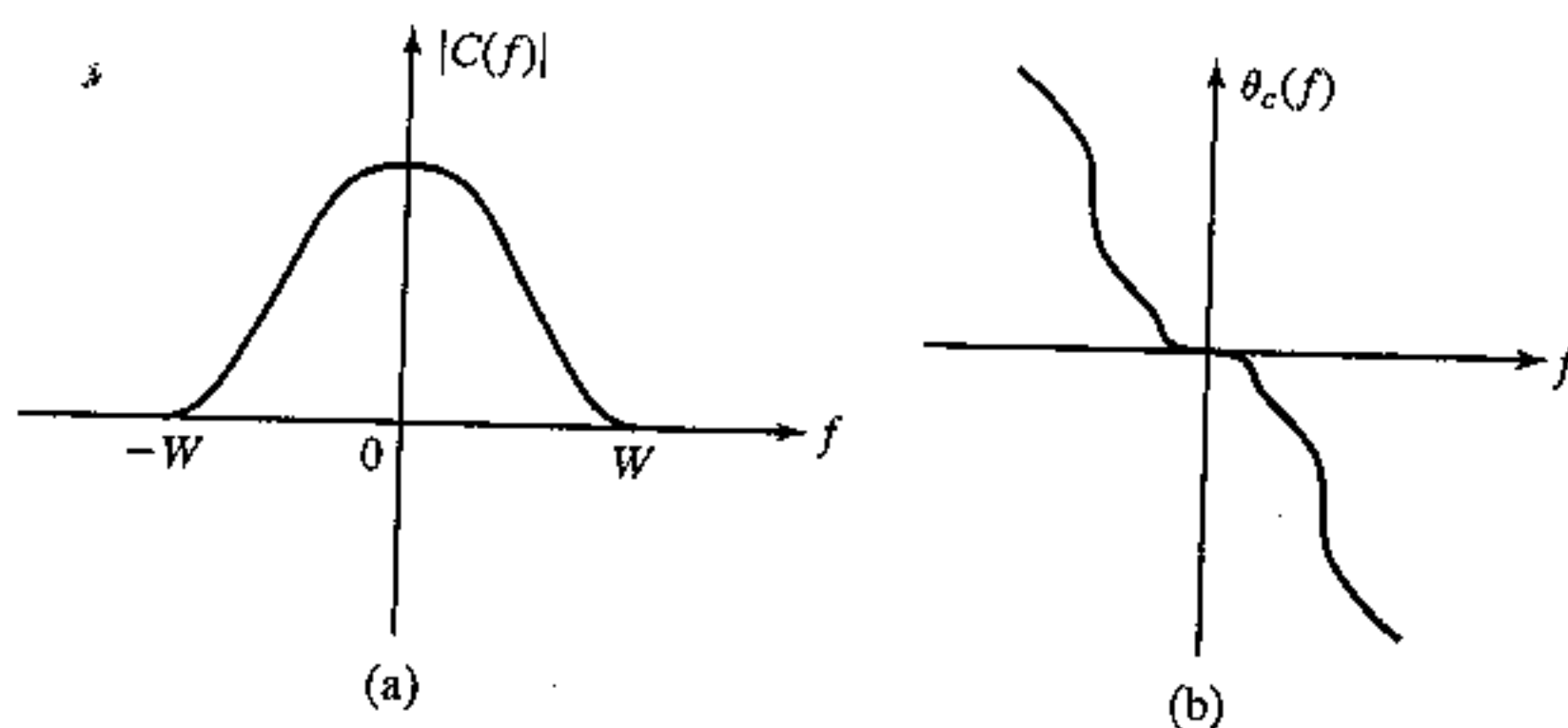


Figure 9.15 Channel characteristics illustrating (a) amplitude distortion and (b) phase distortion.

AMPLITUDE DISTORTION - HIGHER FREQUENCY COMPONENTS, AS AN EXAMPLE, ARE ATTENUATED
 \Rightarrow SIGNAL IS SMOOTHED



PHASE DISTORTION - CAN BE VIEWED MORE EASILY BY CONSIDERING TIME DELAY VS FREQUENCY OF CHANNEL

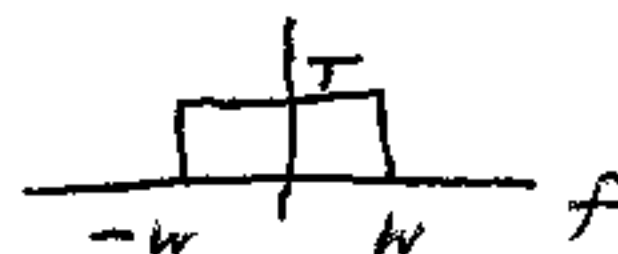
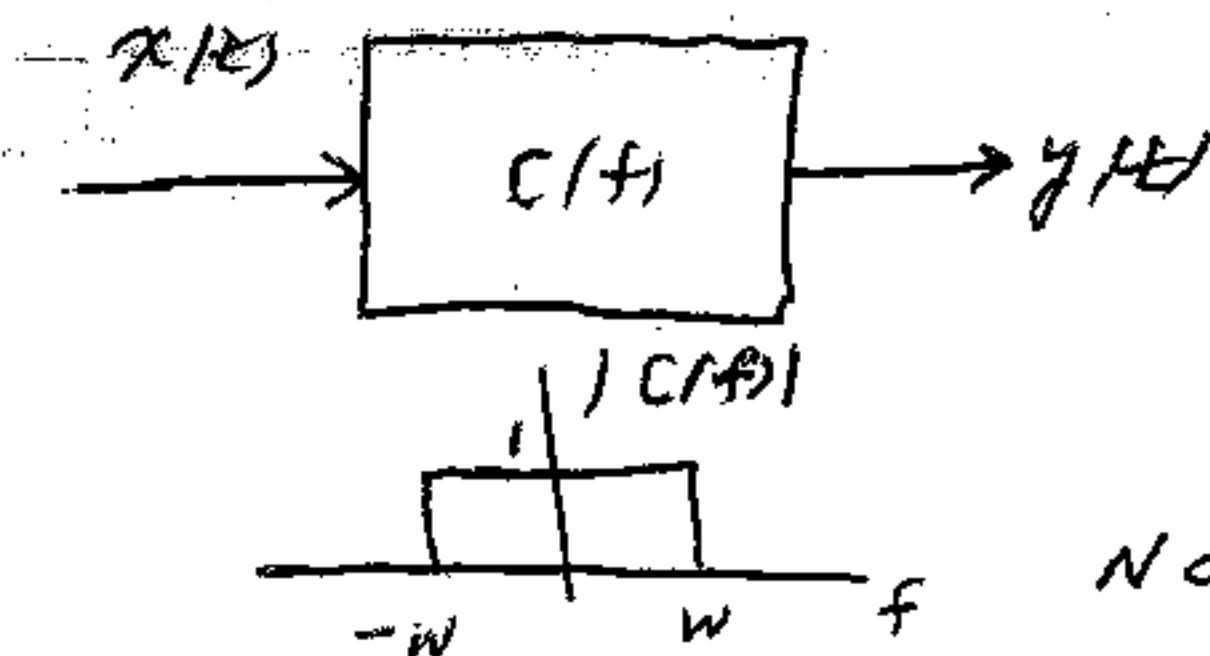
$$\tau(f) = -\frac{1}{2\pi} \frac{d\theta_c(f)}{df}$$

$$= -\frac{1}{2\pi} \frac{d(-2\pi f t_0)}{df} = t_0 \quad |f| \leq W$$

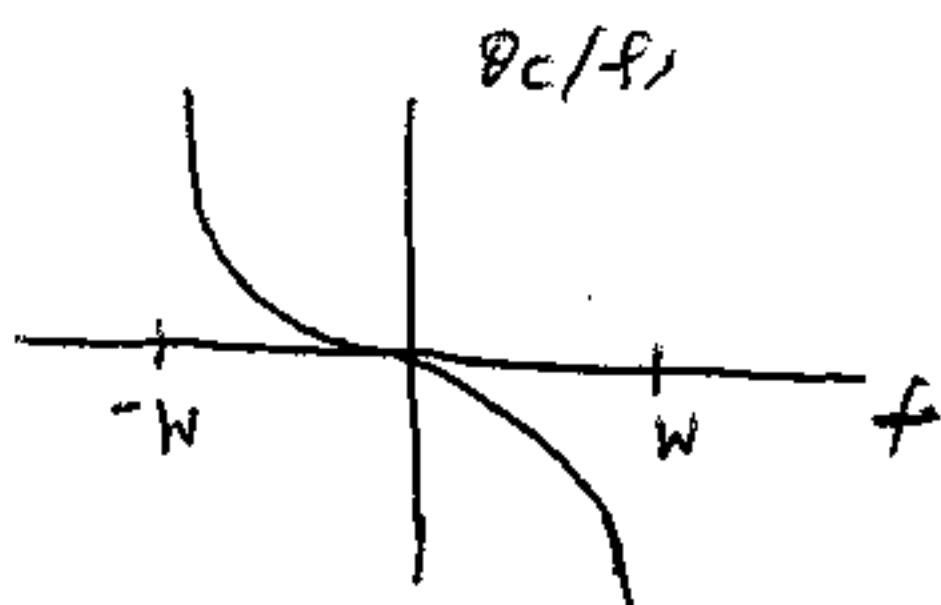
FOR IDEAL CHANNEL,

FOR NONIDEAL CHANNEL, $\theta_c(f)$ IS NONLINEAR
 $\Rightarrow \tau(f)$ DEPENDS ON $f \Rightarrow$ FREQ. COMPONENTS
 "ARRIVE" AT DIFFERENT TIMES. ALSO,
 CALLED DELAY DISTORTION

