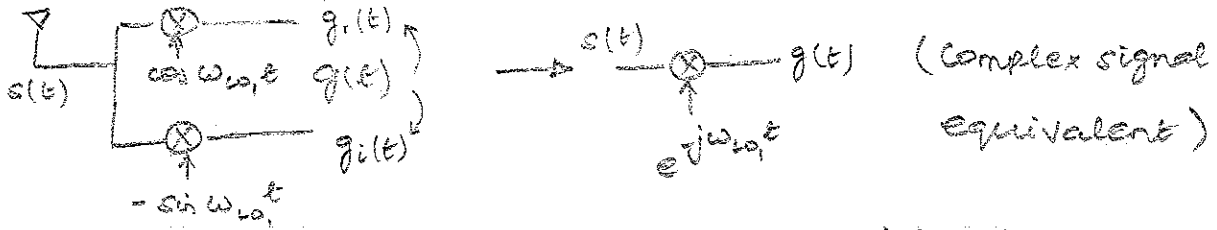


Image-reject receivers:

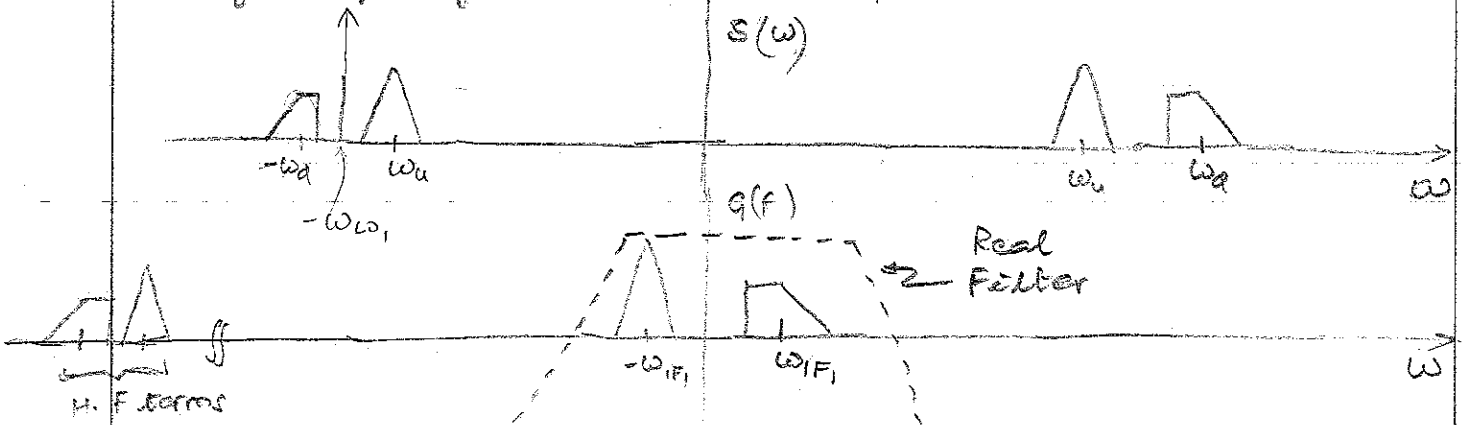
Recall the complex-signal notation in receivers:



$$s(t) = \underbrace{\text{Re} [\tilde{s}_d(t) e^{j\omega_c t}]}_{\text{Desired Signal}} + \underbrace{\text{Re} [\tilde{s}_u(t) e^{j\omega_c t}]}_{\text{Undesired image}}$$

