

ANALOG CIRCUIT DESIGN ORAL NOTES

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Guide to these notes:

- If you paid for these notes, **you've been scammed**. I always give my notes for free.
- There are some additional notes at the end of the documents that could make some questions (like the orchard theorem) clearer to you, or at least, they did to me.
- I tried to explain and justify every step of each question, I hope you can follow my reasonings. If there's something that's "taken for granted" (except for filter theory that's purely mathematical and it's not explained in this course), there's probably an additional note in the end of the document
- This is a helpful guide to be read together with the lectures PDFs. I don't recommend to fully rely on the stupid things I wrote. Btw, PDFs contain typos so beware of that too
- Questions #24,#25,#1 are a bit meh, I didn't really know what to say about those. I really recommend not to use those as reference
- Question everything you read because during the oral you will be asked exactly that. Just to make few examples:
 - (Question #22 on $1/f$ noise) why do we take $\beta = \frac{1}{4}$? Can't we generalize the reasoning to all energy levels? Of course, but it's more complex. We analyze the traps at Fermi level just to demonstrate that if we consider multiple τ we end up with several lorentian shapes. Of course, on a more general view, there will be different families of traps (ions, defects on lattice, etc..) + different energy levels that will lead to something like $\frac{1}{f^\beta}$
 - (questions #14 and #15) How do we size the square length and area Λ, A_0 ? Is there any reasoning behind that? (See additional notes at the end of the document)
- NEVER EVER TAKE ANY FORMULA OR SYMBOL FOR GRANTED. The exact moment you memorize anything without a clear understanding of what you learned you will fail the oral. It's guaranteed. You must be able to justify anything that gets out your mouth or pen.
- Tip that helped me the most: don't memorize every step but memorize the first and last steps. Then memorize the track you need to follow to get from start to the end, it helps the flow of your speech and it won't take too much brain space
- It took me about 7 days to write these notes from scratch, so if my writing isn't clear, well, I'm sorry C: hope it helps anyway. Also, I speak maccheroni and I'm well aware of the English mistakes I made. However I didn't have enough time to be my grammar nazi
- There are two questions #19, the given pdf with the oral topics had two #19 and I didn't see that until I was done writing everything. Just ignore this mistake

If you're having any issue with this document just send an email to giacomo.tombolan@mail.polimi.it

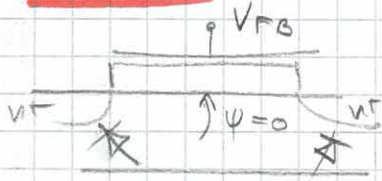
Topics for the orals

1. MOSFET's figures of merit: maximum voltage gain and cut-off frequency. Dependences on bias (strong, moderate and weak inversion).
2. Independent/interacting capacitors and poles. Extension of the time constant method. Middlebrook's theorem. Examples with RC networks.
3. The prototypical differential stage: from resistive to active loads. Common mode feedback and single ended option.
4. Single ended differential stage with mirror: Bias, input and output voltage swings, differential gain, Common mode gain.
5. Two-stage CMOS OTA: topology, frequency response, Miller compensation.
6. Two-stage CMOS OTA: frequency compensation with the nulling resistor. Implementing the nulling resistor
7. Two-stage CMOS OTA: frequency compensation with ideal voltage and current buffers. Impact of the buffer finite resistance.
8. Nested Miller Compensation
9. OTA Linear response. In-band zero-pole doublets and features of the settling response.
10. The slew rate limit. Impact on settling time. CMOS-OTA: Internal and external slew rate limits. Improving SR with class AB output stages.
11. CMOS single stage amplifiers: telescopic and folded cascode structures. Motivations, performance and linear swing
12. Output stages: Class A vs. class B output stage. Efficiency and distortion.
13. Output stages: Total harmonic distortion and feedback
14. Variability and matching: Relative matching of resistors. Pelgrom's formula
15. Variability and matching: Relative matching of threshold voltage values. Pelgrom's formula
16. OTA: Offset. Deterministic and statistical contributions to input referred offset.
17. OTA: Common-mode rejection ratio. Deterministic and statistical limits to CMRR
18. Quantitative description of noise: the power spectral density concept, thermal noise in resistors and MOSFETs
19. Input referred noise sources of a two-port network. Definitions and derivation. Extension to the differential stage
19. Noise models: The Nyquist argument for the thermal noise power spectral density
20. Noise models: Shot noise model. Application to p-n junctions and MOSFETs in weak inversion
21. Trapping noise: trapping noise in a resistor
22. McWorther model of the $1/f$ noise in MOSFETs. Tvidis formula.
23. Introduction to analog filters: Ideal performance. Limits of the causal response. Group delay and signal distortion.
24. LP filters: Filter mask and numerical parameters (selectivity, discrimination). Families of LP filter functions (Butterworth, Chebyshev, Bessel, ...) and their properties.
25. Mapping: Motivations, HP to LP transformation, BP to LP transformation
26. Active cells: The Sallen-key. Sizing options and sensitivity.
27. The universal cell. Deriving the block-diagram. Properties. The Tow-Thomas cell.
28. Ladder networks: Orchard theorem, implementation with active integrators. Flow-graph derivation procedure. Denormalization.
29. Ladder networks: implementations with gyrators. Topologies of gyrators (OP-AMP, OTA based). Properties and limitations
30. Dynamic range in filters: number of equivalent bits, guidelines to reduce the noise floor.
31. Impact of OP-AMP non-idealities on reference radial frequency and Q factor of a biquad cell.
32. Switched capacitor filters: Motivations, concepts, implementation of the ideal integrator. Stray insensitive topologies
33. SC filters: sampling, transfer function, output spectrum. Anti-aliasing filter and clock frequency.
34. SC filters: trade-off between settling time and charge sharing in sizing the switches.

