

EE206: Communications Principles Tutorial

Assignment 8&9

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1. An FM signal has an amplitude of 5 volts, carrier frequency of 120 MHz and peak frequency deviation of 75 KHz. The modulating message signal is a single tone of 4 KHz with amplitude of 2 volts. The FM signal is transmitted through a transmission line using a bandwidth based on Carson's rule. The gain of the transmission line is 1 in the pass-band. At the output of the transmission line, the FM signal is demodulated using a demodulator including a limiter, differentiator, envelop detector, D.C. blocking and an ideal low-pass filter which has a bandwidth of 4 KHz and unity gain in the pass-band. The signal amplitude at the output of the limiter is assumed to be 2 volt. The white noise at the input of the transmission line has a double-sided PSD of $\eta/2 = 10^{-6}$ W/Hz.

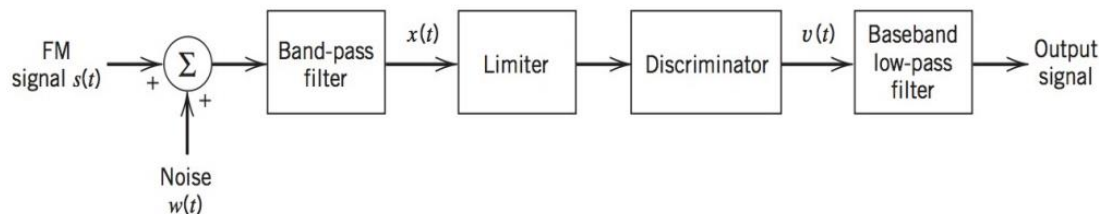
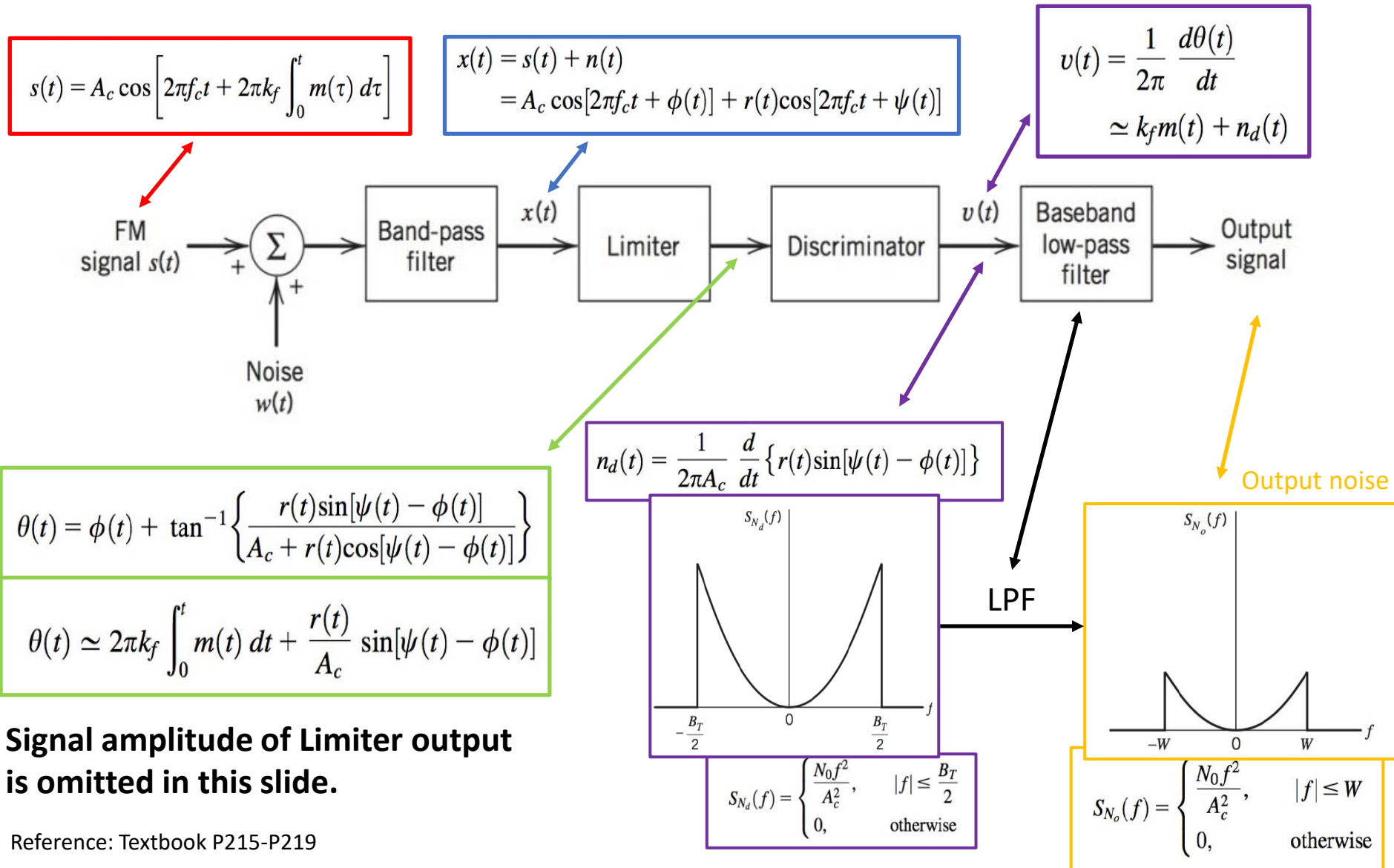