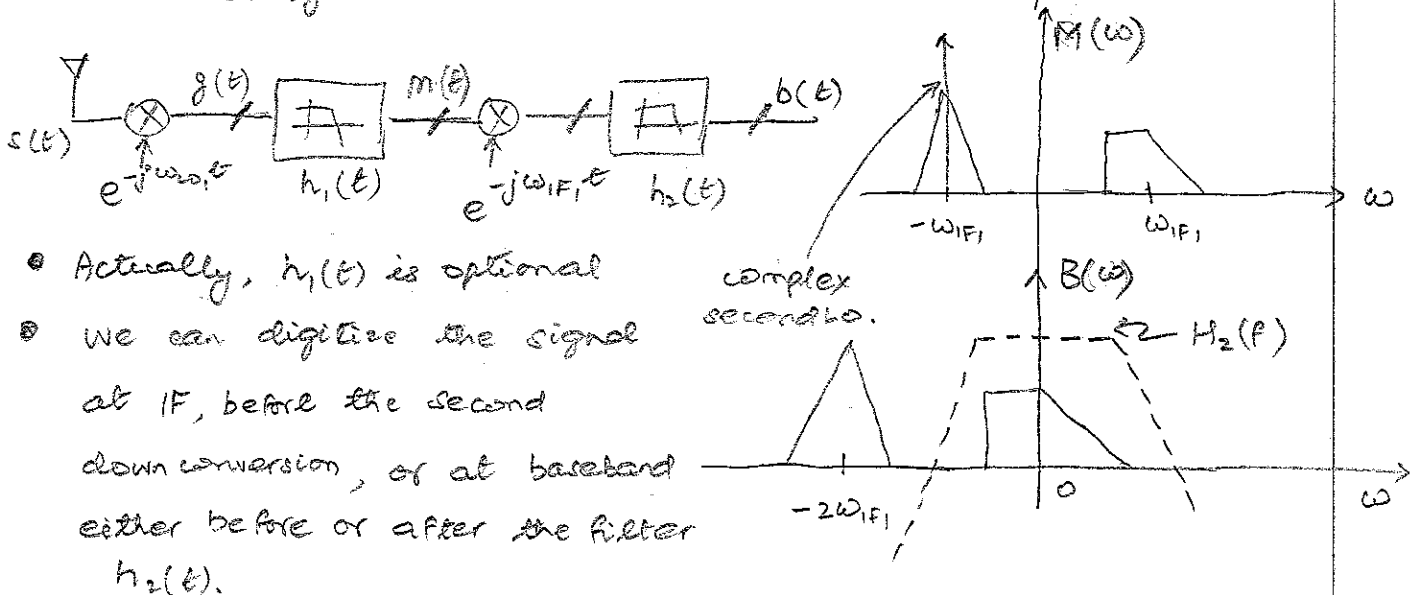
IF frequency $\omega_{IF} = \omega_d - \omega_{c0}$

Image frequency $\omega_u = \omega_d - 2\omega_{IF}$



How to isolate desired channel?

(a) Weaver Receiver: Use a second complex downconversion followed by a filter with a real-valued impulse response.



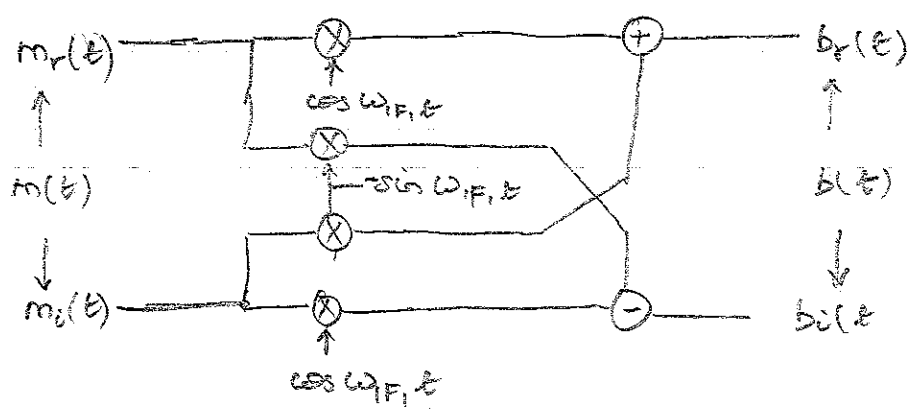
- Actually, $h_1(t)$ is optional
- We can digitize the signal at IF, before the second down conversion, or at baseband either before or after the filter $h_2(t)$.

• But $m(t)$ is a complex-valued signal. How do we do complex downconversion?

→ Recall that $m(t) = m_r(t) + jm_i(t)$ where $m_r(t)$ and $m_i(t)$ are real-valued.

$$\begin{aligned} \therefore b(t) &= m(t) e^{-j\omega_{IF}t} \\ &= [m_r(t) + jm_i(t)] [\cos\omega_{IF}t - j\sin\omega_{IF}t] \\ &= [m_r(t)\cos\omega_{IF}t + m_i(t)\sin\omega_{IF}t] \leftarrow b_r(t) \\ &\quad + j[-m_r(t)\sin\omega_{IF}t + m_i(t)\cos\omega_{IF}t] \leftarrow b_i(t) \end{aligned}$$

