

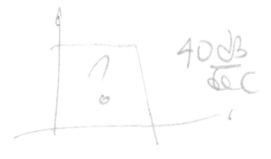
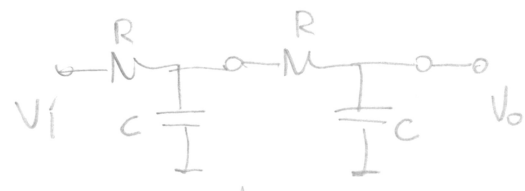
# ACTIVE FILTERS

FILTROS PASSIVOS :



cai muito pouco ...

Soluçao: cascata → duplo # polos



... MAS entra o problema do loading nos estagios Eles se "comegam"

Posso:



BUFFER KOLA estagios

na saída gera  $Z_{out} = 0$  posso colocar na entrada p/ dar  $Z_i = \infty$

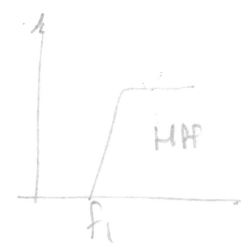
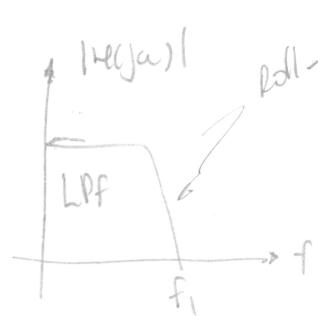


$Z_i = \infty$

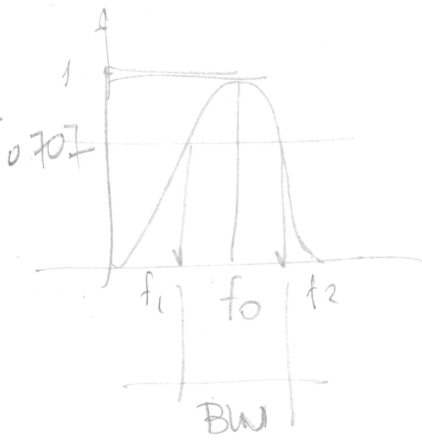
ACTIVE FILTERS → usam opamps. Outra vantagem podem resultar em ganhos, Passivos não conseguem ir além do 0 dB. ponto negativo → opamps ã chegam freqs. altas (max ≈ 1MHz)

## TIPOS FILTROS

ALL PASS  
mexe só na fase (equalização)



Lembrando



BW = banda  
passagem

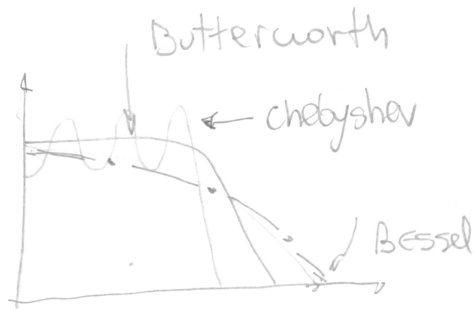
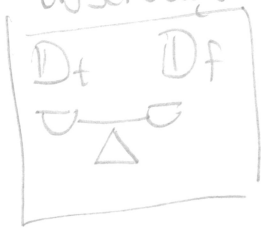
Linear: cai 0,707 do  
MÁXIMO (

dB: cai -3 dB

Q = FATOR QUALIDADE =  $f_0 / BW$       Q =  $1/\alpha$   
 $\alpha$ : damping factor

PROTÓTIPOS

observações → Trade off



obs: mesmo n°  
pólos/zeros

Chebyshev → cai + rápido (Bom) mas paga o preço com  
Ripple na banda de passagem Phase not flat. ( $> 20 \frac{dB}{dec}$ )

Butterworth → suave, sem ripple, mas cai devagar. Phase  
not flat. ( $20 \frac{dB}{dec}$ )

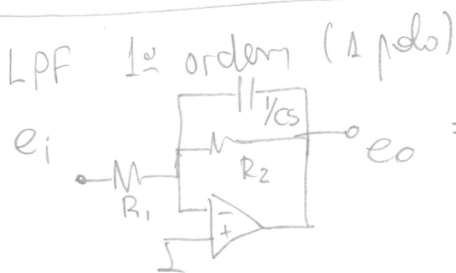
Bessel : Linear phase charact., sem overshoot/undershoot.  
Não distorce shape IDT ?