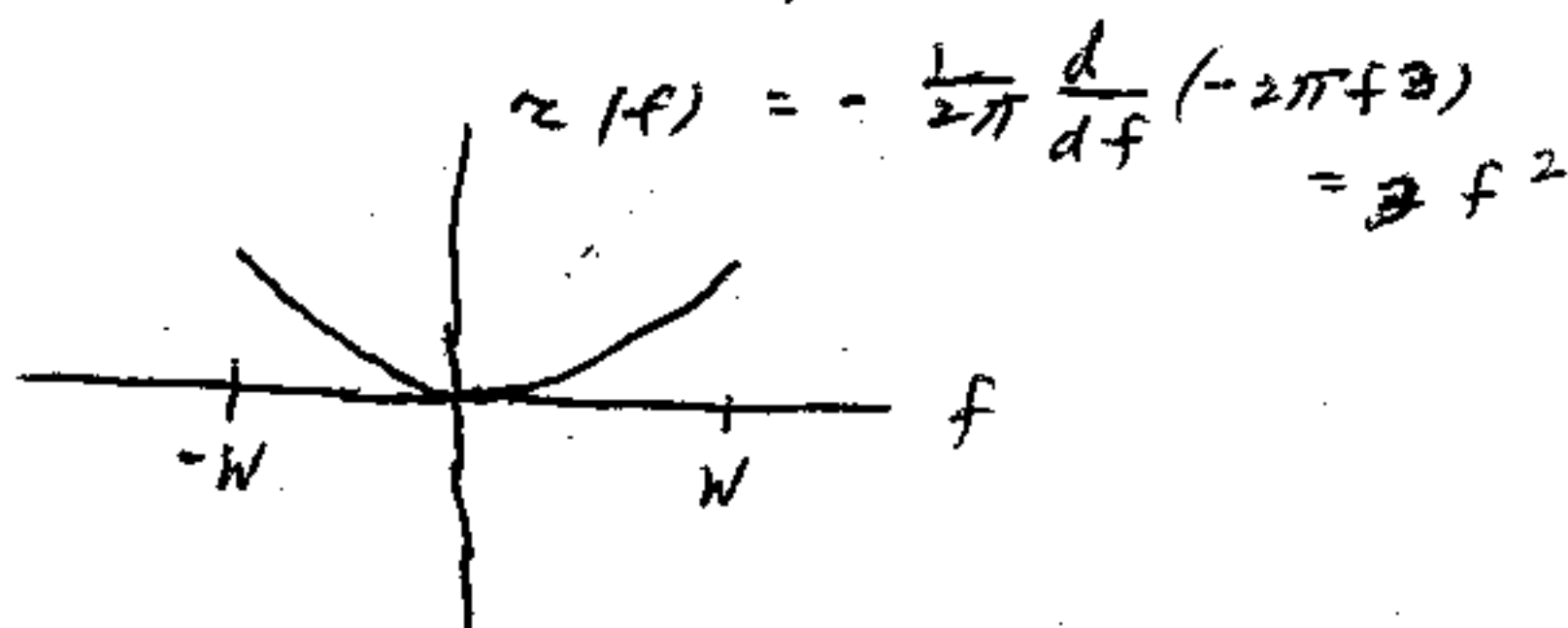
SINC
 CONSIDER A PULSE INPUT, $x(t) = \frac{\sin 2\pi W t}{2\pi W t}$



NO AMPLITUDE
 DISTORTION



PHASE DISTORTION



DELAY DISTORTION

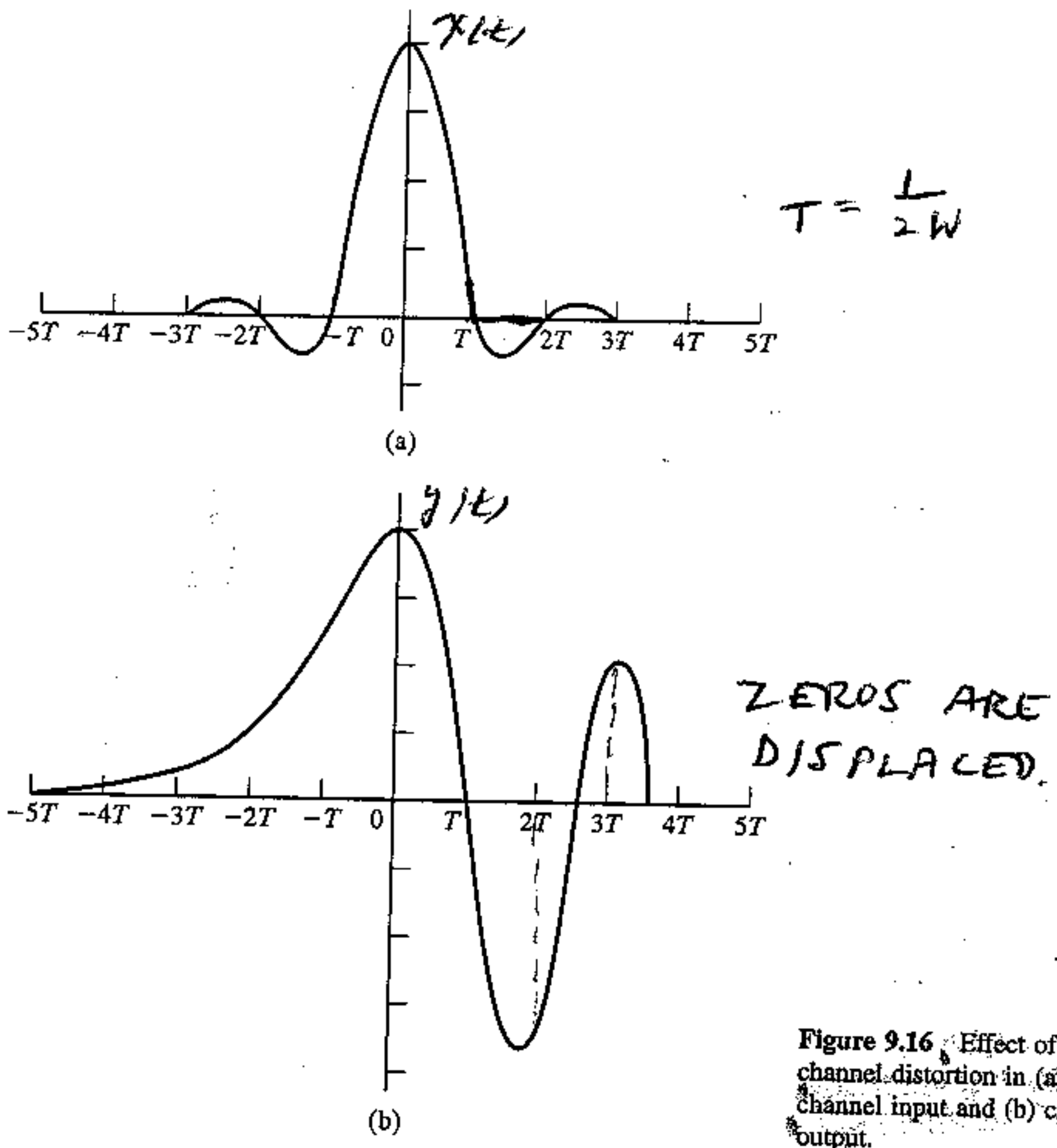


Figure 9.16 Effect of channel distortion in (a) channel input and (b) channel output.

