

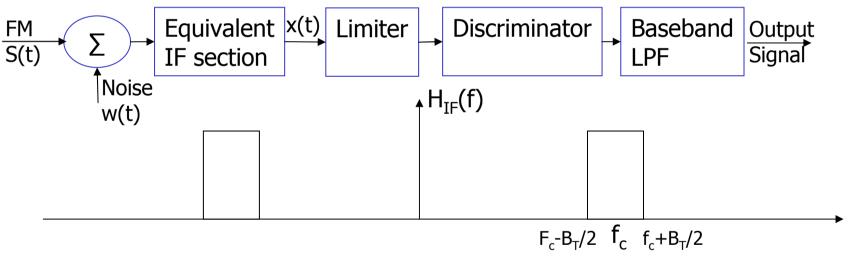
# **Communication Systems II**

Noise in CW modulation FM Modulation

# 4

### **FM Receiver Model**

- W(t) is modeled as white Gaussian noise with zero mean and PSD of N₀/2.
- FM signal of center frequency  $f_c$  and BW = $B_T$



- IF filter is assumed IBPF with bandwidth B<sub>T</sub>
  - We can use the narrow band noise representation in terms of its inphase and quadrature components.

The NBN at the IF output is defined as

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

Equivalently

$$n(t) = r(t)\cos(2\pi f_c t + \psi(t))$$

$$r(t) = \sqrt{n_c^2(t) + n_s^2(t)}$$

$$\psi(t) = \tan^{-1}\frac{n_s(t)}{n_c(t)}$$

• r(t) is Rayleigh distributed and phase is uniformely distributed over  $2\pi$ 

The FM signal at the IF output is

$$s(t) = A_c \cos(2\pi f_c t + 2\pi k_f \int_0^t m(t)dt)$$

$$\phi(t) = 2\pi k_f \int_0^t m(t)dt$$

$$s(t) = A_c \cos(2\pi f_c t + \phi(t))$$

- The total signal at the output of IF section is
  - X(t)=s(t)+n(t)

$$x(t) = A_c \cos(2\pi f_c t + \phi(t)) + r(t) \cos(2\pi f_c t + \psi(t))$$

Represent x(t) by means of a phasor diagram

$$x(t) = A_c \cos(2\pi f_c t + \phi(t)) + r(t) \cos(2\pi f_c t + \psi(t))$$

• The relevant phase  $\theta(t)$  can be calculated

$$\theta(t) = \phi(t) + \tan^{-1} \left\{ \frac{r(t)\sin(\psi(t) - \phi(t))}{A_c + r(t)\cos(\psi(t) - \phi(t))} \right\}$$

resultant 
$$r(t)$$

$$\theta(t)-\phi(t)$$

$$A_{c}$$

- Envelope of x(t) is not of interest to us
- The output of the ideal discriminator will be proportional to θ'(t)/2π

The output of the ideal discriminator 
$$\theta(t) = \phi(t) + \tan^{-1} \left\{ \frac{r(t)\sin(\psi(t) - \phi(t))}{A_c + r(t)\cos(\psi(t) - \phi(t))} \right\}$$

$$\therefore \theta(t) \approx \phi(t) + \frac{r(t)}{A_c} \sin(\psi(t) - \phi(t))$$

$$\theta(t) \approx 2\pi k_f \int_0^t m(t)dt + \frac{\mathbf{r}(t)}{A_c} \sin(\psi(t) - \phi(t))$$

Assumptionsv(t) discrimination

 Carrier to noise ratio is large compared with unity

v(t) discriminator output = 
$$\frac{1}{2\pi}\theta'(t)$$

$$v(t) = k_f m(t) + n_d(t)$$

where 
$$n_d(t) = \frac{1}{2\pi} \frac{d}{dt} \left\{ \frac{\mathbf{r}(t)}{A_c} \sin(\psi(t) - \phi(t)) \right\}$$