Linear phase  
→ distortions



pois daí  $\tau_g = -\frac{\partial \phi}{\partial \omega} = cte !$

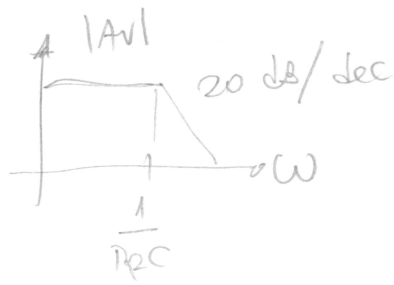
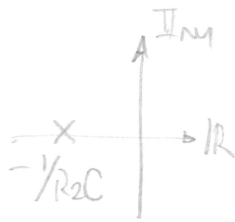
tabelas contém  
coeficientes p/  
cada filtro  
Chebyshev ... N



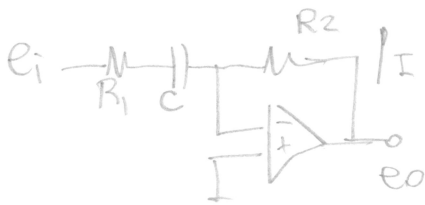
$$Z_p = R_2 // 1/s = R_2 / (R_2 s + 1)$$

$$I = \frac{e_o}{Z_p} = -\frac{e_i}{R_1} \rightarrow \frac{e_o}{e_i} = H(s) = \frac{-1/R_1 C}{s + \frac{1}{R_2 C}}$$

polo simples:  $-1/R_2C$



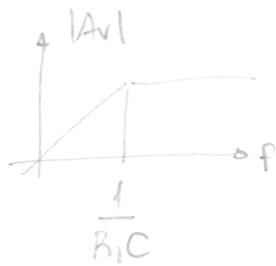
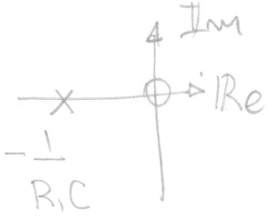
HPF 1º ordem



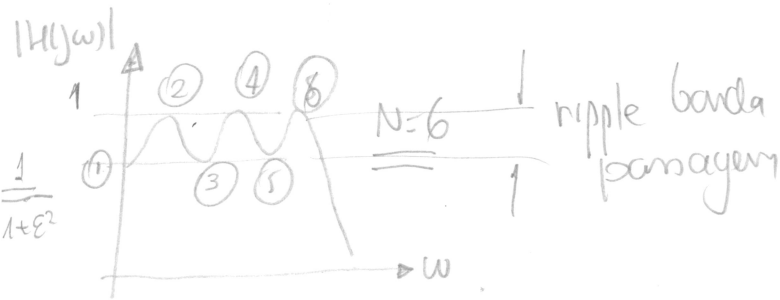
$$Z_S = R_1 + (1/CS) = \frac{R_1CS + 1}{CS}$$

$$I = \frac{e_o}{R_2} = -\frac{e_i}{Z_S}$$

$$\frac{e_o(s)}{e_i} = H(s) = -\frac{R_2}{Z_S} = -\frac{R_2}{R_1} \frac{s}{s + \frac{1}{R_1C}}$$



Chebyshev - melhor solução em termos tamanho (# polos) vs. rolloff (paga preço resposta Dt no shape contido)

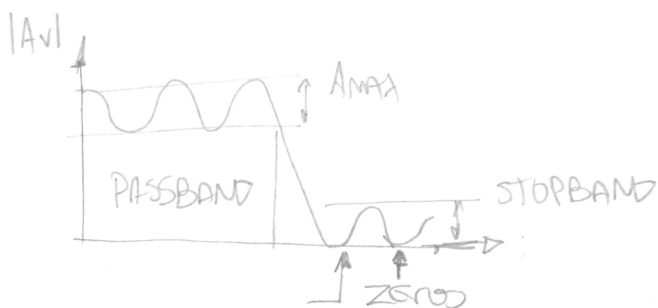


typ -> 1 dB  
3 dB  
5 dB  
quanto menor ripple melhor mas filtro fica + complicado

