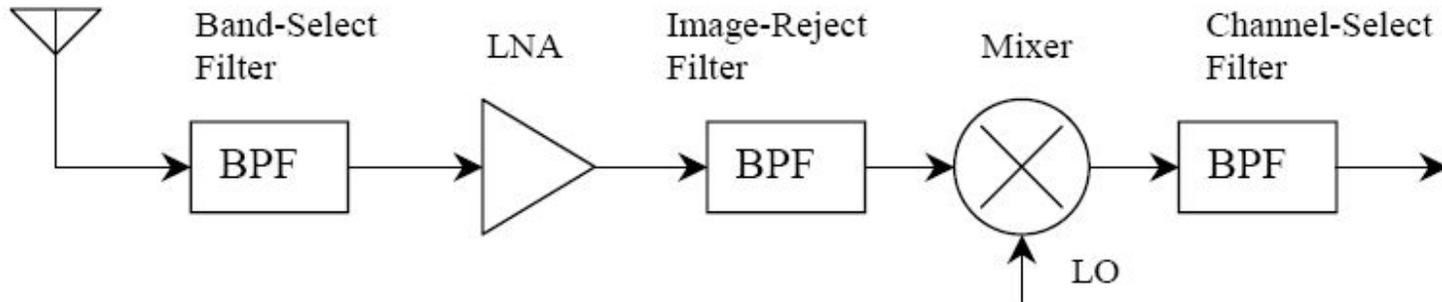


Up-/Down- conversion in frequency

Why?

- Amplification 120+ dB without parasitic feedback
- Tunable receiver without too many tuned filters
- Easier narrowband filtering

Superheterodyne receiver



IF

$$\cos(\omega_1 t)\cos(\omega_2 t) = 1/2 \cos((\omega_1 - \omega_2)t) + 1/2 \cos((\omega_1 + \omega_2)t)$$

$\omega_1 - \omega_2$ - IF or beat frequency

Intermediate frequency is usually lower than RF (easier to amplify/filter/process).

Examples (receivers):

FM radio: 10.7 MHz

AM radio: 465 kHz

analog TV: 30 MHz, 45 MHz

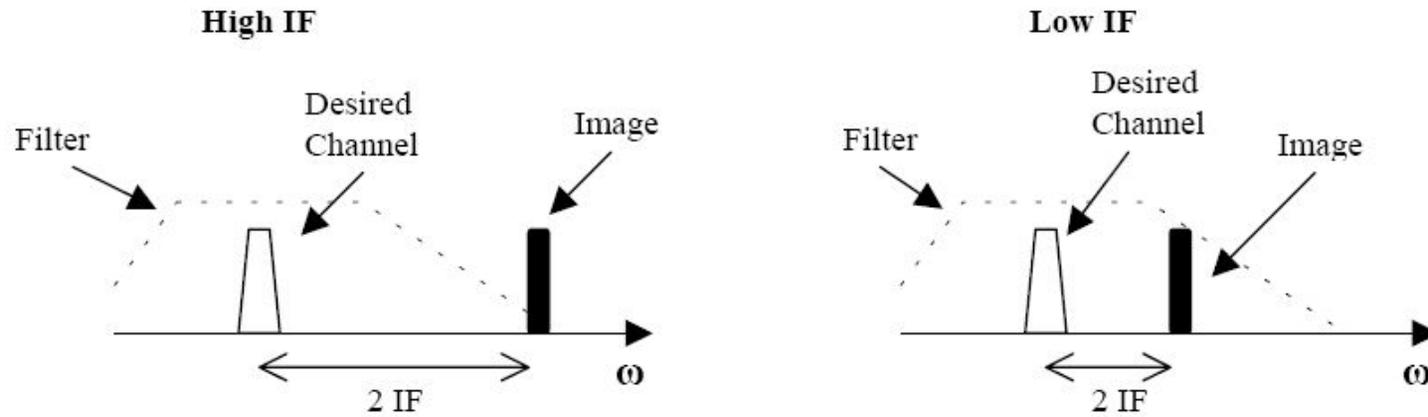
Satellite equipment: 70 MHz second, 950-1450 first IF (L-band) *double conversion receiver!!*