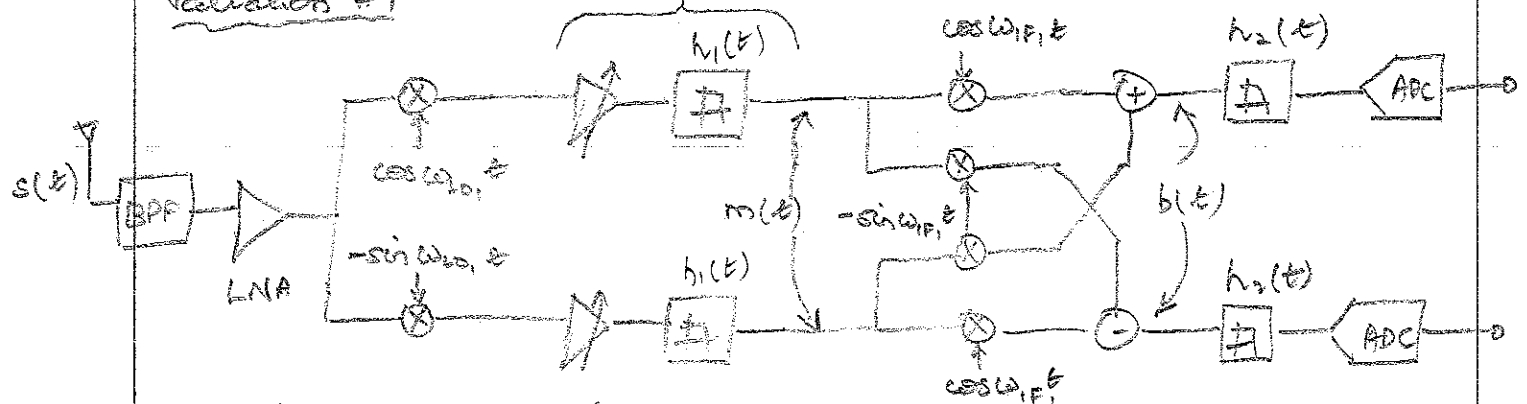
⇒ Implement complex mixing with 4 real multiplications and 2 real additions:



Complete Weaver Receiver:

Variation #1

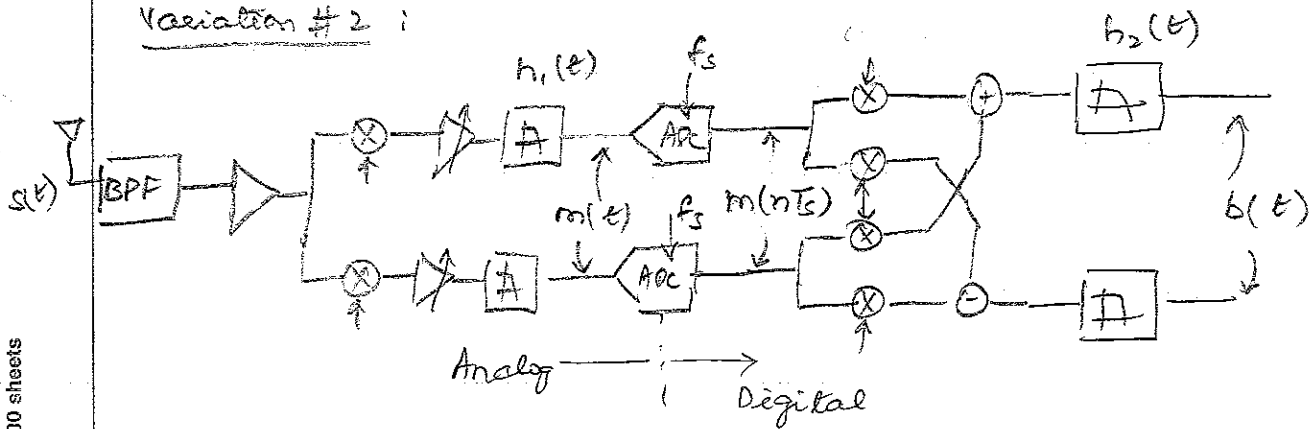
Optional!