1) MOS FET + FET. Dependences on strong, moderate + weak inv.

Let's see first some characteristics of the MOSFET:

- $V_{GT} = V_{FB} \rightarrow$ off state $\rightarrow \psi_s = 0$



Substrate forms diodes

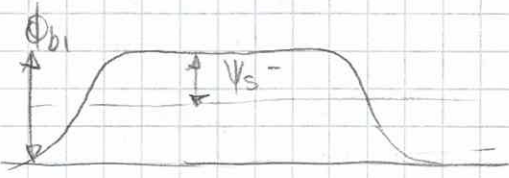


with both S, D. We therefore just have

a ϕ_{bi} (built-in potential) related to the diodes.

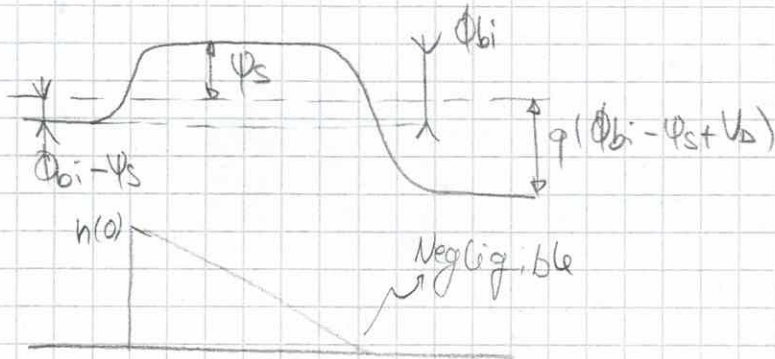
For $V_{GT} < V_T$ we are sub-threshold

- $V_{GS} < V_T$ and $V_{DS} = V_D = 0$



Energy barrier is now lower

- $V_{FB} < V_{GS} < V_T$ and $V_{DS} > 0$ ($V_S = 0$)



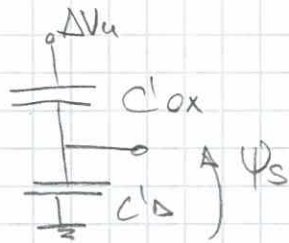
Since current is low \rightarrow no ohmic drop \rightarrow potential energy along the channel remains fairly constant. However, on the source we have a ϕ_{bi} that (Boltzmann) leads to a carrier concentration of $n(0) = N_s e^{-\frac{q \phi_{bi}}{kT}}$ while on the drain side, $V_D > 0$ will lead to a negligible concentration.