Terrestrial MW link: 250 MHz, 70 MHz

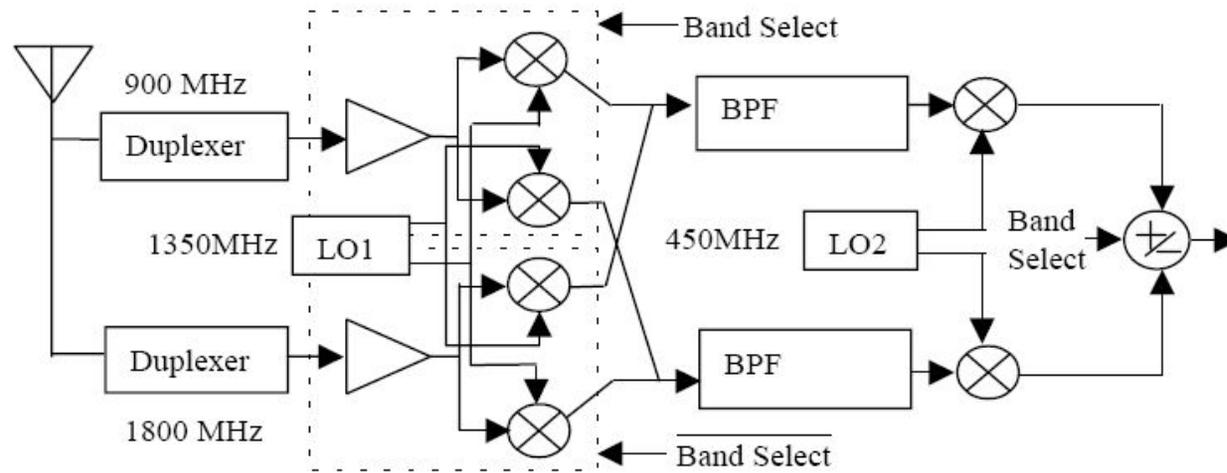
Radar: 30 MHz, 70 MHz

Image band

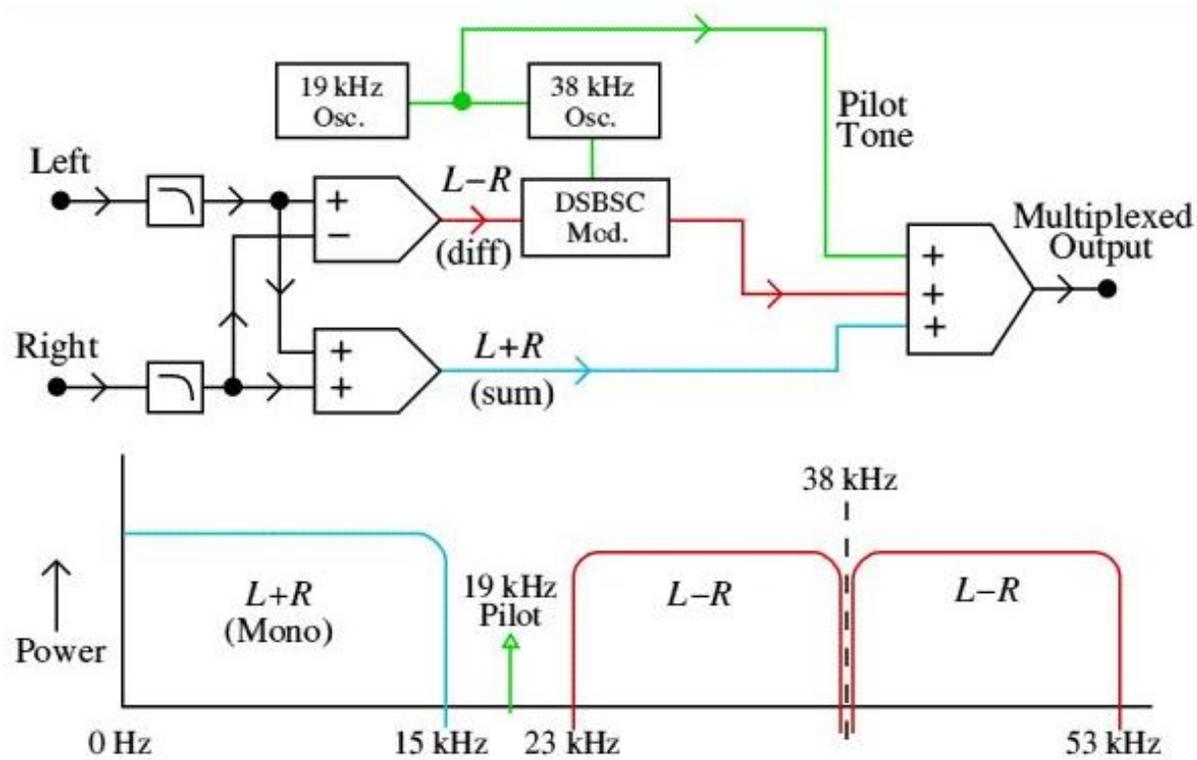
$$f_{RF} = f_{LO} \pm f_{IF} \text{ (which one?)}$$