- Fully analog Weaver ← $h_2(t)$ performs channel selection
- Can use high-IF before ADC

Example "A 1.9GHz Wideband Double Conversion CMOS Receiver..."
 J. C. Rudell et al. JSSC Dec 1997, pp 2071-2088

Variation # 2 :



- $h_1(t)$ required to perform anti-aliasing for ADC.
- $h_2(t) \rightarrow$ digital channel-select filter.
- IF is usually low, otherwise ADC requires very high sampling rate
(Advanced topics here: bandpass ADC's, sub-sampling)
- If sampling rate of ADC is chosen such that $f_{IF} = \frac{f_s}{4}$, then the 4 mixers in the second downconversion become AND gates: since

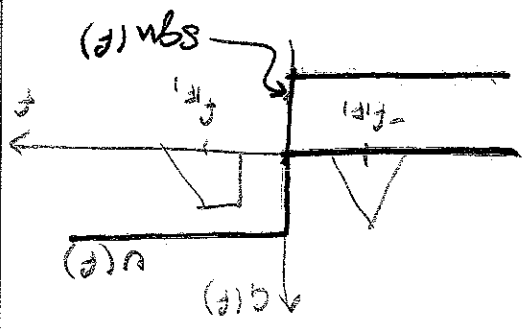
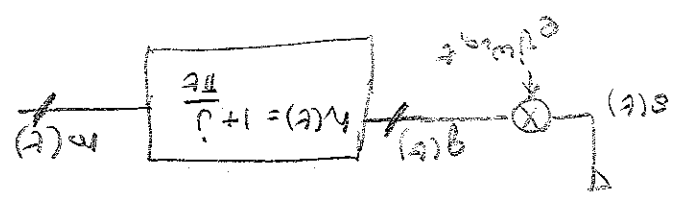
$$\cos \omega_{IF} t \rightarrow \cos \omega_{IF} nT_s = \cos \frac{2\pi f_{IF} n}{f_s} = \cos \frac{n\pi}{2} = \left\{ \dots, +1, 0, 1, 0, \dots \right\}$$

$$\text{and } \sin \omega_{IF} t \rightarrow \sin \omega_{IF} nT_s = \sin \frac{2\pi f_{IF} n}{f_s} = \sin \frac{n\pi}{2} = \left\{ \dots, 0, 1, 0, -1, \dots \right\}$$

Example: S. Dow et al., "A dual-band direct-conversion/VLIF transceiver for GSM/GPRS/PCS", IEEE ISSCC, Feb 2002, pp 230-231

\rightarrow This is the topology of choice for modern GSM cellphones

OPTIONAL: Hilbert Transforms & the Generalized Hersey Receiver!



- Suppose we have a filter $h_1(t)$ whose Fourier transform is the unit-step function $U(f)$, i.e., $h_1(t) \xrightarrow{F.T.} H_1(f) = U(f)$
- We can create the desired signal by filtering the complex signal $g(f)$ with the complex filter $H_1(f) = U(f)$



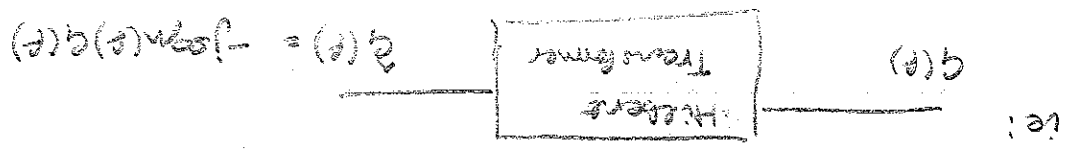
$$M(f) = U(f) \cdot g(f) = \frac{1 + \operatorname{sgn}(f)}{2} g(f)$$

$$= \frac{1}{2} g(f) + \frac{j}{2} \{ -\operatorname{sgn}(f) \} g(f)$$

- Now, recall that $-\operatorname{sgn}(f)g(f)$ is just the F.T. of the

Hilbert transform of $g(t)$.

(Ref: Haykin, "Communication Systems", 2nd ed Wiley & Sons)



Input response = $\frac{1}{j} \leftarrow$ real-valued.

$$\therefore M(f) = \frac{1}{2} [g(f) + j\hat{g}(f)]$$

$$\text{But } q(f) = g_r(f) + jg_i(f)$$

$$\Rightarrow M(f) = \frac{1}{2} [g_r(f) + jg_i(f) + j\hat{g}_r(f) + j\hat{g}_i(f)]$$