This concentration leads to a bipolar-like diffusion current that is

$$I_{DIFF} = I_s e^{\frac{q(V_{GS} - V_T)}{kT}} \quad g_D = \frac{\partial I}{\partial V_{GS}} = I_s \cdot \frac{q}{kT} e^{\frac{q(V_{GS} - V_T)}{kT}} = q \frac{I_{DS}}{kT}$$

$$= \frac{I_{DS}}{n V_{TH}}$$

n :



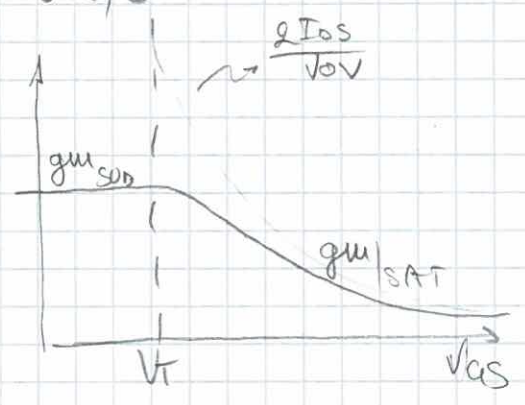
$$\Delta\psi_s = \Delta V_g \cdot \frac{C'_{ox}}{C'_{ox} + C'_d} \rightarrow \Delta\psi_s = \frac{\Delta V_g}{n}$$

with $n = 1 + \frac{C'_d}{C'_{ox}} \sim 1.5$

Maximum gain

$g_{m|SUB} = \frac{I_{D,S}}{nV_{TH}}$

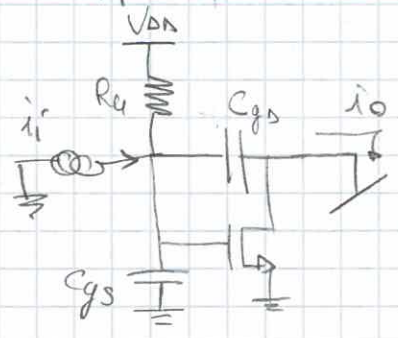
$g_{m|SAT} = \frac{2I_{D,S}}{V_{OV}}$



$\mu|SUB = \frac{I_{D,S}}{nV_{TH}} \cdot \frac{V_A}{I_{D,S}} = \frac{V_A}{nV_{TH}} \rightarrow \text{MAX } \mu \text{ ACHIEVABLE}$

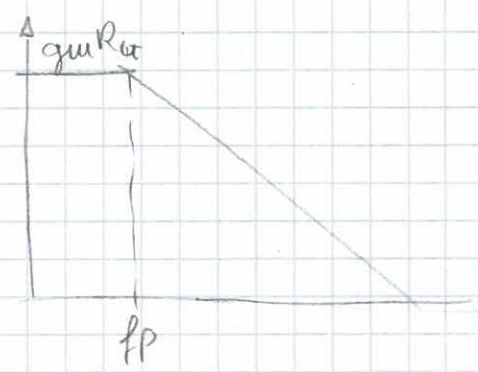
This will be the max achievable gain

Cutoff frequency



$C_{DC} = -g_m R_e = \frac{i_{out}}{i_{in}}$

$f_p = \frac{1}{2\pi C_{ox} \cdot R_e}$



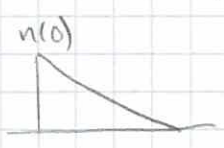
$g_m R_e \cdot \frac{1}{2\pi R_e C_{ox}} = \frac{g_m}{C_{ox} \cdot 2\pi}$

We define the gain-BW product as $f_T = \frac{g_m}{C_{ox}} \cdot \frac{1}{2\pi}$

$f_T = \frac{2\mu V_{OV}}{2\pi W L C'_{ox}} = \frac{2 \cdot \frac{1}{2} \mu C'_{ox} \frac{V_{OV}}{L}}{2\pi W L C'_{ox}} = \frac{\mu V_{OV}}{2\pi L^2} = \frac{\bar{v}}{2\pi L} = \frac{1}{2\pi t_{in}}$

$\frac{V_{OV}}{L} = \text{electric field } \bar{v} = \mu E = \mu \frac{V_{OV}}{L}$

This is not true for SUB-thresh:



$\bar{J}_{DIFF} = q D_n \left(\frac{n(0)}{L} \right) \rightarrow \frac{\Delta n}{\Delta x}$ diffusion density along the channel

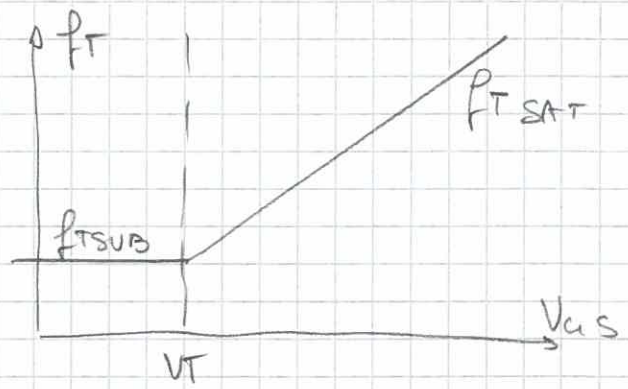
Stored charge will be $Q' = q \cdot \frac{n(0) \cdot L}{2}$ area of the triangle