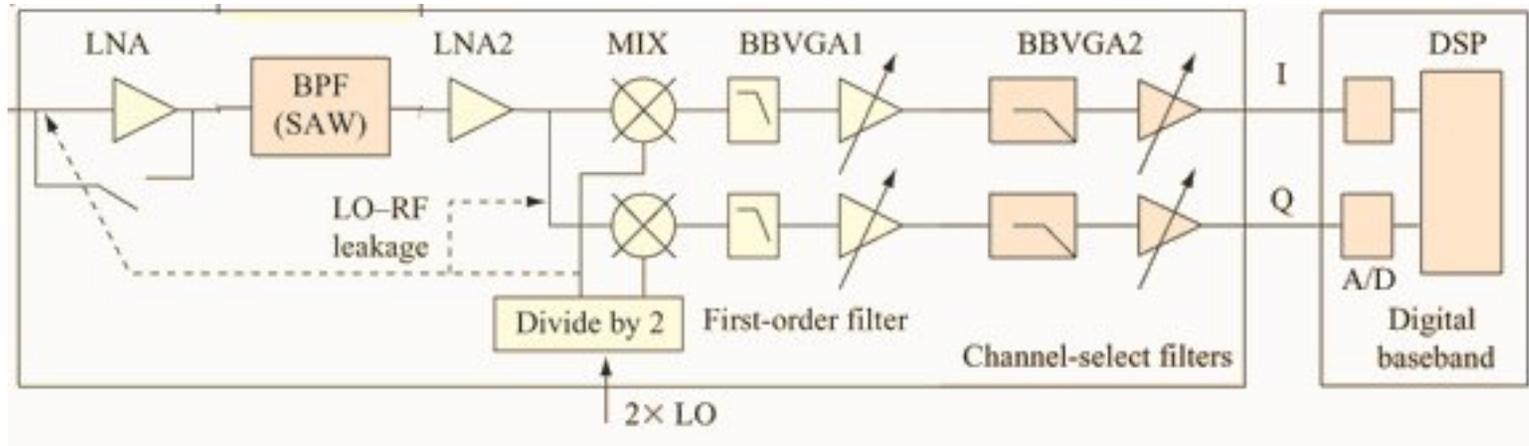
Dualband receiver example



Stereo FM example



Direct conversion receiver (homodyne)



- LO leakage \longrightarrow DC bias \longrightarrow saturation of BB amp
- Linearity (little gain control before mix)
- 2xLO - to produce phase-shifted “sin” & “cos” LO
- Synchronization of LO with carrier (radar \longrightarrow easy, other \longrightarrow PLL, digital cancel, ...)
- no image band!