$$= \frac{1}{2} [g_r(f) - \hat{g}_i(f) + j\{g_i(f) + \hat{g}_r(f) + g_i(f)\}]$$

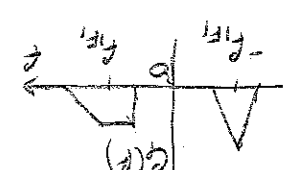
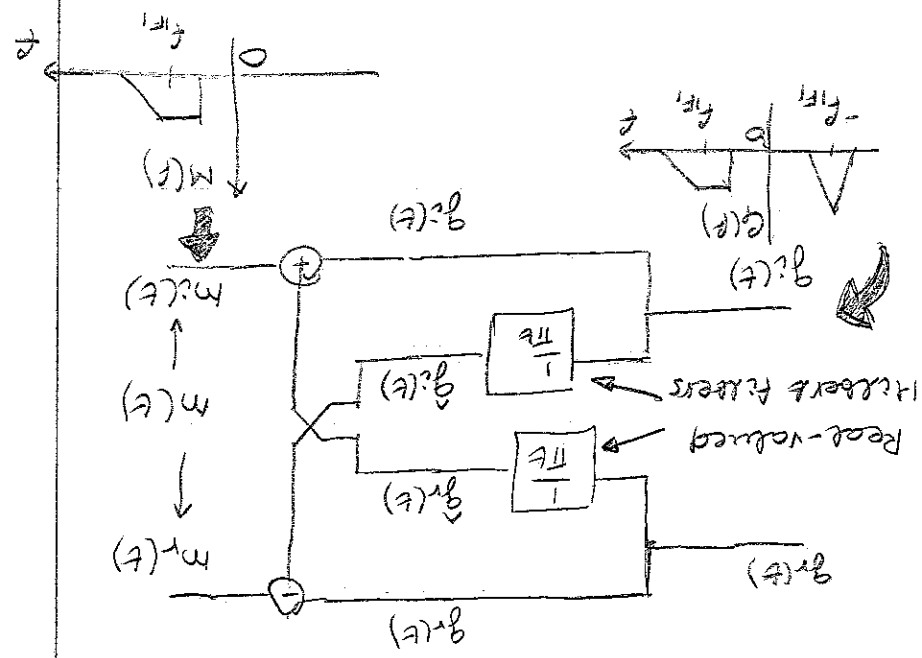
$M_r(f)$ $M_i(f)$

The caret symbol denotes Hilbert transform.

Note that $G_r(f)$, $G_i(f)$, $G_r(f)$ and $G_i(f)$ are each Fourier

transforms of real-valued signals, and are therefore, physically

impementable.

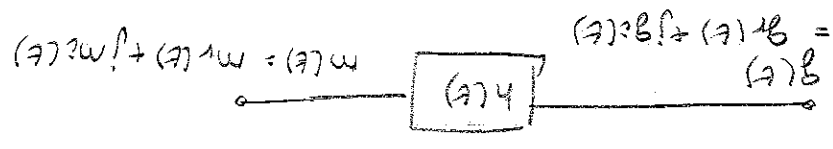


We can express this in a compact form using the

notion of a complex-valued filter. Specifically, define

$$h(f) = h_r(f) + j h_i(f).$$

Here $h_r(f) = 1$ and $h_i(f) = \frac{1}{\pi}$ (impulse-response of Herbert filter)



The input-output convolution follows the rules of complex addition, multiplication & superposition:

$$m(t) = g(t) * h(t) = (g_r + j g_i) * (h_r + j h_i) = (g_r * h_r - g_i * h_i) + j (g_r * h_i + g_i * h_r)$$

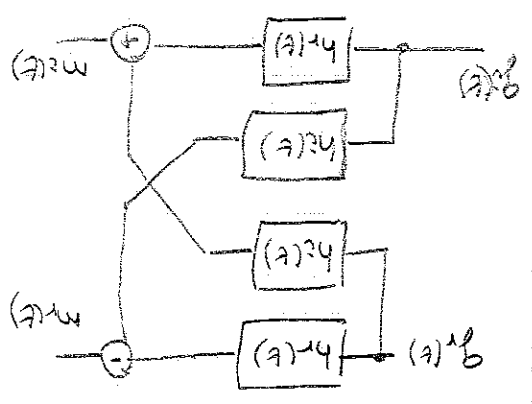
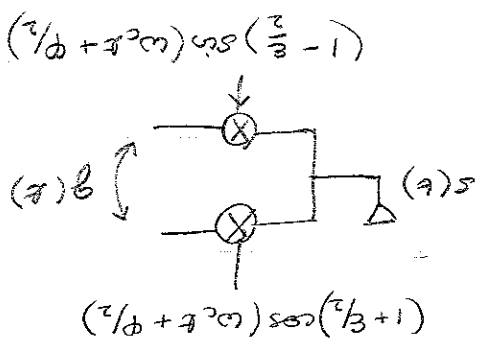