FIGURE 6.7 Noisy model of an FM receiver.



Signal amplitude of Limiter output is omitted in this slide.

Reference: Textbook P215-P219

a. Determine the SNR at the input and the output of the demodulator, respectively.

$$A_c = 5\text{V}, \quad f_c = 120\text{MHz}$$

$$\Delta f = k_f A_m = 75\text{kHz}, \quad A_m = 2\text{V}, \quad f_m = 4\text{kHz}$$

$$\frac{\eta}{2} = 10^{-6}\text{W/Hz}$$

By Carson's Rule,

$$\text{BW} = 2(\beta + 1)f_m = 2\left(\frac{\Delta f}{f_m} + 1\right)f_m = 2\left(\frac{75\text{k}}{4\text{k}} + 1\right)4\text{k} = 158\text{kHz}$$

Input signal power:

$$S_i = \frac{1}{2}A_c^2 = 12.5\text{W}$$

Input noise power:

$$N_i = \frac{\eta}{2} \times 2 \times \text{BW} = 2 \times 10^{-6} \times 158\text{k} = 0.316\text{W}$$

Input SNR:

$$\text{SNR}_i = \frac{S_i}{N_i} = 15.97\text{dB}$$

- a. Determine the SNR at the input and the output of the demodulator, respectively.

Output signal:

$$y(t) = A_l 2\pi k_f m(t)$$

where $V_l = 2V$, denotes the signal amplitude at the output of the limiter.

Output signal power:

$$S_o = \frac{1}{2} A_l^2 (2\pi k_f)^2 A_m^2 = \frac{1}{2} A_l^2 (2\pi \Delta f)^2$$

Output noise power:

$$N_o = \frac{A_l^2 8\pi^2 \eta f_{LPF}^3}{3A_c^2}$$

Output SNR:

$$SNR_o = \frac{S_o}{N_o} = \frac{3\Delta f^2 A_c^2}{4\eta f_{LPF}^3} = 59.16\text{dB}$$

b. Determine if the system is under the threshold.

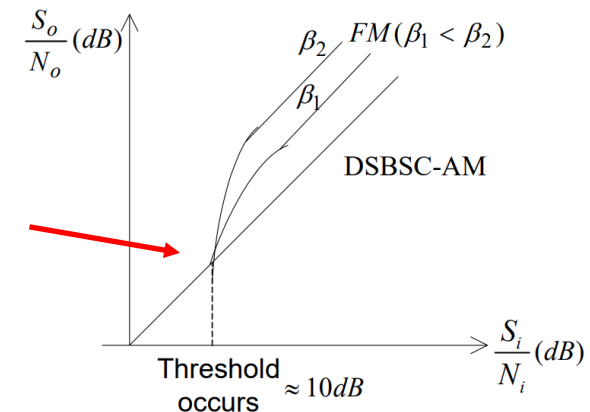
Threshold Phenomenon

- The output SNR formula was derived based on the small noise assumption. For large noise, “**threshold phenomenon**” occurs in the FM receiver.
- As discussed in the capture effect, when noise is small, the desired signal captures the noise.
- When noise is strong, it can effectively capture the signal.
- The “**threshold**” is defined as the **minimum** S_i / N_i that validates the output SNR formula and makes the FM receiver work properly.
- This threshold occurs when $\frac{S_i}{N_i} \approx 10 \text{ dB}$.