Transmitter - modulation

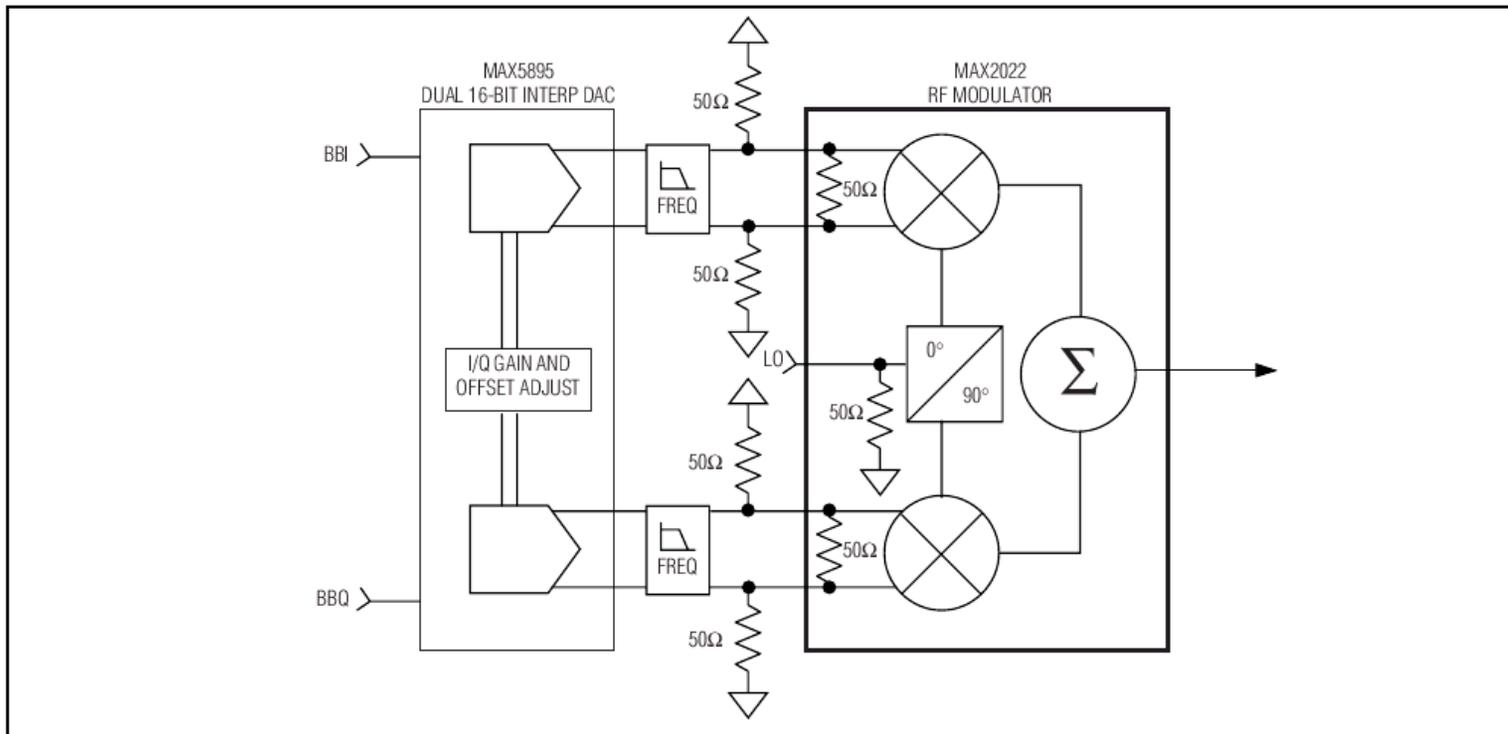
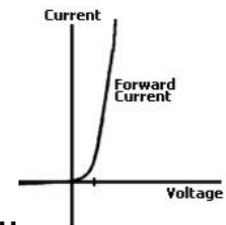
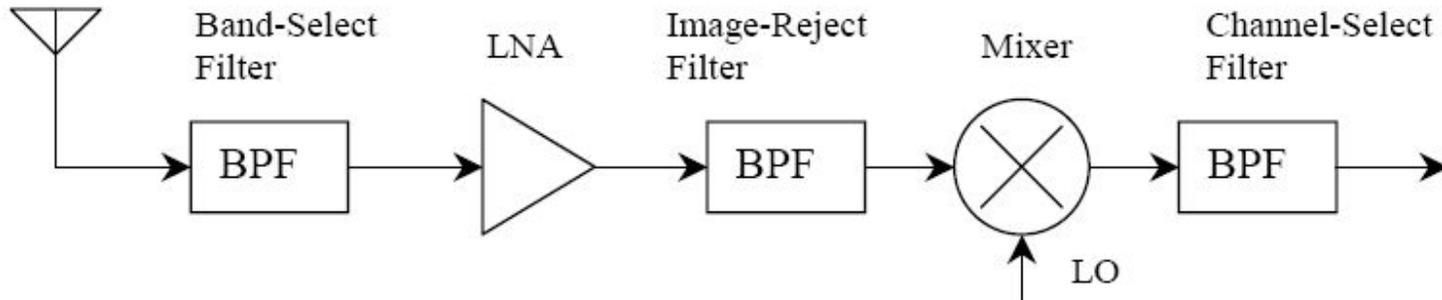