The noise component can be simplified further to

$$n_d(t) \approx \frac{1}{2\pi} \frac{d}{dt} \left\{ \frac{\mathbf{r}(t)}{A_c} \sin(\psi(t)) \right\}$$

however

$$n_s(t) = r(t)\sin(\psi(t))$$

$$n_d(t) \approx \frac{1}{2\pi} \frac{d}{dt} \left\{ \frac{\mathbf{n}_s(t)}{A_c} \right\}$$

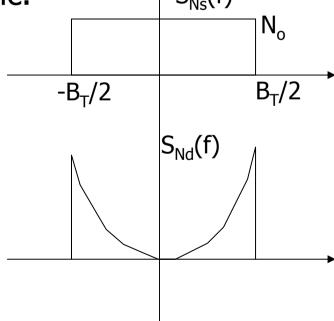
- The power in the output message is k<sub>f</sub><sup>2</sup>P where p is power in transmitted message
- Required to find the power in the noise component
  - Derivative in time  $\rightarrow$  j2 $\pi$ f in frequency

$$\therefore S_{n_d}(f) = \frac{f^2}{A_c^2} S_{n_s}(f)$$

- With the Equivalent IF filter has ideal response.
  - It follows that the narrow-band noise n(t) will have a PSD similar in shape to the transfer function of the IF filter
  - Therefore the quadrature component of the NBN will have the ideal low pass characteristic.  $\uparrow S_{Ns}(f)$
  - The corresponding PSD of n<sub>d</sub> will be

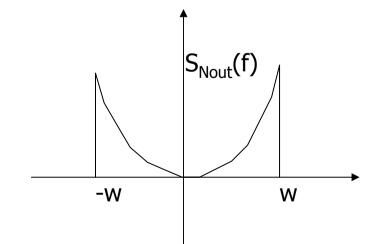
$$\therefore S_{n_d}(f) = \frac{f^2}{A_c^2} S_{n_s}(f)$$

$$S_{n_d}(f) = \begin{cases} \frac{N_o f^2}{A_c^2}, & |f| \leq \frac{B_T}{2} \\ 0, & elsewhere \end{cases}$$



- The discriminator output is followed by a low pass filter with bandwidth equal to the bandwidth of the message W.
  - W>> $B_T/2$
  - Therefore the noise component  $n_d(t)$  will have rejection region.
  - The corresponding PSD of n<sub>d</sub> after LPF will be

$$S_{n_{\text{out}}}(f) = \begin{cases} \frac{N_o f^2}{A_c^2}, & |f| \leq W\\ 0, & elsewhere \end{cases}$$



 The average output noise power is determined by integrating the PSD S<sub>nout</sub> from –W to W

average power of output noise = 
$$\frac{N_o}{A_c^2} \int_{-W}^{W} f^2 df$$
  
=  $\frac{2N_o}{3A^2} W^3$ 

 Note that the average output noise power is inversely proportional to the carrier power A<sub>c</sub><sup>2</sup>/2