DESIGN OF $G_R(f)$, $G_T(f)$

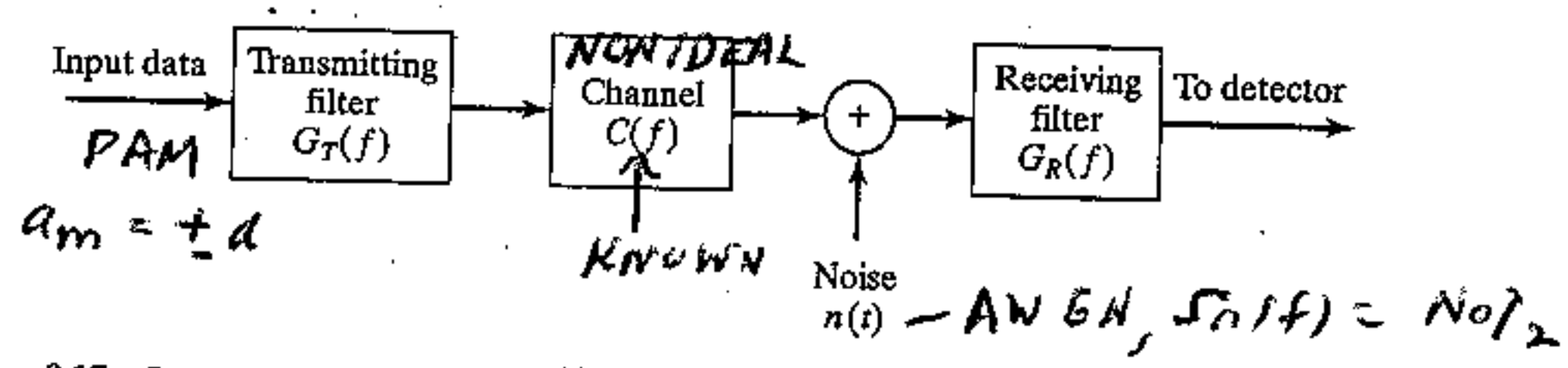


Figure 9.17 System configuration for the design of $G_T(f)$ and $G_R(f)$.

HOW DO WE CHOOSE $G_T(f)$, $G_R(f)$ IF CHANNEL IS NONIDEAL BUT KNOWN?
 WHAT IS LOSS DUE TO CHANNEL?

APPROACH:

1) GUARANTEE NO ISI

$$\Rightarrow G_T(f) C(f) G_R(f) = X_{rc}(f) e^{-j2\pi f t_0} \quad |f| \leq W$$

(t_0 CHOSEN FOR REALIZABILITY)

2) CHOOSE

$$G_T(f) C(f) = \sqrt{X_{rc}(f)} e^{-j2\pi f t_0} \quad \text{WHY?}$$

3) CHOOSE

$$G_R(f) = \sqrt{X_{rc}(f)} e^{-j2\pi f t_r} \quad \text{WHY?}$$

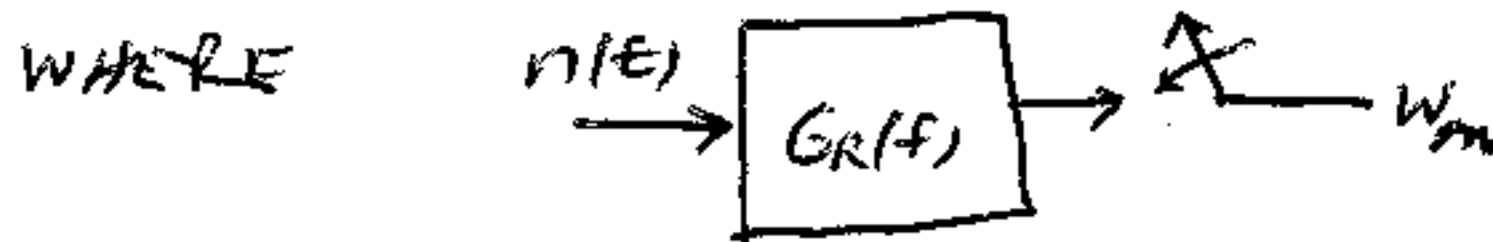
(NOTE: THIS IS A MATCHED FILTER,
MATCHED TO CHANNEL OUTPUT)

TO DETERMINE LOSS DUE TO NONIDEAL
CHANNEL, WITH NO ISI,

$$y_m = a_m + w_m = \pm d + w_m$$

$$\begin{aligned} \text{DECIDE } +d & \text{ IF } y_m > 0 & (\text{EQUIPROBABLE} \\ & & \text{SIGNALS)} \\ & -d & \text{ IF } y_m < 0 \end{aligned}$$

$$y_m \sim N(\pm d, \sigma_w^2)$$



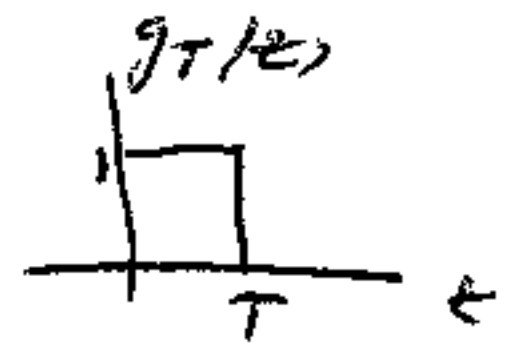
$$\sigma_w^2 = \int_{-\infty}^{\infty} |G_R(f)|^2 \frac{N_0}{2} df = \frac{N_0}{2} \int_{-W}^W X_{RC}(f) df$$

$$P_2 = Q\left(\frac{d-0}{\sigma_w}\right) = Q\left(\sqrt{d^2/\sigma_w^2}\right)$$

PERFORMANCE IMPROVES WITH INCREASING d^2/σ_w^2 . HOLDING AVERAGE TRANSMIT POWER FIXED, WE CAN FWD DEGRADATION.

RECALL THAT $s(t) = a_m g_T(t)$

$$P_{AV} =$$



$$\text{AVERAGE POWER} = E[a_m^2] \frac{1}{T} \int_{-\infty}^{\infty} g_T^2(t) dt$$

$$= \frac{d^2}{T} \int_{-W}^W |G_T(f)|^2 df = \frac{d^2}{T} \int_{-W}^W \frac{X_{RC}(f)}{|C(f)|^2} df$$

$$\Rightarrow d^2 = \frac{P_{AV} T}{\int_{-W}^W \frac{X_{RC}(f)}{|C(f)|^2} df}$$

$$\text{ALSO, } \sigma_w^2 = \frac{N_0}{2} \int_{-W}^W X_{RC}(f) df$$

$$= \frac{N_0}{2} \int_{-W}^W X_{RC}(f) e^{j2\pi f t} df \Big|_{t=0}$$

$$X_{RC}(0) = 1$$

$$= N_0/2$$

$$\frac{d^2}{\sigma_w^2} = \frac{2 P_{AV} T / N_0}{\int_{-W}^W \frac{X_{RC}(f)}{|C(f)|^2} df}$$

FOR IDEAL CHANNEL $|C(f)| = 1 \Rightarrow \frac{d^2}{\sigma_w^2} = \frac{2 P_{AV} T}{N_0}$

THUS DEGRADATION IN "SNR" d^2/σ_w^2 IS

$$10 \log_{10} \int_{-W}^W \frac{X_{RC}(f)}{|C(f)|^2} df \text{ dB}$$

NOTE THAT THERE IS NO LOSS DUE TO PHASE DISTORTION SINCE THE TRANSMIT FILTER IS

$$\begin{aligned} G_T(f) &= \frac{\sqrt{X_{RC}(f)}}{C(f)} e^{-j2\pi f t_0} \\ &= \frac{\sqrt{X_{RC}(f)}}{|C(f)| e^{j\theta(f)}} = \frac{\sqrt{X_{RC}(f)}}{|C(f)|} e^{-j\theta(f) + j\theta(f) - j2\pi f t_0} \end{aligned}$$

↑
PHASE COMPENSATION

WHERE DOES LOSS COME FROM?

ASSUMING $|C(f)| < 1$ (NO AMPLIFIERS IN CHANNEL) AND $X_{RC}(f)$ IS FIXED

$$P_{AV} = \text{CONSTANT} = \frac{d^2}{T} \int_{-W}^W \frac{X_{RC}(f)}{|C(f)|^2} df$$

AS $|C(f)|$ DECREASES, $\int_{-W}^W \frac{X_R(f)}{|C(f)|^2} df$

INCREASES AND SO d^2 MUST BE DECREASED
TO MAINTAIN AVERAGE TRANSMIT POWER.

NOTE: WOULD BE BETTER TO OPTIMIZE
OVER CHOICE OF $G(f)$ (WAS JUST
SELECTED WITHOUT JUSTIFICATION)

CHANNEL EQUALIZATION

WHAT DO WE DO IF $C(f)$ IS UNKNOWN?
COULD EVEN BE TIME VARYING, FOR
EXAMPLE, TELEPHONE CALL SETUP -
CONNECTIONS SWITCHED AS NEEDED, OR
CELL PHONES FROM CARS THAT ARE
MOVING.