Figure 1. MAX5895 DAC Interfaced with MAX2022

Mixer

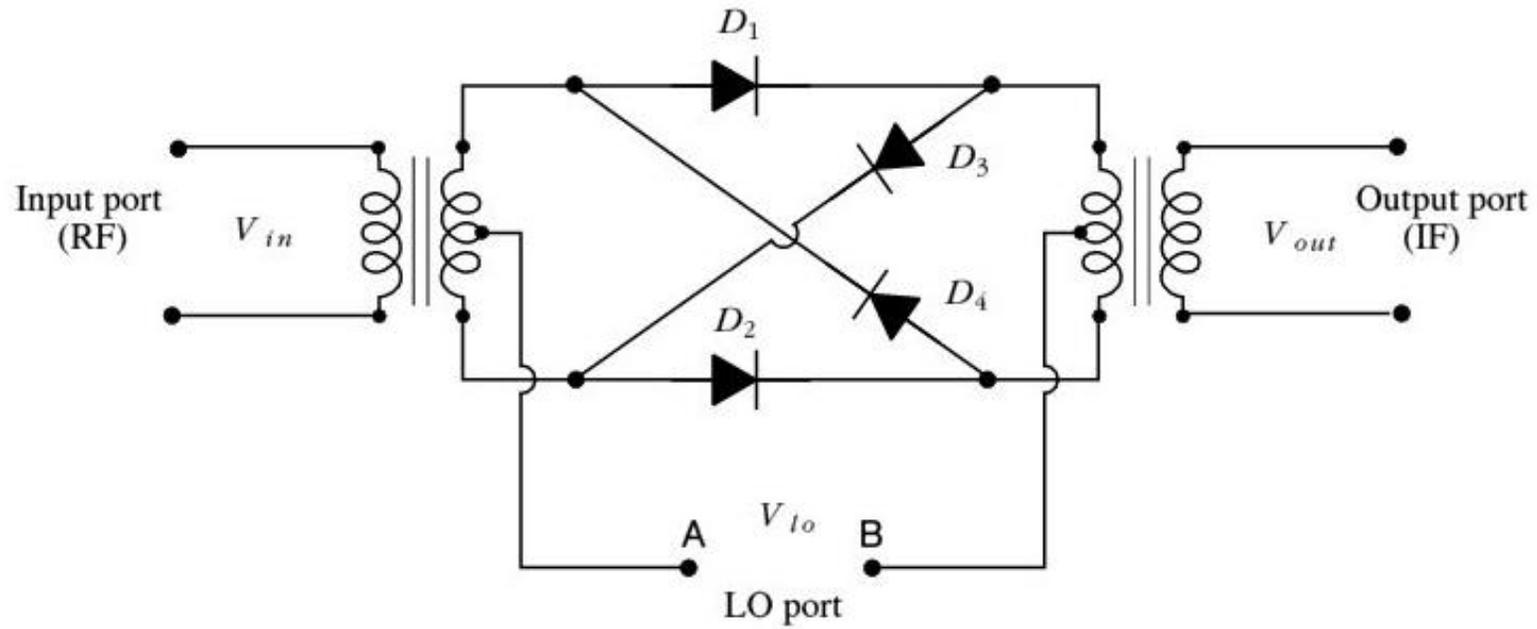


How to multiply two signals? \rightarrow use any element with x^2 type nonlinearity...