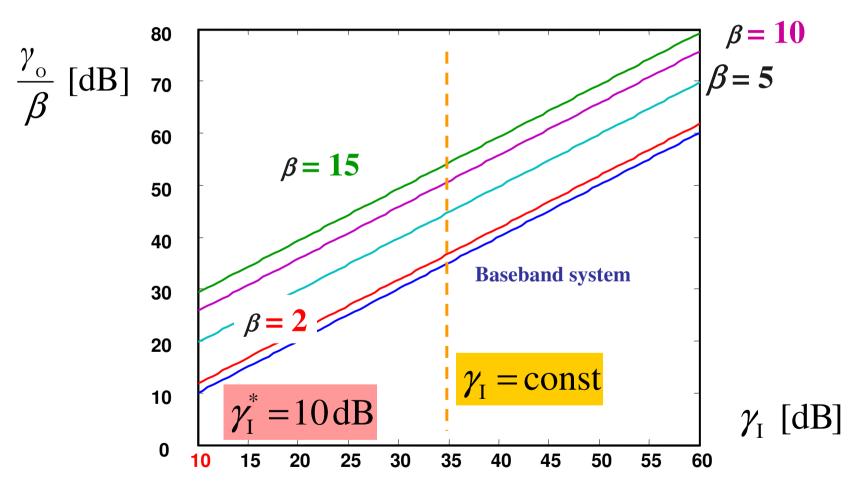
$$SNR_{c,FM} = \frac{A_c^2}{2N_oW}$$

$$SNR_{o,FM} = \frac{3A_c^2k_f^2P}{2N_oW^3}$$

$$Figure \text{ of Merit} = \frac{3k_f^2P}{W^2}$$



### FM Noise Characteristic



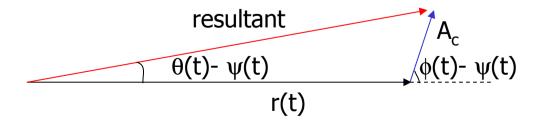
### What if the noise is bigger than the signal

Represent x(t) by means of a phasor diagram

$$x(t) = A_c \cos(2\pi f_c t + \phi(t)) + r(t) \cos(2\pi f_c t + \psi(t))$$

- $r(t) >>> A_c$
- The relevant phase θ(t) can be calculated

$$\theta(t) = \psi(t) + \tan^{-1} \left\{ \frac{A_c \sin(\phi(t) - \psi(t))}{r(t) + A_c \cos(\phi(t) - \psi(t))} \right\}$$



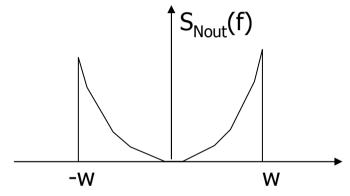
# Noise >>> FM reception

- Envelope of x(t) is not of interest to us
- The output of the ideal discriminator will be proportional to  $\theta'(t)/2\pi$
- $\theta(t) = \psi(t) + \tan^{-1} \left\{ \frac{A_c \sin(\phi(t) \psi(t))}{r(t) + A_c \cos(\phi(t) \psi(t))} \right\}$
- $\therefore \theta(t) \approx \psi(t) + \frac{A_c}{r(t)} \sin(\phi(t) \psi(t))$
- $\theta(t) \approx \psi(t) + \frac{A_c}{r(t)} \sin(2\pi k_f \int_0^t m(t)dt \psi(t))$
- v(t) discriminator output =  $\frac{1}{2\pi}\theta'(t)$
- $\mathbf{v}(\mathbf{t}) = \frac{d\boldsymbol{\psi}(t)}{2\pi dt} + n_d(t)$
- where  $n_d(t) = \frac{1}{2\pi} \frac{d}{dt} \left\{ \frac{A_c}{\mathbf{r}(t)} \sin(2\pi k_f \int_0^t m(t) dt \psi(t)) \right\}$

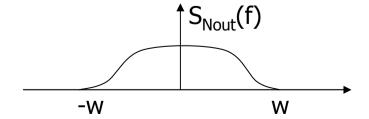
### Assumptions

- Carrier to noise ratio is small compared with unity
- V(t) is complete noise
  - No detection → FM threshold effect
  - SNR>1

The PSD of the noise at the receiver output has a square law dependence on the operating frequency

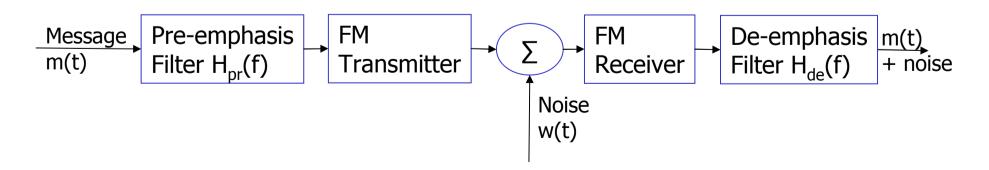


- The PSD of a typical message source (audio, video) has the following general shape
- The PSD falls significantly at high frequencies while the noise PSD increases rapidly with frequency