ONE APPROACH IS TO USE EQUALIZING
FILTER AT RECEIVER, WHICH MAY BE
ADAPTIVE (TO ACCOMMODATE TIME-VARYING
CHANNELS)

WITHOUT EQUALIZER WE HAVE

$$y_m = x_0 a_m + \sum_{\substack{n=-\infty \\ n \neq m}}^{\infty} a_n x_{m-n} + w_m$$

OR

$$\begin{aligned}
 y_m &= \sum_{n=-\infty}^{\infty} a_n x_{m-n} + W_m \\
 &= \sum_{n=-\infty}^{\infty} x_n a_{m-n} + W_m \\
 &= x_0 a_m + \underbrace{\sum_{\substack{n=-\infty \\ n \neq 0}}^{\infty} x_n a_{m-n}}_{\text{ISI}} + W_m
 \end{aligned}$$

THIS IS WHAT IS OBTAINED IF WE IGNORE EFFECT OF NONIDEAL CHANNEL BY LETTING $G_T(f)G_R(f) = X_{RC}(f)e^{-j2\pi ft_0}$

USUALLY ISI ONLY EXTENDS INTO A FEW ADJACENT SYMBOL INTERVALS OR

$$y_m = x_0 a_m + \sum_{\substack{n=-L_1 \\ n \neq 0}}^{L_2} x_n a_{m-n} + W_m$$

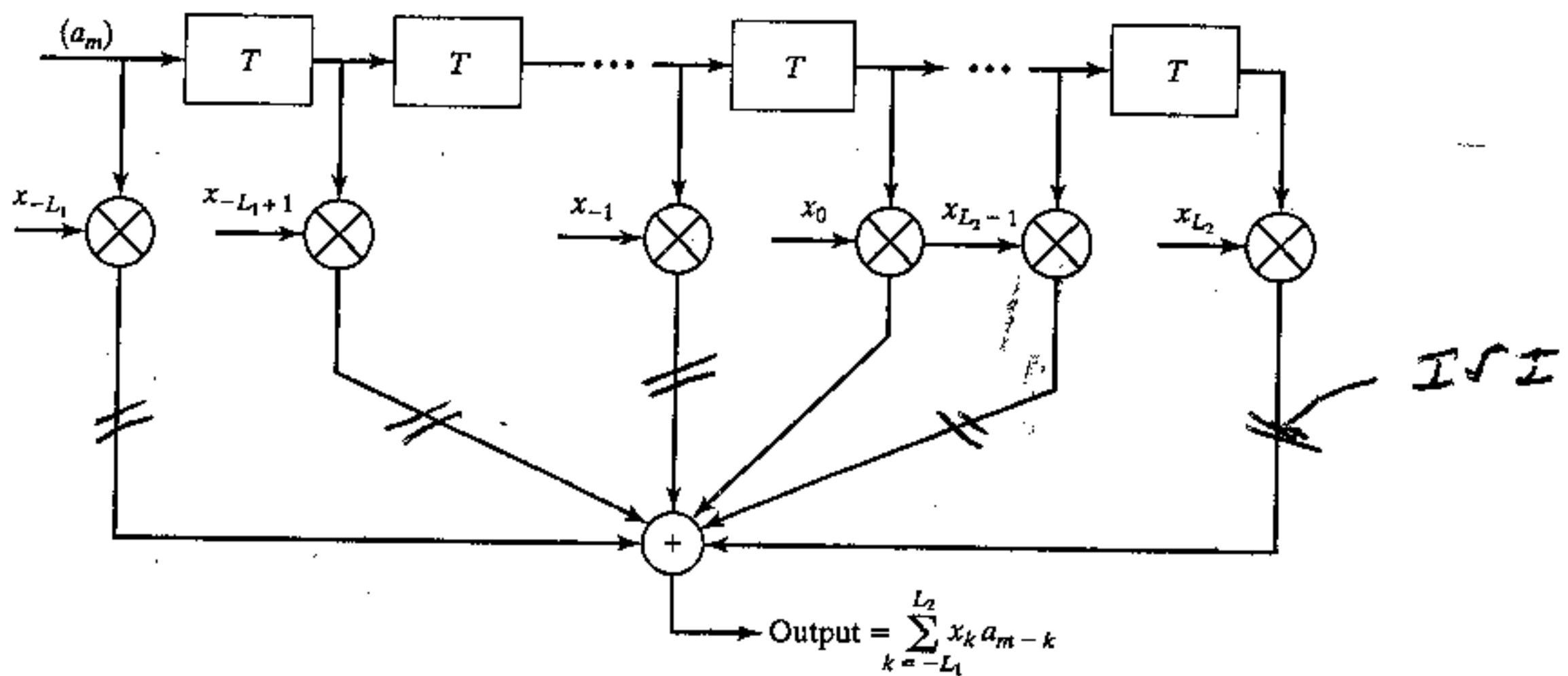
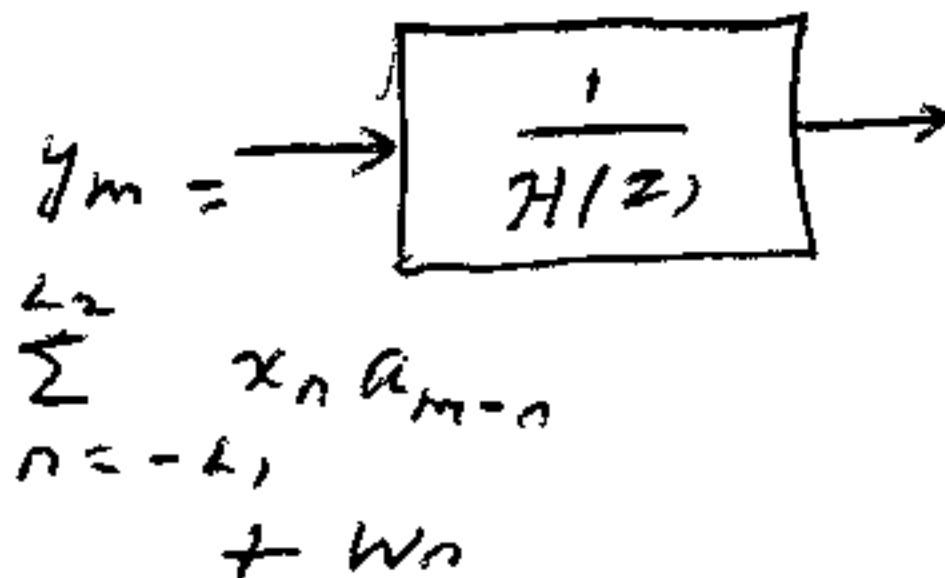


Figure 9.18 Equivalent discrete-time channel filter. (JUST AN FIR FILTER)

$$H(z) = \sum_{k=-L_1}^{L_2} x_k z^{-k}$$

WANT $H(z) = x_0$ OR $x_k = 0$ FOR $k \neq 0$.

ONE POSSIBLE APPROACH IS TO INSERT EQUALIZING FILTER AT RECEIVER,



IIR FILTER NOT PRACTICAL, ALSO WILL ENHANCE NOISE

$$Y(z) = H(z)A(z) + W(z)$$

↑ UNDO THIS

BETTER PRACTICAL SOLUTION IS FIR ANALOG EQUALIZER, $G_E(z) = \sum_{n=-N}^N c_n z^{-n}$ OR

$$g_E(t) = \sum_{n=-N}^N c_n \delta(t - nT)$$

SO THAT AT THE OUTPUT OF EQUALIZER

$$\begin{aligned} g(t) &= x(t) * g_E(t) \\ &= x(t) * \sum_{n=-N}^N c_n \delta(t - nT) \\ &= \sum_{n=-N}^N c_n x(t - nT) \end{aligned}$$