So $t_{DIFF} = \frac{Q'}{\bar{J}_{DIFF}} = \frac{\frac{1}{2} q n(0) \cdot L}{q D_n \frac{n(0)}{L}} = \frac{L^2}{2 D_n}$

So

$$f_{SUB} = \frac{1}{2\pi t_{DIFF}} = \frac{D_n}{\pi L^2}$$

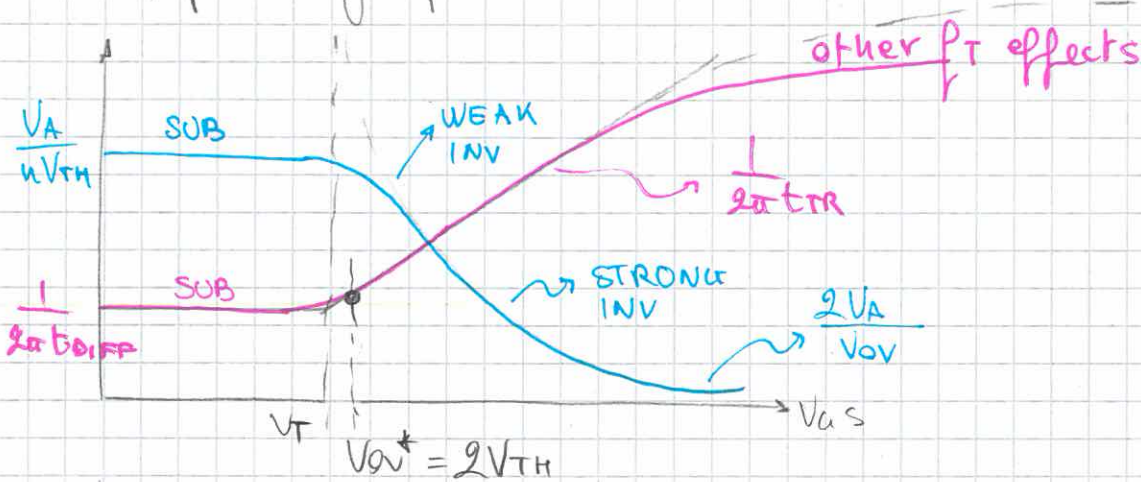
$$f_{SAT} = \frac{g_m}{2\pi C_{ox}} = \frac{\mu V_{ov}}{2\pi L^2}$$



$$f_{SUB} = f_{SAT} \rightarrow \frac{L^2}{\mu V_{ov}} = \frac{L^2}{2D_n} \Rightarrow \frac{2D_n}{\mu} = V_{ov}^*$$

Since $D_n = \frac{kT}{q} \mu$ $V_{ov}^* = \frac{2kT}{q} \approx 51 \text{ mV @ } 300K$

Real, first graph



2) Time constant method + examples + extension

For a LTI network the transfer function is:

$$T(s) = T(0) \frac{N(s)}{D(s)} = \frac{a_m s^m + a_{m-1} s^{m-1} + \dots + 1}{b_n s^n + b_{n-1} s^{n-1} + \dots + 1}$$

$N(s)$ are zeros $D(s)$ are poles. For a 3 capacitors network

$$b_1 = C_1 R_1^{(0)} + C_2 R_2^{(1)} + C_3 R_3^{(0)}$$

$$b_2 = C_1 C_2 R_1^{(0)} R_2^{(1)} + C_1 C_3 R_1^{(0)} R_3^{(2)} + C_2 C_3 R_2^{(0)} R_3^{(2)}$$

$$b_3 = C_1 C_2 C_3 R_1^{(0)} R_2^{(1)} R_3^{(1,2)}$$

$$a_1 = C_1 R_{01}^{(0)} + C_2 R_{02}^{(0)} + C_3 R_{03}^{(0)}$$

$$a_2 = C_1 C_2 R_{01}^{(0)} R_{02}^{(1)} + C_1 C_3 R_{01}^{(0)} R_{03}^{(2)} + C_2 C_3 R_{02}^{(0)} R_{03}^{(2)}$$

$$a_3 = C_1 C_2 C_3 R_{01}^{(0)} R_{02}^{(1)} R_{03}^{(1,2)}$$

$$D(s) = b_3 s^3 + b_2 s^2 + b_1 s + 1 \quad N(s) = a_3 s^3 + a_2 s^2 + a_1 s + 1$$