- One way to improve the noise performance of the system is to slightly reduce the bandwidth of the post detection low pass filter so to reject large amount of noise power → unacceptable
- Another method is to use a pre-emphasis at the transmitter and a de-emphasis at the receiver as shown in the following figure





- In this method, emphasize the high frequency components of the message signal prior to modulation in the transmitter.
- The low frequency and the high frequency portions of the message are equalized to occupy the whole frequency band allocated to it.
- At the discriminator output, inverse process is performed to de-emphasize the high frequency components to restore the original signal power distribution of the message.
  - The high frequency portion of the noise will be reduced thus increasing the output signal to noise ratio of the system



In order to produce an undistorted version of the original message at the receiver output, the preemphasis and de-emphasis must ideally have transfer function that is inverse to each other

$$H_{de}(f) = \frac{1}{H_{pe}(f)}, \quad -W \le f \le W$$

- This makes the message power independent of this operation.
- For the noise component

after the discriminator 
$$S_{n_d}(f) = \begin{cases} \frac{N_o f^2}{A_c^2}, & |f| \leq \frac{B_T}{2} \\ 0, & elsewhere \end{cases}$$

- Thus the modified PSD of the noise at the deemphasis output is equal to  $\left| H_{de}(f) \right|^2 S_{n_s}(f)$
- Since the final LPF has BW= W<<B<sub>→</sub>/2

Av. o/p noise power

with de-emphasis = 
$$\frac{N_o}{A^2} \int_{0}^{w} f^2 |H_{de}(f)|^2 df$$

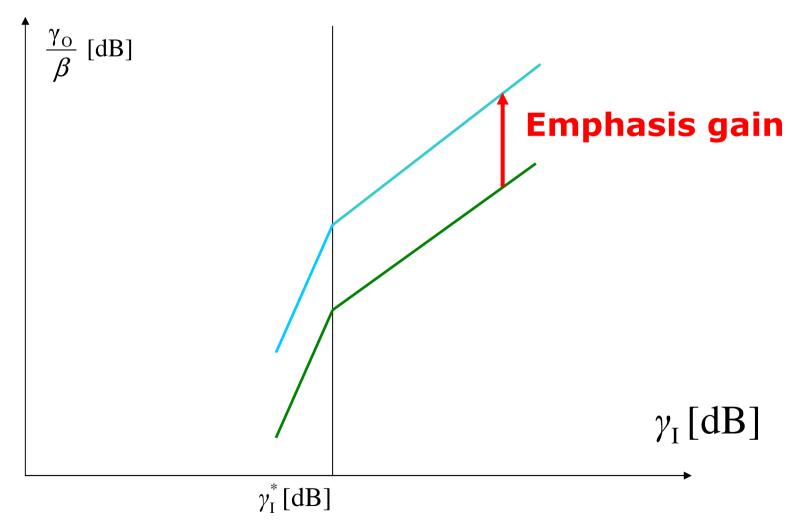
with de-emphasis =  $\frac{N_o}{A_c^2} \int_{-w}^{w} f^2 |H_{de}(f)|^2 df$ The improvement in o/p SNR due to de-emphasis is

$$D = \frac{\text{Average o/p noise power without emphasis}}{\text{Average o/p noise power with emphasis}}$$