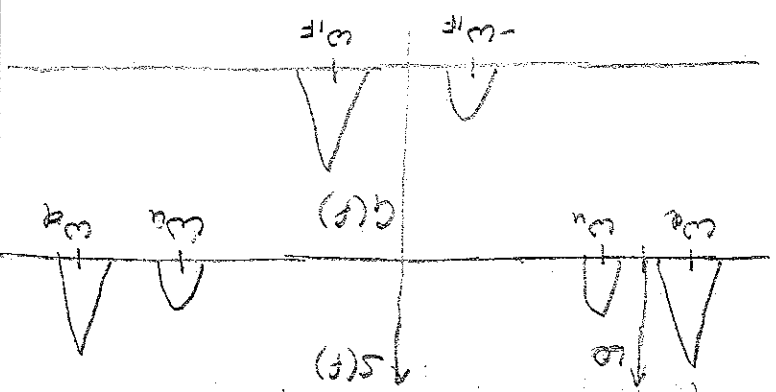
Image rejection ratio (IRR)

Until now, we have assumed perfect quadrature mixing
 - equal gain and phase-shifts in each I & Q paths
 - I and Q LO's are exactly 90° apart.
 Under these assumptions, the image signal is cancelled perfectly. In practice, the image is attenuated imperfectly due to gain & phase mismatches:

Consider a simple model:



With perfect quadrature



With I/Q mismatch!

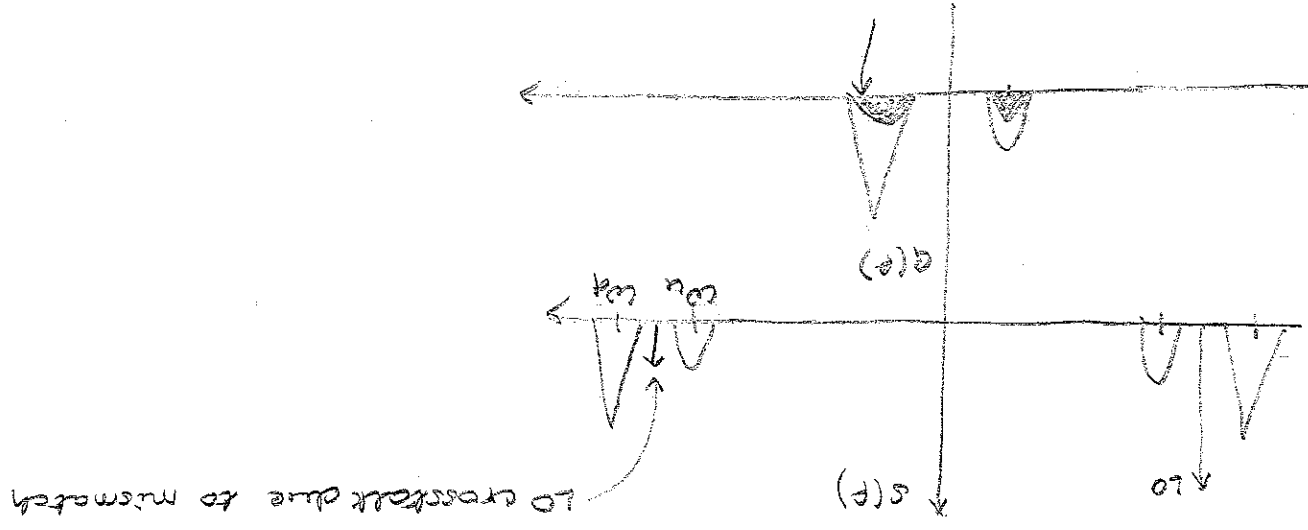


Image imperfectly attenuated
 → degrades SNR.

• Instead of down conversion as $g(t) = s(t)e^{j\omega_c t}$,
 we get down conversion as $g(t) = s(t)[A_1 e^{j\omega_c t} + A_2 e^{-j\omega_c t}]$

Where A_1 & A_2 are complex constants dependent on the gain mismatch ϵ & phase mismatch ϕ .