$$|T(jw)| = \frac{1}{\sqrt{1 + \epsilon^2 \cos^2 [N A \cos(w/w_p)]}}$$

$\uparrow$   
 $w < w_p \cos$   
 $w > w_p \cosh$

protótipo: encaixa crit nessa resposta

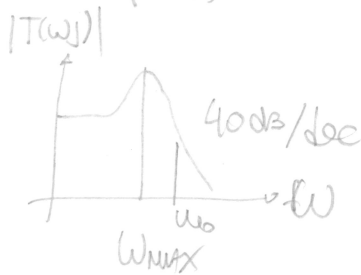


quanto + polos (N) mais rápido "cai" mas mais complexo é o filtro

FILTROS 2º ordem

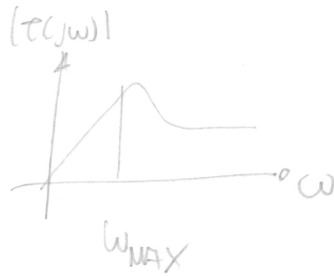
$$T(s) = \frac{a_2 s^2 + a_1 s + a_0}{s^2 + \left(\frac{\omega_0}{Q}\right) s + \omega_0^2}$$

LPF  $\rightarrow T(s) = \frac{a_0}{s^2 + s\frac{\omega_0}{Q} + \omega_0^2}$



zero =  $\emptyset$   
 polos: 2 complexos conjugados

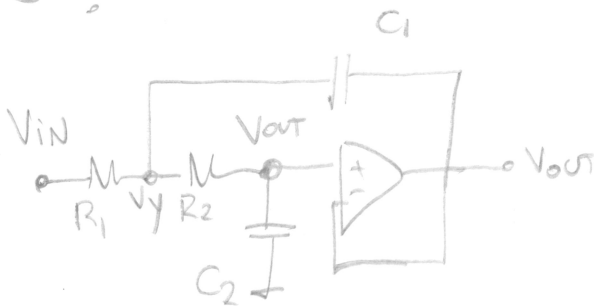
HAF  $T(s) = \frac{a_2 s^2}{s^2 + s\frac{\omega_0}{Q} + \omega_0^2}$



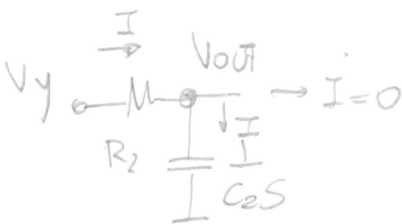
2 zeros origem  
 2 polos complexos conjugados

FILTROS PASSIVOS  $\rightarrow$  2º ordem precisa L  
 vantagens opamp active filter  $\rightarrow$  SINTETIZO efeito L com R C !

SALLEN KEY:  
 (LPF)



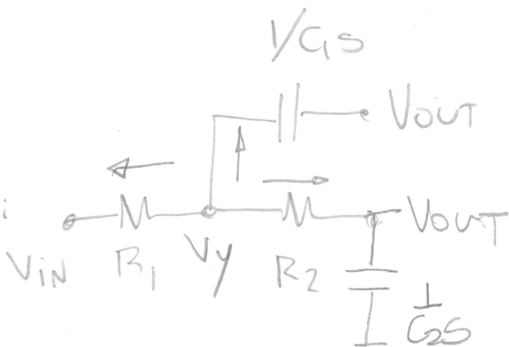
opamp  $\rightarrow$  buffer  
 Liga Vout sobre o capacitor



$$\frac{V_y - V_{out}}{R_2} = \frac{V_{out}}{1/C_2 S} \rightarrow V_y = V_{out} [R_2 C_2 S + 1]$$

ANALISANDO

nos Vy:



$$\frac{V_y - V_{in}}{R_1} + V_{out} C_1 S + (V_y - V_{out}) C_1 S = 0$$

+ facil que usar  
 $V_y = \frac{V_{out}}{R_2}$  !

consente 'e' a mesma!!!

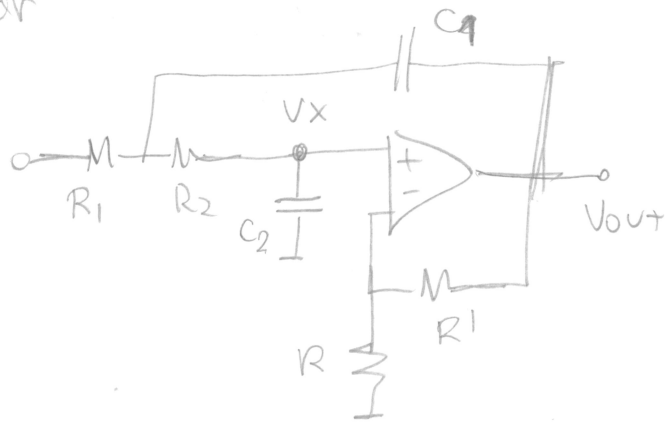
$$\frac{V_{out}(R_2 C_2 s + 1) - V_{in}}{R_1} + V_{out} C_2 s + [V_{out}(R_2 C_2 s + 1) - V_{out}] C_1 s = 0$$