$$D = \frac{2W^3}{3\int_{-w}^{w} f^2 \left| H_{de}(f) \right|^2 df}$$

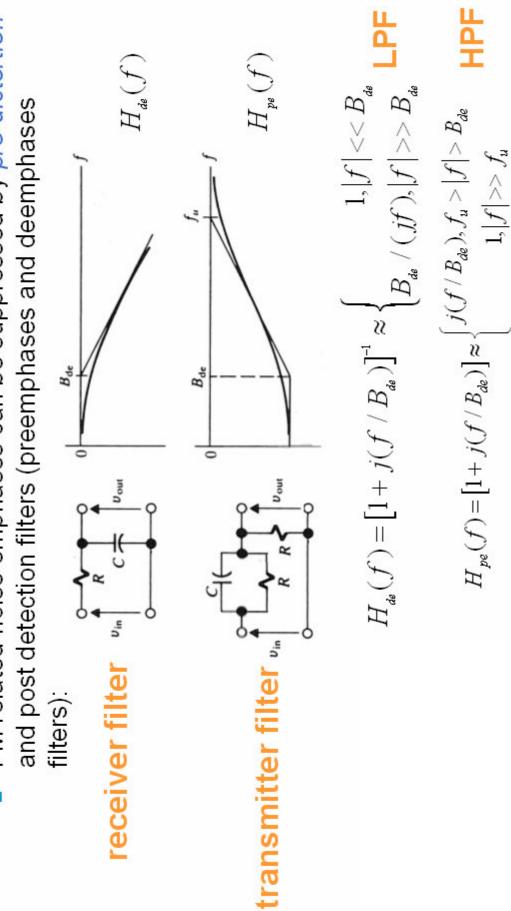


### Noise characteristics



# FM preemphases and deemphases filters

FM related noise emphases can be suppressed by pre-distortion and post detection filters (preemphases and deemphases



### FM post-detection S/N with deemphases

 Deemphases filter (that is a lowpass filter connected after detector) can suppress noise further. FM post-detection noise PSD and total noise power without deemphases:

$$G_{FM}(f) = \frac{N_o f^2}{2S_R} \Pi \left( \frac{f}{B_T} \right) \quad N_D = \int_{-W}^{W} G_{FM}(f) df = \frac{N_o W^3}{3S_R}$$

With deemphases filter (for simplification assume W/B<sub>de</sub>>>1):

$$\begin{split} N_{_{D}} &= \int_{_{-W}}^{W} G_{_{FM}}(f) \big| H_{_{de}}(f) \big|^2 \, df = \frac{\mathsf{N_o} B_{_{de}}^3}{S_{_R}} \Bigg| \frac{W}{B_{_{de}}} - \arctan \frac{W}{B_{_{de}}} \Bigg| \approx \mathsf{N_o} B_{_{de}}^2 W \, / \, S_{_R} \end{split}$$
 where 
$$\big| H_{_{de}}(f) \big|^2 = \frac{1}{1 + (f \, / B_{_{de}})^2} \\ S_{_{D}} / N_{_{D}} &= \frac{f_{_{\Delta}}^2 S_{_{x}}}{\mathsf{N_o} B_{_{de}}^2 W \, / \, S_{_{R}}} = \frac{S_{_{x}} S_{_{R}}}{\mathsf{N_o} W} \bigg( \frac{f_{_{\Delta}}}{B_{_{de}}} \bigg)^2 = S_{_{x}} \gamma \bigg( \frac{f_{_{\Delta}}}{B_{_{de}}} \bigg)^2 \end{split}$$

### Example

FM radio

$$f_{A} = 75 \text{ kHz}, W = 15 \text{ kHz}, D = 5, S_{x} = 1/2, B_{de} = 2.1 \text{ kHz}$$

Without deemphases

$$S_{_D}/N_{_D} = 3D^2S_{_X}\gamma$$

$$= (3 \times 5^2 \times \frac{1}{2})\gamma = 38\gamma \qquad \gamma = \frac{S_{_R}}{N_{_D}W}$$
hases

With deemphases

$$S_{\rm D}/N_{\rm D} = (f_{\rm A}/B_{\rm de})^2 S_{\rm x} \gamma \approx 640 \gamma$$

- Therefore if DSB or SSB system could be exchanged to FM system 640 fold transmission power savings could be achieved.
- Note, however that the required transmission bandwidth is now about 220 kHz /15 kHz = 15 times larger!