AND WHEN SAMPLED

$$q(mT) = \sum_{n=-N}^N c_n x(mT - nT)$$

$$= \begin{cases} 1 & m = 0 \\ 0 & m \neq 0 \end{cases}$$

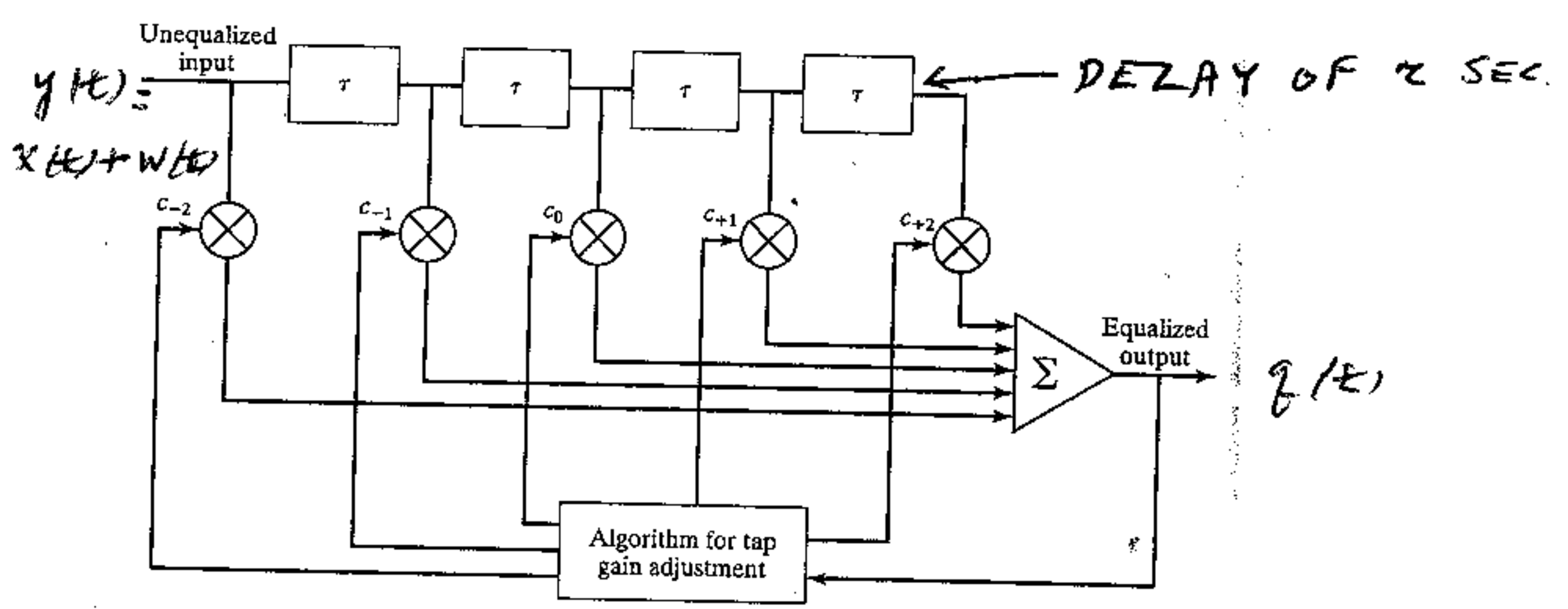


Figure 9.20 Linear transversal filter.

$\tau = T$ OR $T/2$ (FRACTIONALLY SPACED TAPS)

NOT POSSIBLE TO HAVE $q(mT) = 0$ FOR ALL $m \neq 0$. INSTEAD, FORCE $q(mT) = 0$ FOR $m = \pm 1, \pm 2, \dots, \pm N$ AND $q(0) = 1$
 \Rightarrow CALLED ZERO-FORCING EQUALIZER.

HAVE $2N + 1$ COEFFICIENTS AND $2N + 1$ CONSTRAINTS \Rightarrow CAN NOW SOLVE FOR c_n 'S.

EXAMPLE : $q(mT) = \sum_{n=-2}^2 c_n x(mT - nT)$

| | | | | | | | |
|-----------------|---------|---------|----------|----------|----------|---|---|
| $m \setminus n$ | -2 | -1 | 0 | 1 | 2 | | |
| -2 | $x(0)$ | $x(-T)$ | $x(-2T)$ | $x(-3T)$ | $x(-4T)$ | $\begin{bmatrix} c_{-2} \\ c_{-1} \\ c_0 \\ c_1 \\ c_2 \end{bmatrix}$ | $= \begin{bmatrix} 0 \\ 0 \\ 1 \\ 0 \\ 0 \end{bmatrix}$ |
| -1 | $x(T)$ | $x(0)$ | $x(-T)$ | $x(-2T)$ | $x(-3T)$ | | |
| 0 | $x(2T)$ | $x(T)$ | $x(0)$ | $x(-T)$ | $x(-2T)$ | | |
| 1 | $x(3T)$ | $x(2T)$ | $x(T)$ | $x(0)$ | $x(-T)$ | | |
| 2 | $x(4T)$ | $x(3T)$ | $x(2T)$ | $x(T)$ | $x(0)$ | | |

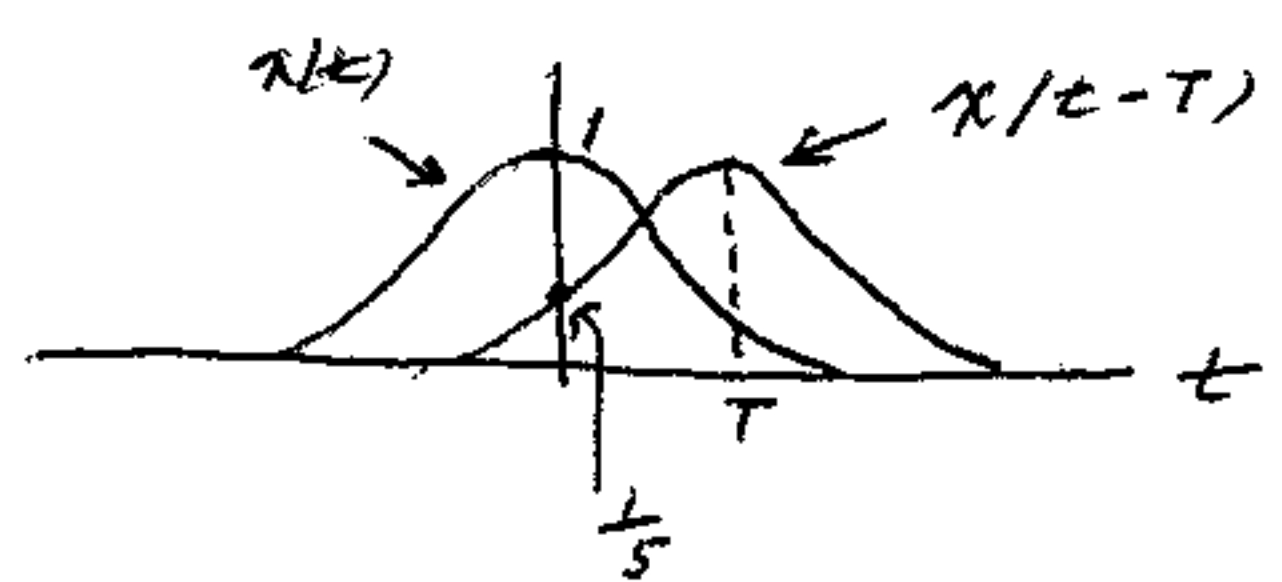
$\underbrace{\hspace{15em}}_{\underline{x}} \quad \underbrace{\hspace{2em}}_{\underline{c}} \quad \underbrace{\hspace{2em}}_{\underline{q}}$

CAN EASILY SOLVE FOR c_n 'S.

EXAMPLE : FRACTIONALLY SPACED EQUALIZER
 $\tau = T/2$. SAMPLING RATE AT
 EQUALIZER OUTPUT IS $1/\tau = 2 \times \frac{1}{T}$
 OR TWICE AS LARGE - REDUCES
 ALIASING FREQUENCIES IF $\alpha = 1$
 USED FOR RAISED COSINE PULSE.

$$x(t) = \frac{1}{1 + (2t/T)^2}$$

$\frac{1}{T}$ = SYMBOL RATE
 $2/T$ = OUTPUT RATE
 OF EQUALIZER



$$q(mT) = \sum_{n=-2}^2 c_n x(mT - nT/2)$$

$$\tau = T/2$$

$$= 1 \quad m = 0$$

$$= 0 \quad m = \pm 1, \pm 2$$

$$\text{NEED } x((m-n/2)T) = \frac{1}{1 + \left[\frac{2(m-n/2)T}{T} \right]^2}$$

$$= \frac{1}{1 + (2m-n)^2} = [X]_{mn}$$

$m = -2, \dots, 2$
 $n = -2, \dots, 2$

FOR EXAMPLE $m = -2, n = -2$

$$[X]_{-2,-2} = \frac{1}{1 + (-4+2)^2} = \frac{1}{5}$$

$$[X]_{-2,0} = \frac{1}{1 + (-4-0)^2} = \frac{1}{17}$$

SEE BOOK FOR X . SOLUTION IS

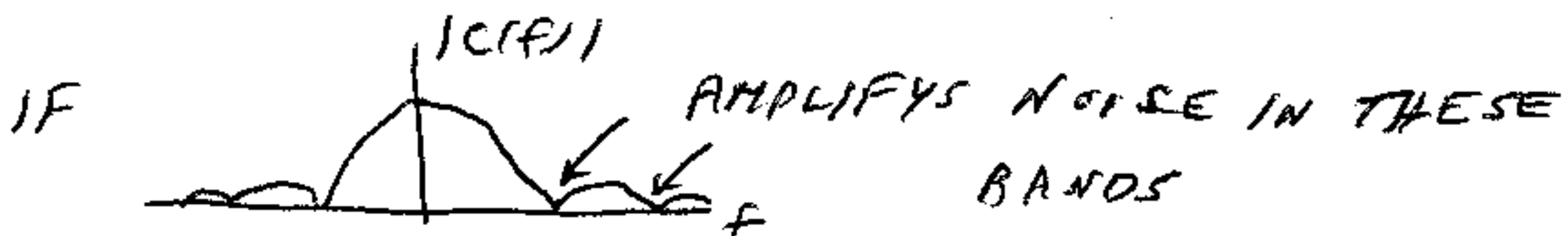
$$C_{\text{opt}} = \begin{bmatrix} -2.2 \\ 4.9 \\ -3 \\ 4.9 \\ -2.2 \end{bmatrix}$$

FREQ. RESPONSE?