Resolvo e agrego os termos:

$$\frac{V_{out}(s)}{V_{in}} = H(s) = \frac{1}{R_1 R_2 C_1 C_2 s^2 + (R_1 + R_2) C_2 s + 1}$$

2º ordem SGM Indutores

Passo inclusive adicionar ganho!



agora:

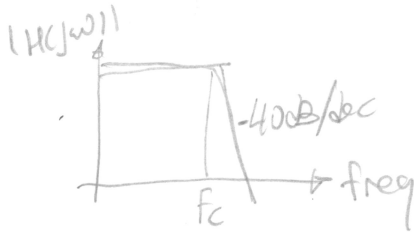
configuração

amplificadora não

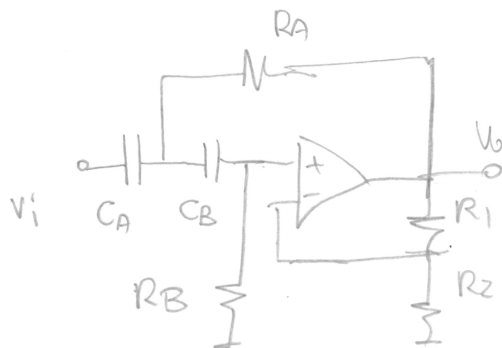
inversora:  $V_{out} = V_x \left[ 1 + \frac{R_1}{R} \right]$

$f_c = \frac{1}{2\pi \sqrt{R_1 R_2 C_1 C_2}}$  cutoff freq

Roll-off: 40dB/dec



SALLENKEY HPF

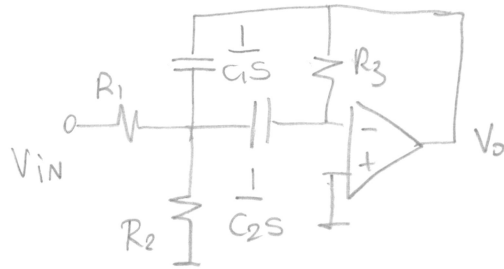


$$\frac{V_o}{V_i} = \frac{s^2 [R_A R_B C_A C_B] \left[ 1 + \frac{R_1}{R_2} \right]}{1 + s [R_A (C_A + C_B) - \frac{R_1}{R_2} R_B C_B] + s^2 [R_A R_B C_A C_B]}$$

$$\omega_0 = \frac{1}{\sqrt{R_A R_B C_A C_B}}$$

BPF

MULTIPLE  
FEEDBACK  
FILTER



$$\frac{V_o}{V_{in}}(s) = \frac{-R_2 R_3 C_2 s}{R_1 R_2 R_3 C_1 C_2 s^2 + R_1 R_2 (C_1 + C_2) s + (R_1 + R_2)}$$

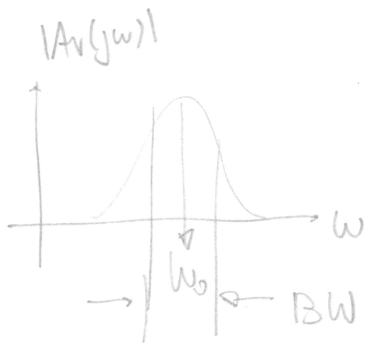
passando p/  
formato

$$\frac{-K \omega s}{s^2 + 2Q\omega_0 s + \omega_0^2} \rightarrow \frac{-\frac{1}{R_1 C_1} s}{s^2 + \frac{C_1 + C_2}{R_3 C_1 C_2} s + \frac{R_1 + R_2}{R_1 R_2 R_3 C_1 C_2}}$$

$$\omega_0 = \frac{1}{C} \sqrt{\frac{1 + R_1/R_2}{R_1 R_3}}$$

$$BW = 2/R_3 C$$

$$Q = \frac{1}{2} \sqrt{\frac{R_3}{R_1}} \sqrt{\left(1 + \frac{R_1}{R_2}\right)}$$



Referências

Thomas L. Floyd, Electronic Devices, 7th Edition, ch. 15 "Active Filters"

Hernando L.F. Conque, Analog Electronics Applications, ch. 20 "Filters"

Sedra Smith, Microelectronics, 7th ed, ch. 17, "Filters & Tuned Amplifiers"