With simple algebra,
 $A_1 = \cos \phi/2 + j \epsilon/2 \sin \phi/2$
 $A_2 = \frac{\epsilon}{2} \cos \phi/2 + j \sin \phi/2$

Image rejection ratio IRR = $\frac{\text{Signal/Image @ input}}{\text{Signal/Image @ output}}$

$$= \frac{1 + 2(1 + \epsilon) \cos \phi + (1 + \epsilon)^2}{\epsilon^2 + \phi^2} \approx \frac{1 - 2(1 - \epsilon) \cos \phi + (1 + \epsilon)^2}{4}$$

$(\epsilon = \frac{\Delta G}{G} = \text{relative gain mismatch} \ \& \ \phi \text{ is phase mismatch in radians})$

For $\epsilon = 1\%$ and $\phi = 1^\circ$ → IRR = 40dB

• These numbers are typical in IC implementations.

• With digital calibration, IRR can be improved to 60-80dB

OPTIONAL: Derivation of $A_1, \Delta A_2$:

$$LD = (1 + \frac{\epsilon}{2}) \cos(\omega t + \phi) - j(1 - \frac{\epsilon}{2}) \sin(\omega t - \phi) - \frac{\epsilon}{2}$$

$$(1 + \frac{\epsilon}{2}) (\cos \omega t \cos \phi - \sin \omega t \sin \phi) + j(1 - \frac{\epsilon}{2}) (\sin \omega t \cos \phi - \cos \omega t \sin \phi)$$

$$= \cos \phi \left[(1 + \frac{\epsilon}{2}) \cos \omega t - j(1 - \frac{\epsilon}{2}) \sin \omega t \right]$$

$$- \sin \phi \left[(1 + \frac{\epsilon}{2}) \sin \omega t + j(1 - \frac{\epsilon}{2}) \cos \omega t \right]$$

$$= \cos \phi \left[(\cos - j \sin) + \frac{\epsilon}{2} (\cos + j \sin) \right] - \sin \phi \left[(\sin + j \cos) + \frac{\epsilon}{2} (\sin - j \cos) \right]$$

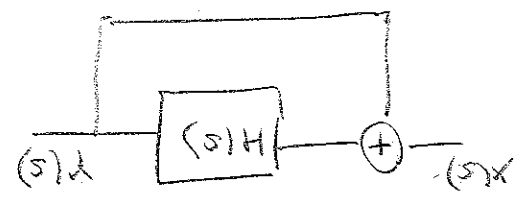
$$= \cos \phi \left(e^{-j \omega t} + \frac{\epsilon}{2} e^{j \omega t} \right) - \sin \phi \left(j e^{j \omega t} - j \frac{\epsilon}{2} e^{-j \omega t} \right)$$

$$= \underbrace{\left(\cos \phi + j \frac{\epsilon}{2} \sin \phi \right)}_{A_1} e^{-j \omega t} + \underbrace{\left(\frac{\epsilon}{2} \cos \phi + j \sin \phi \right)}_{A_2} e^{j \omega t}$$

Voltage-Controlled Oscillators (VCO's)

Autonomous circuit, always used in RF transmitters for up/down conversion: Used in phase-locked loop frequency synthesizer for RF channel selection

Feedback model:



$$Y(s) = \frac{H(s)}{1-H(s)}$$

⇒ If at $s = j\omega$, $H(j\omega) = 1$

then $Y(s) = \infty$ ⇒ possible oscillations with constant amplitude.

Barkhausen criteria for oscillations:

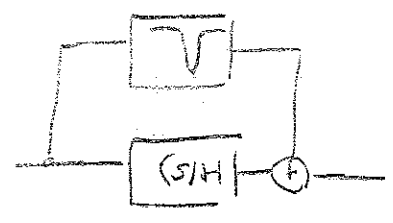
Two necessary conditions are required for steady-state oscillation. At $s = s_0$, $H(s_0) = 1$

⇒ $|H(j\omega_0)| = 1$

$\angle H(j\omega_0) = 360^\circ$

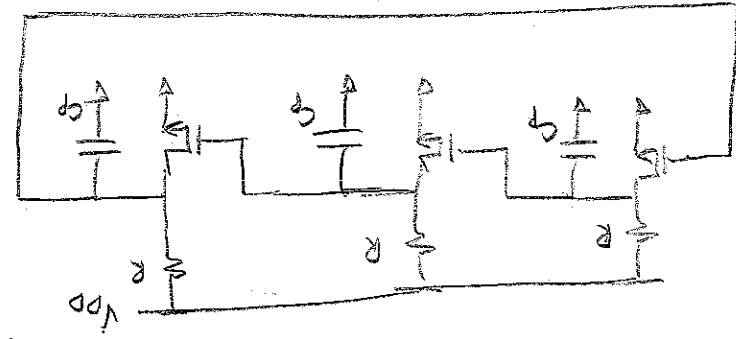
In many RF oscillators, an LC tank is used outside the loop to stabilize frequency:

Note: If we have DC negative feedback, we would have a 180° phase shift around the loop at DC. $H(s)$ would need to contribute only 180° of excess phase with $|H(j\omega)| > 1$ for oscillations



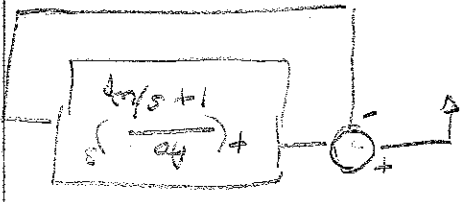
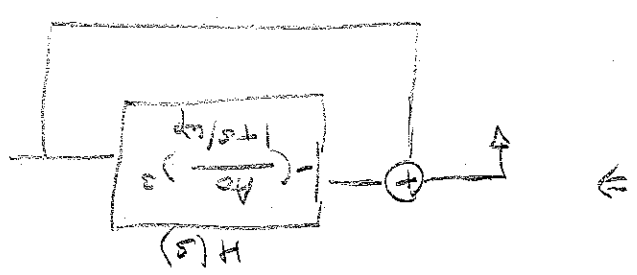
Barkhausen criterion not sufficient for oscillations ⇒ even if met, oscillations may not occur (Ex: CMOS inverters in a loop)

Example: Consider 3-stage C-s ring oscillator:

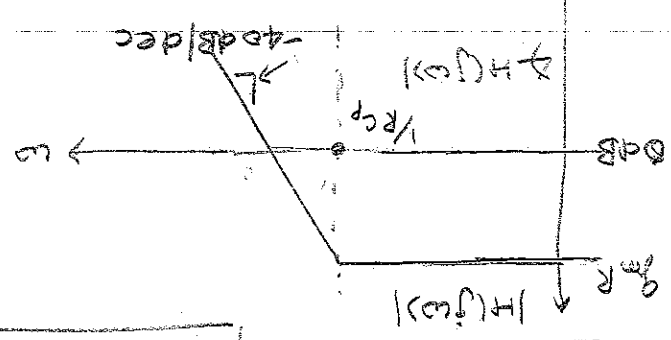


At DC, inverting $\Rightarrow 180^\circ$ phase shift around loop

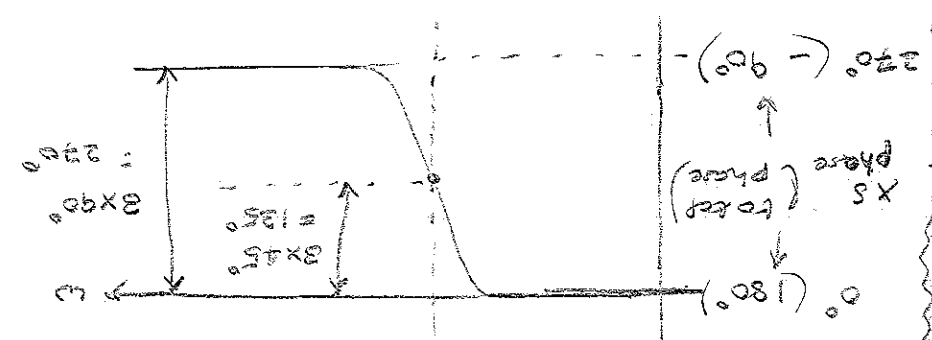
Small signal response of each stage: $A_0(s) = \frac{A_0}{1 + s/pc}$



$$H(s) = \frac{A_0}{(1 + s/pc)^3}$$



For oscillation, Foster phase shift around loop = 360° at ω_x



i.e., $\pi - 3\pi \omega_x^{-1} = 2\pi$ or $\pi \omega_x^{-1} = \frac{2\pi}{3} \Rightarrow \omega_x = \sqrt{3} pc$

At $\omega = \sqrt{3} pc$, $|H(j\omega)| = \frac{A_0}{3} \left[1 + \frac{\omega^2}{pc^2} \right]^{3/2} = \frac{A_0}{8}$

For oscillation, we require $|H(j\omega)| = 1 \Rightarrow A_0 = 8$