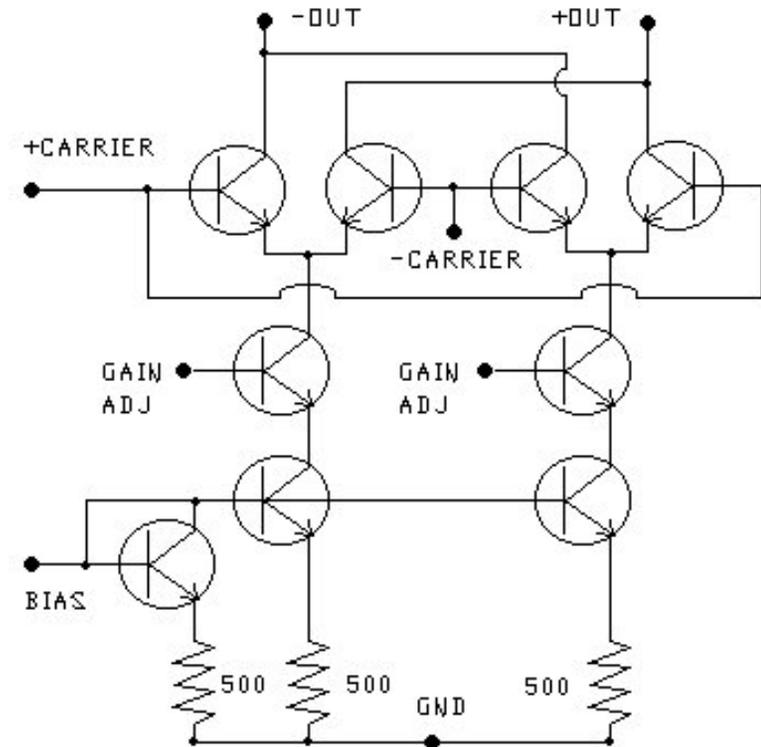
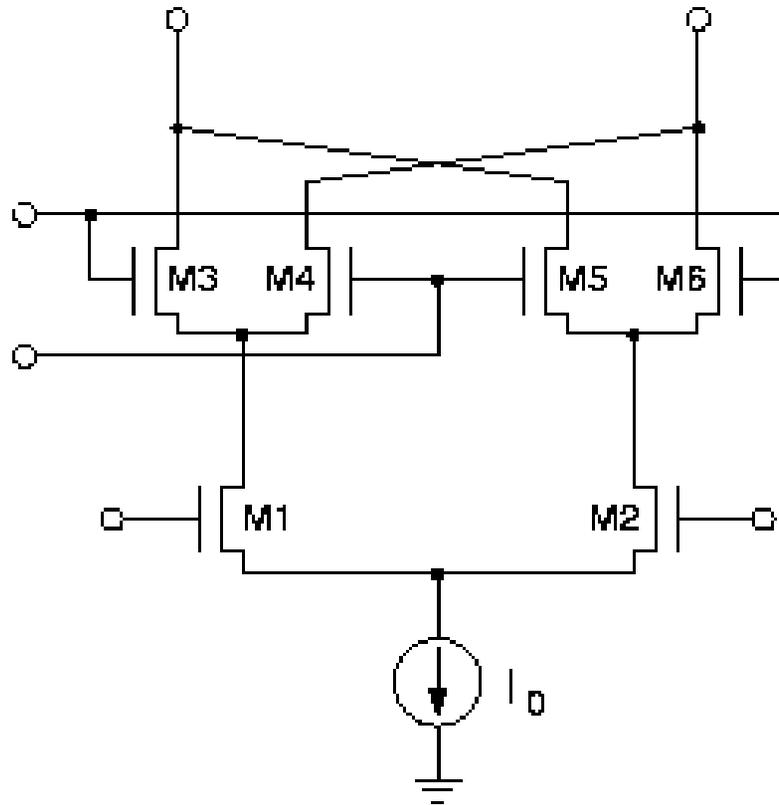
$$(\cos(\omega_1 t) + \cos(\omega_2 t))^2 = \cos(\omega_1 t)^2 + \cos(\omega_2 t)^2 + 2\cos(\omega_1 t)\cos(\omega_2 t)$$