MMSE EQUALIZERS

PROBLEM WITH PREVIOUS EQUALIZER - NO ACCOUNTING FOR NOISE. ATTEMPTS TO IMPLEMENT

$$G_E(f) = \frac{1}{C(f)}$$



BETTER TO INCLUDE NOISE EFFECTS BY USING MEAN SQUARE ERROR AS DESIGN CRITERION.

OUTPUT OF EQUALIZER IS -

$$z(z) = \sum_{n=-N}^N c_n y(z-nT)$$

$$\Rightarrow z(mT) = \sum_{n=-N}^N c_n y(mT-nT)$$

DESIRED OUTPUT IS TRANSMITTED SYMBOL a_m . HENCE, CHOOSE c_n 'S TO MINIMIZE MSE

$$MSE = E \left[(z(mT) - a_m)^2 \right]$$

$$= E \left[\left(\sum_{n=-N}^N c_n y(mT-nT) - a_m \right)^2 \right]$$

$$= E \left[\sum_n \sum_k c_n c_k y(mT-nT) y(mT-kT) - 2 \sum_n c_n y(mT-nT) a_m + a_m^2 \right]$$

$$= \sum_n \sum_k c_n c_k E \left[y(mT-nT) y(mT-kT) \right] - 2 \sum_n c_n E \left[y(mT-nT) a_m \right] + E \left[a_m^2 \right]$$

SINCE a_m IS IID SEQUENCE AND NOISE IS AWGN, CAN SHOW $a_m, y(t)$ ARE JOINTLY STATIONARY.

$$MSE = \sum_n \sum_k c_n c_k R_y(n-k) - 2 \sum_k c_k R_{AY}(k) + E(a_m^2)$$

$$\text{WHERE } R_y(n-k) = E[y(mT-n\tau)y(mT-k\tau)]$$

$$R_{AY}(k) = E[y(mT-k\tau)a_m]$$

TO FIND c_n 'S DIFFERENTIATE

$$\frac{\partial MSE}{\partial c_l} = \sum_n \sum_k (c_n \delta_{kl} + \delta_{nl} c_k) R_y(n-k) - 2 R_{AY}(l) + 0$$

$$= \sum_n c_n R_y(n-l) + \sum_k c_k R_y(l-k) - 2 R_{AY}(l)$$

$$= 2 \left(\sum_n c_n R_y(n-l) - R_{AY}(l) \right) = 0$$

$$l = -N, \dots, N$$

$$\sum_{n=-N}^N c_n R_y(l-n) = R_{AY}(l)$$

$$l = -N, \dots, N$$

$2N+1$ LINEAR EQUATIONS IN $2N+1$ UNKNOWN \Rightarrow SOLVE FOR c_n 'S

TO FIND $R_Y(n)$, $R_{AY}(n)$ WE USE TEST SIGNALS (BEFORE DATA IS SENT) AND USE ESTIMATES

$$\hat{R}_Y(n) = \frac{1}{K} \sum_{k=1}^K y(kT - nT) y(kT)$$

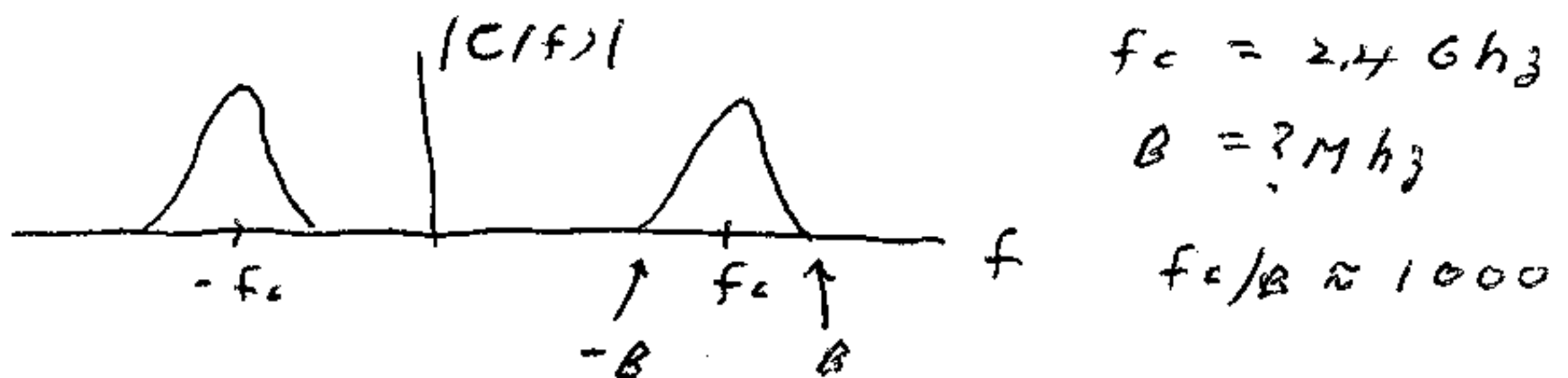
$$\hat{R}_{AY}(n) = \frac{1}{K} \sum_{k=1}^K y(kT - nT) a_k$$

↑ KNOWN TO RECEIVER

SEE BOOK FOR ADAPTIVE VERSION OF THIS EQUALIZER.

CHAPTER 10 - CARRIER MODULATION (10.1-10.5 - SELECTED TOPICS)

CONSIDERED BASEBAND CHANNELS - PASSBANDS WERE NEAR $f=0$, E.G., WIRELINE. NOW CONSIDER BANDPASS CHANNELS, E.G., WIRELESS COMMUNICATIONS AT 2.4 GHz (G = 10^9 = "GIGA")



TO TRANSMIT THROUGH CHANNEL A BASEBAND SIGNAL :

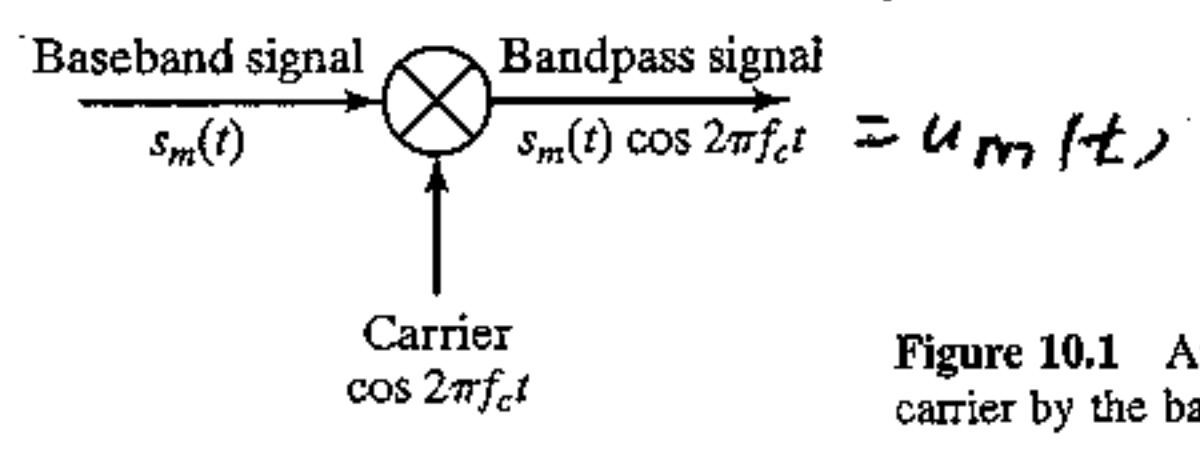


Figure 10.1 Amplitude modulation of the sinusoidal carrier by the baseband signal.

$$u_m(t) = s_m(t) \cos 2\pi f_c t \quad f_c = \text{CARRIER FREQ.}$$

$$= s_m(t) \left(\frac{1}{2} e^{j2\pi f_c t} + \frac{1}{2} e^{-j2\pi f_c t} \right)$$

$$\Rightarrow U_m(f) = \frac{1}{2} S_m(f - f_c) + \frac{1}{2} S_m(f + f_c)$$

THIS IS A DOUBLE SIDEBAND SUPPRESSED CARRIER AM SIGNAL.