Answer: Because $SNR_i > 10 \text{ dB}$, the system is not under the threshold.

Textbook P217:

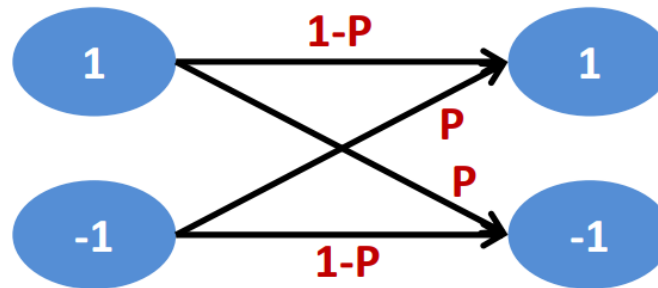
We assume that the carrier-to-noise ratio measured at the discriminator input is large compared with unity.



- D1.1

Given the following communication channel, if each information bit is repeated 4 times (code rate = 1/5) at the transmitter and $P=0.9$, how to achieve a good bit error rate at the receiver? What will the bit error rate be?

Binary Symmetric Channel:



Invert -1, 1 in receiver.

$$\begin{aligned}
 & \Pr(\text{Error Occurs}) \\
 &= \Pr\left(\text{More than } \frac{n}{2} \text{ bits are wrong}\right) \\
 &= \sum_{i=\lceil \frac{n}{2} \rceil}^n \Pr(i \text{ bits are wrong}) \\
 &= \sum_{i=\lceil \frac{n}{2} \rceil}^n \binom{n}{i} P^i (1-P)^{n-i}
 \end{aligned}$$

- D2.1

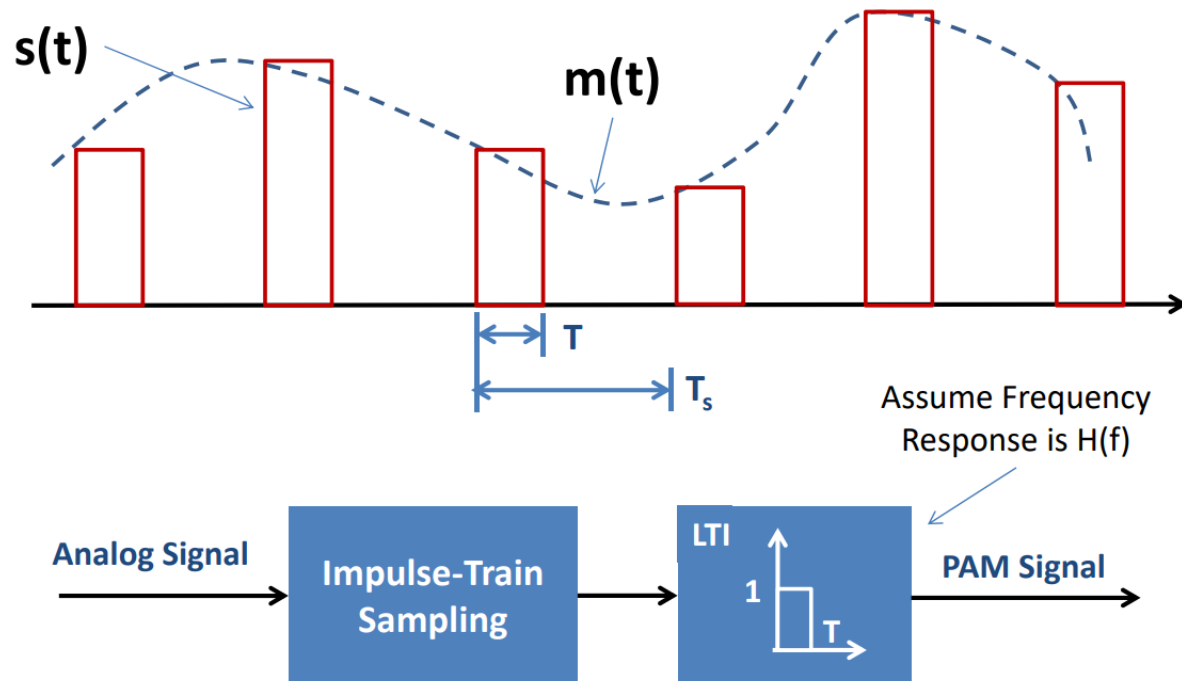
Plot the spectrum of a PAM wave produced from the following modulating signal