... and filter out $\cos^2(\omega t) = 1/2 + 1/2\cos(2\omega t)$ components

Diode mixer



Balanced mixer circuits



Matched filter

Task: detect a signal $s(t)$ in noise. *Is There Anybody Out There?(and when?)*

Idea: use LTI filter that maximizes pSNR=(signal instantaneous power)/(noise power) ratio at some instant (assume e.g. at $t = 0$)

Derivation: assume noise is white with PSD N_0 ; after a filter $H(j\omega)$ noise power is

$$P_N = N_0/4\pi \int_{-\infty}^{+\infty} |H(j\omega)|^2 d\omega$$

After the $H(k\omega)$ filter the signal spectrum becomes

$$Y(j\omega) = S(j\omega)H(j\omega)$$

thus the signal at $t = 0$ (by inv. Fourier transf.)

$$y(0) = 1/2\pi \int_{-\infty}^{+\infty} S(j\omega)H(j\omega)e^{j\omega 0} d\omega$$

We want to maximize $|y(0)|^2/P_n$

Now we put our spectra into the Schwarz inequality:

$$\left| \int S(j\omega)H(j\omega)d\omega \right|^2 \leq \int |S(j\omega)|^2 d\omega \int |H(j\omega)|^2 d\omega$$

so

$$pS/N = \frac{|y(0)|^2}{P_n} = \frac{|\int S(j\omega)H(j\omega)d\omega|^2}{N_0/2 \int |H(j\omega)|^2 d\omega} \leq 2/N_0 \int |S(j\omega)|^2 d\omega$$

If we guess find $H(j\omega)$ such that the above \leq becomes $=$, nobody will find anything better..

For complex numbers $|X|^2 = X \cdot X^*$ so

$$\frac{\int S(j\omega)H(j\omega)d\omega \int S^*(j\omega)H^*(j\omega)d\omega}{\int H(j\omega)H^*(j\omega)d\omega} \leq \int S(j\omega)S^*(j\omega)d\omega$$

if we put $H(j\omega) = S^*(j\omega)$ we got it! \longrightarrow (please recall from circuit theory how we compensate reactive power to maximize power drained from the source)

Matched filter - conclusions

$$H(j\omega) = S^*(j\omega) \longrightarrow H(t) = S(-t)$$

We may modify it a little bit:

- delay in time by length t_s of $s(t)$ \longrightarrow so that $h_d(t) = h(t - t_s)$ is causal
- scale it by any constant (equality holds)

As now $|y(0)|^2 = 1/2\pi \int_{-\infty}^{+\infty} |S(j\omega)|^2 d\omega = E =$ signal energy of $s(t)$ (Parseval),
pSNR at the output of an ideal matched filter equals

$$pSNR = \frac{2E}{N_0}$$

... but other signal parameters (BW, t_s ...) may be used for whatever we want. (technical reasons, sidelobes.....)

Matched filter - variations

- ...with non-white noise:

$$H(j\omega) = \frac{S^*(j\omega)}{|N(j\omega)|^2} = \frac{1}{N(j\omega)} \cdot \frac{S^*(j\omega)}{N^*(j\omega)}$$

(whitening + matched to a spoiled $s(t)$)

- Mismatched filter - a matched filter modified a bit, e.g. to reduce sidelobes:
 - modification by windowing
 - modification by optimization techniques
- **mis**matched filter is not optimal for $pSNR$, but by losing a little bit of $pSNR$ we may make it optimal in some other sense