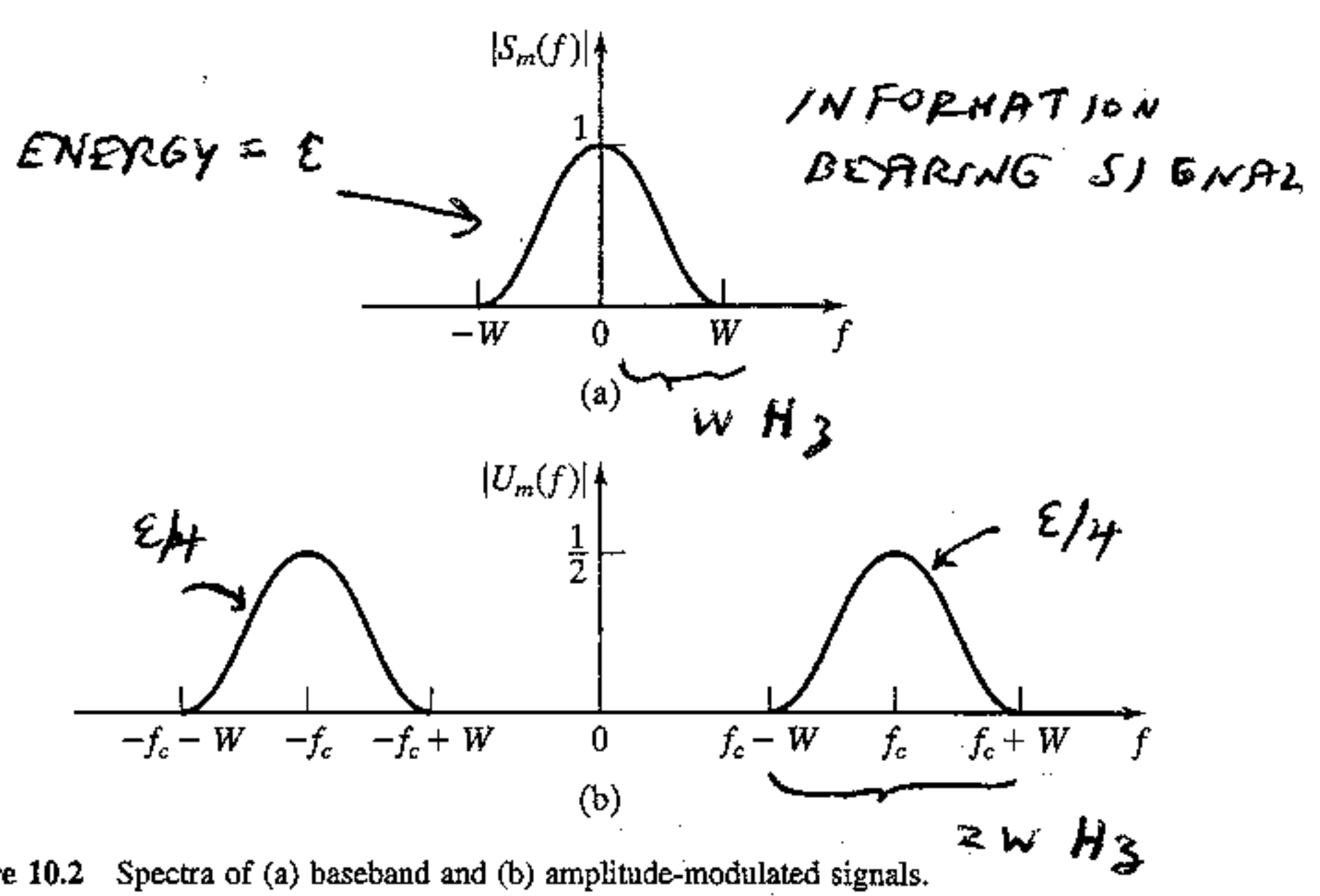
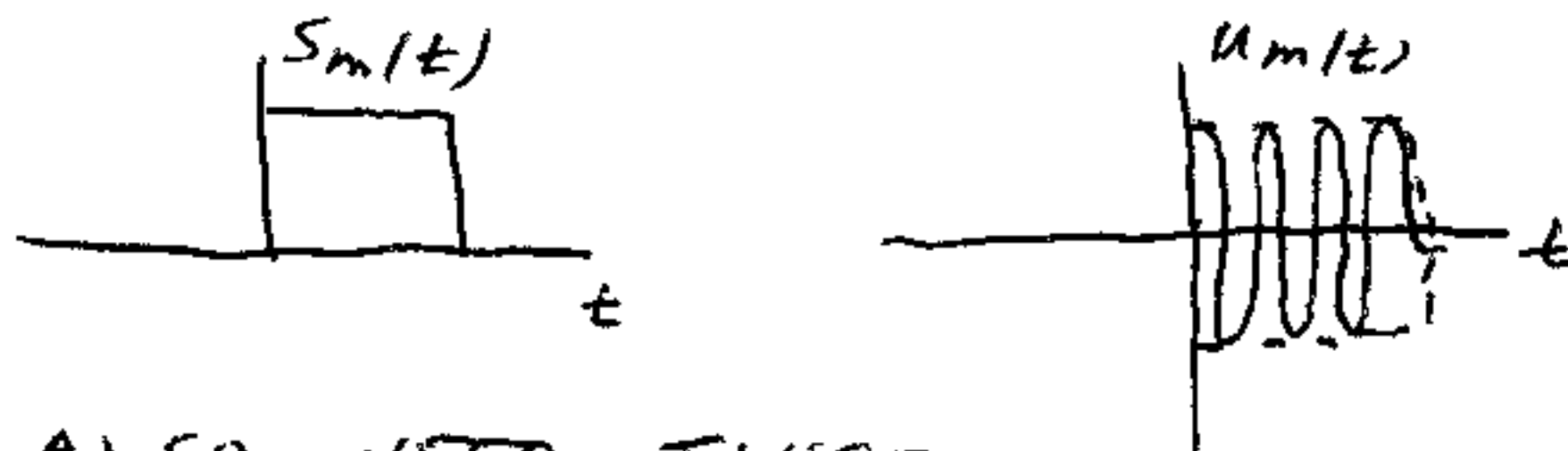


Figure 10.2 Spectra of (a) baseband and (b) amplitude-modulated signals.

$$E = \int_{-\infty}^{\infty} |S_m(f)|^2 df \quad \left(= \int_{-\infty}^{\infty} s_m^2(t) dt \right)$$

$\frac{1}{2}$ OF ENERGY LOST DUE TO CARRIER



ALSO NEED TWICE
THE BANDWIDTH.

GEOMETRIC REPRESENTATION

LET $s_m(t) = \sqrt{E_s} \psi_m(t)$ $m = 1, 2, \dots, M$
 \uparrow
 ORTHONORMAL BASEBAND
 BANDPASS

WANT SET OF ORTHONORMAL WAVEFORMS

$$\Rightarrow u_m(t) = s_m(t) \cos 2\pi f_c t$$

$$= \sqrt{E_s} \psi_m(t) \cos 2\pi f_c t$$

ASIDE: $u_m(t)$ 'S STILL ORTHOGONAL SINCE

$$\int_{-\infty}^{\infty} \psi_i(t) \cos 2\pi f_c t \psi_j(t) \cos 2\pi f_c t dt =$$

$$\int_{-\infty}^{\infty} \psi_i(t) \psi_j(t) \left(\frac{1}{2} \cos 4\pi f_c t + \frac{1}{2} \right) dt$$

$$\approx \frac{1}{2} \int_{-\infty}^{\infty} \psi_i(t) \psi_j(t) dt = \frac{1}{2} \delta_{ij}$$

TO MAKE ORTHONORMAL, LET

$$\psi_m(t) \rightarrow \sqrt{2} \psi_m(t) \text{ OR}$$

$$s_m(t) = \sqrt{\frac{E_s}{2}} \underbrace{\sqrt{2} \psi_m(t)}_{\psi'_m(t)}$$

$$\Rightarrow u_m(t) = \sqrt{E_s/2} \underbrace{\psi'_m(t) \cos 2\pi f_c t}_{\psi_{c,m}(t)}$$