Since the coefficients are related to the circuit topology, order of capacitors doesn't matter

We can denote roots of the denominator so that

$$D_3(s) = \left(1 - \frac{s}{p_3}\right) \left(1 - \frac{s}{p_2}\right) \left(1 - \frac{s}{p_1}\right) = b_3 s^3 + b_2 s^2 + b_1 s + 1$$

By comparing the expressions we find that:

$$b_1 = - \left(\frac{1}{p_3} + \frac{1}{p_2} + \frac{1}{p_1} \right)$$

If there's a dominant pole out of the three, $\frac{1}{p}$ will be much higher; for example $\frac{1}{p_1} \gg \frac{1}{p_2}, \frac{1}{p_3}$ so $\frac{1}{p}$

$$b_1 = C_1 R_1^{(0)} + C_2 R_2^{(0)} + C_3 R_3^{(0)} \approx - \frac{1}{p_1}$$

On the other hand, for high frequency

$$b_3 s^3 + b_2 s^2 + b_1 s + 1 \approx b_3 s^3 + b_2 s^2 = s (s b_3 + b_2)$$

Therefore
$$P_H = -\frac{b_2}{b_3} = \frac{C_1 C_2 R_1 R_2^{(1)} + C_1 C_3 R_1 R_3^{(1)} + C_2 C_3 R_2 R_3^{(1)}}{C_1 C_2 C_3 R_1^{(1)} R_2^{(1)} R_3^{(1)}}$$

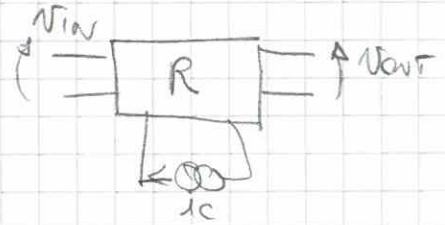
We can simplify the terms and find that

$$P_H = -\frac{b_2}{b_3} = \frac{-1}{C_3 R_3^{(1,2)}} - \frac{1}{C_2 R_2^{(1,3)}} - \frac{1}{C_1 R_1^{(2,3)}} =$$

$$= -\left(\frac{1}{\tau_3^\infty} + \frac{1}{\tau_2^\infty} + \frac{1}{\tau_1^\infty}\right) = -\frac{1}{\tau_{\infty}}$$

Zero of a network with a single pole

LTI network \rightarrow $V_{out} \propto$ linear to V_{in} , i.e.
 $V_c \propto$ linear to V_{in} , i.e.



$$\begin{cases} V_{out} = A_0 V_{in} + R_m i_c \\ V_c = B_0 V_{in} + R_1 i_c \end{cases}$$

Since $\frac{1}{sC} i_c = V_c \rightarrow i_c = -sC V_c$

$$V_c = B_0 V_{in} - sC R_1 V_c \rightarrow V_c = \frac{B_0 V_{in}}{(1 + sR_1 C)}$$

$$V_{out} = V_{in} \left[A_0 - s \frac{R_m C B_0}{1 + sR_1 C} \right]$$

$$= V_{in} A_0 \frac{1 + sC \left[R_1 - \frac{R_m B_0}{A_0} \right]}{1 + sC R_1}$$

pole is $\tau_p = R_1 C$

For the zero to occur, it must be $A_0 V_{in} + R_m i_c = 0$
 Therefore at the same time

$$\underline{V_c|_{ZERO} = -\frac{B_0 R_m i_c|_{ZERO}}{A_0} + R_1 i_c|_{ZERO}}$$

Resistance seen on C during zero condition will be

$$\underline{\frac{V_c}{i_c}|_{ZERO} = \left(R_1 - \frac{R_m B_0}{A_0} \right)}$$

Results :