$$m(t) = A_m \cos(2\pi f_m t)$$

assuming $f_m = 0.2\text{Hz}$, PAM sampling period $T_s = 1\text{s}$, and pulse duration $T = 0.45\text{s}$.

Pulse Amplitude Modulation

- **PAM**: sampled signal value is represented by the amplitude of pulses



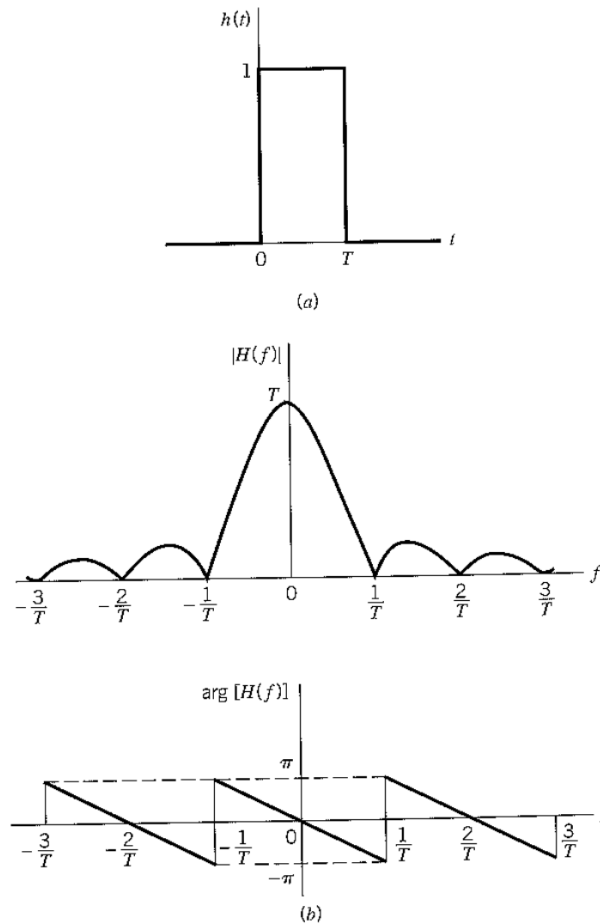


FIGURE 7.6 (a) Rectangular pulse $h(t)$. (b) Spectrum $H(f)$.

$$h(t) = \begin{cases} 1, & 0 < t < T \\ \frac{1}{2}, & t = 0, t = T \\ 0, & \text{otherwise} \end{cases}$$

By definition, the instantaneously sampled version of $m(t)$ is given by

$$m_{\delta}(t) = \sum_{n=-\infty}^{\infty} m(nT_s) \delta(t - nT_s) \quad (7.12)$$

$$\begin{aligned} m_{\delta}(t) \star h(t) &= \int_{-\infty}^{\infty} m_{\delta}(\tau) h(t - \tau) d\tau \\ &= \int_{-\infty}^{\infty} \sum_{n=-\infty}^{\infty} m(nT_s) \delta(\tau - nT_s) h(t - \tau) d\tau \quad (7.13) \\ &= \sum_{n=-\infty}^{\infty} m(nT_s) \int_{-\infty}^{\infty} \delta(\tau - nT_s) h(t - \tau) d\tau \end{aligned}$$

Using the sifting property of the delta function, we thus obtain

$$m_{\delta}(t) \star h(t) = \sum_{n=-\infty}^{\infty} m(nT_s) h(t - nT_s) \quad (7.14)$$

$$S(f) = M_{\delta}(f) H(f) \quad (7.16)$$

$$M_{\delta}(f) = f_s \sum_{k=-\infty}^{\infty} M(f - kf_s) \quad (7.17)$$

TextBook
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