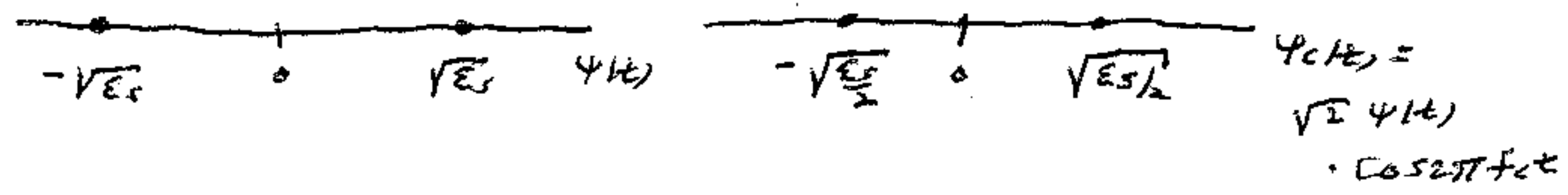
$\psi_{c,m}(t)$ IS BASIS SIGNAL FOR BANDPASS SIGNAL AND NOW

$$\int_{-\infty}^{\infty} \psi_{c,i}(t) \psi_{c,j}(t) dt = \delta_{ij}$$

AND E_s = ENERGY OF $s_m(t)$ - BASEBAND SIGNAL

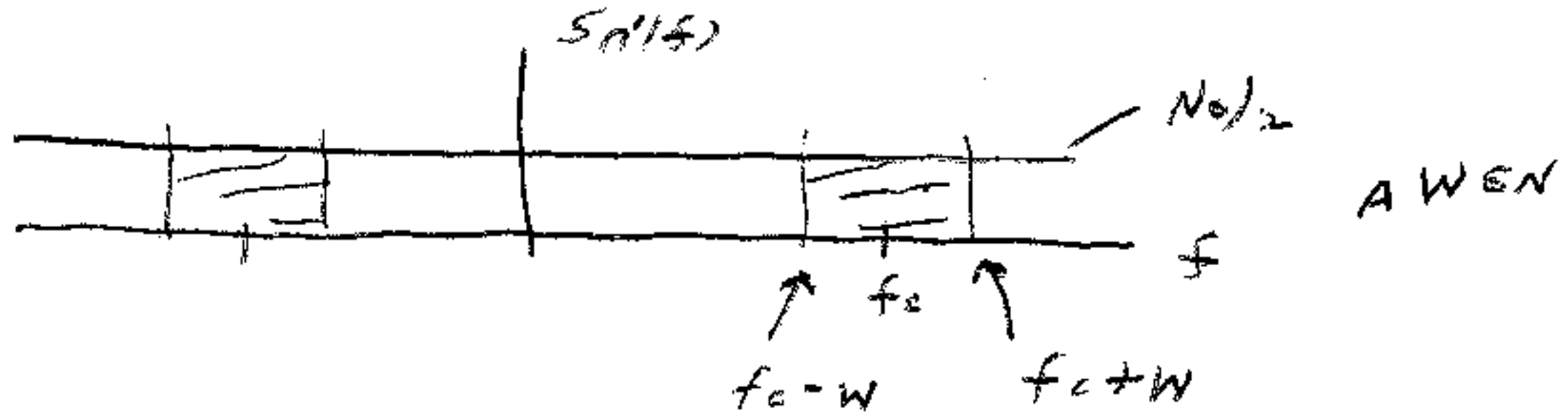
$E_s/2$ = ENERGY OF BANDPASS SIGNAL

EXAMPLE : BINARY PAM

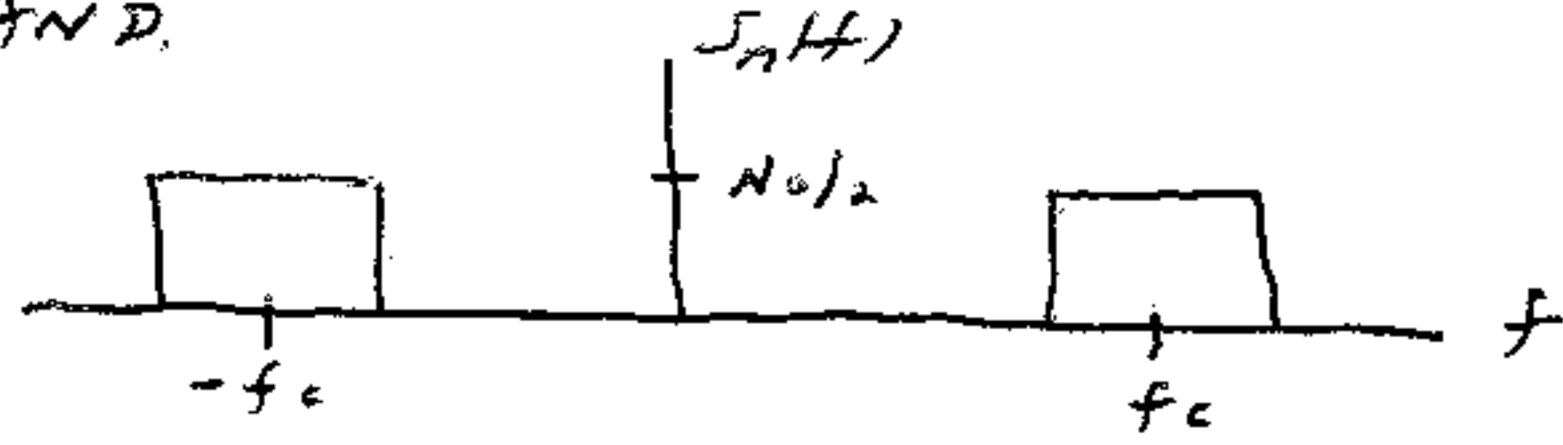


DEMODULATION AND DETECTION

NEED REPRESENTATION OF BANDPASS NOISE



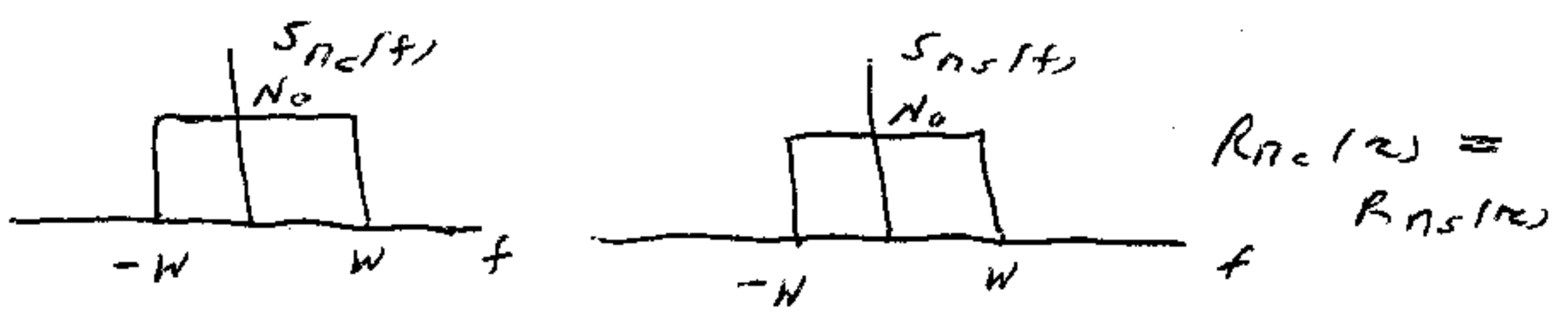
BANDPASS SIGNAL WILL BE FILTERED TO GET RID OF NOISE OUTSIDE OF $(f_c - W, f_c + W)$ BAND.



THIS RANDOM PROCESS CAN BE REPRESENTED AS

$$n(t) = n_c(t) \cos 2\pi f_c t - n_s(t) \sin 2\pi f_c t$$

WHERE $n_c(t)$, $n_s(t)$ ARE EACH GAUSSIAN, ARE INDEPENDENT OF EACH OTHER, AND BOTH ARE ZERO MEAN AND WIDE SENSE STATIONARY WITH PSDs



NOTE THAT $n_c(t)$ (IN-PHASE COMPONENT) AND $n_s(t)$ (QUADRATURE COMPONENT) ARE LOWPASS RANDOM PROCESSES

- 1) SINCE $n_s(t)$, $n_c(t)$ ARE GAUSSIAN RANDOM PROCESS $\Rightarrow n(t)$ IS GAUSSIAN