- $f_{\text{pole}} = \frac{1}{2\pi R_1 C}$

↳ There's a pole

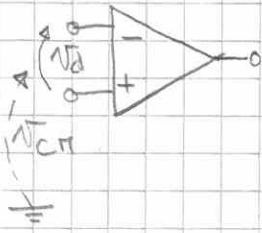
- $f_z = \frac{1}{2\pi R_2 C}$

↳ There's a zero

- DC gain is $\frac{V_{\text{out}}}{V_{\text{in}}}$ with C open

3) Diff stage: from resistive to active load. CM feedback + single ended option

$$V_d = V^+ - V^-$$

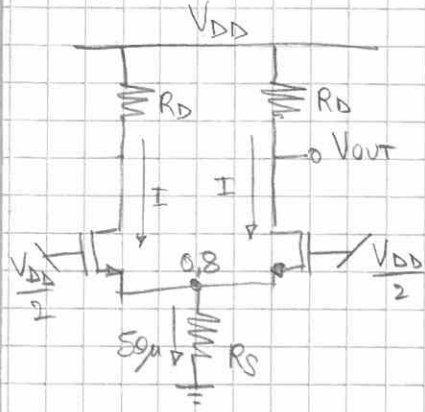


$$V_{out} = G_D \cdot V_d + G_{CM} V_{CM}$$

$$V_{CM} = (V^+ + V^-) / 2$$

Goal: - high differential gain (∞ ideally)
 - low common mode gain (0 ideally)

Differential stage \rightarrow resistors



Put example with: $R_L = 10k\Omega$ $R_S = 16k\Omega$

$$V_B = \frac{V_{DD}}{2} \quad 2I = 50\mu A$$

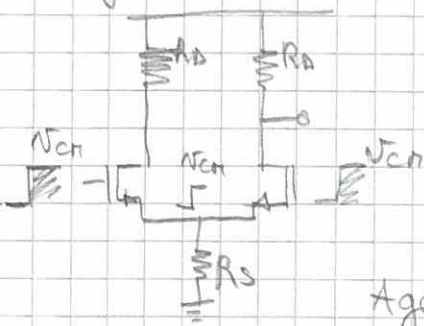
$$V_{out} = g_m \frac{V_d}{2} \cdot R_D = \frac{2I_D}{V_{ov}} \cdot \frac{R_D}{2} V_d$$

$$G_D = \frac{I_D R_D}{V_{ov}} \quad \text{where } I_D R_D = \text{bias voltage drop on } R_D$$

Issue: gain is limited by voltage drop on resistor:

To increase G_D : $\rightarrow R_D \uparrow \rightarrow M_1, M_2$ go out of saturation
 $\rightarrow V_{ov} \uparrow \rightarrow g_m$ can saturate (EKV model)

Solution: load that shows high impedance with low bias voltage drop.



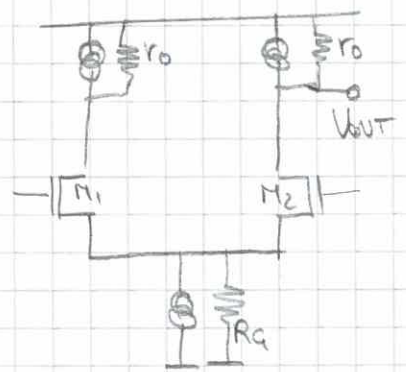
$$V_{out} / V_{CM} = \frac{V_{CM}}{R_S} \cdot \frac{1}{2} \cdot R_D \rightarrow G_{CM} = -\frac{R_D}{2R_S}$$

$$G_{DM} = \frac{I_D R_D}{I_D 2R_S} = \frac{\Delta V_{R_D}}{\Delta V_{R_S}}$$

Again: gain is limited by voltage drops on resistors (ideally $G_{CM} \rightarrow 0$)

Differential stage → active loads

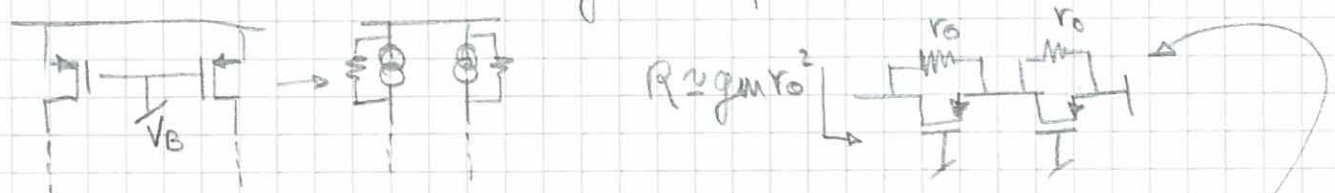
High impedance + low voltage drop → current generators!



$G_D = g_m \cdot \frac{1}{2} \cdot r_o = \frac{\mu}{2}$ → max transistor gain

$G_{CM} = \frac{r_o}{2R_G}$

Current generators are implemented using mosfets.

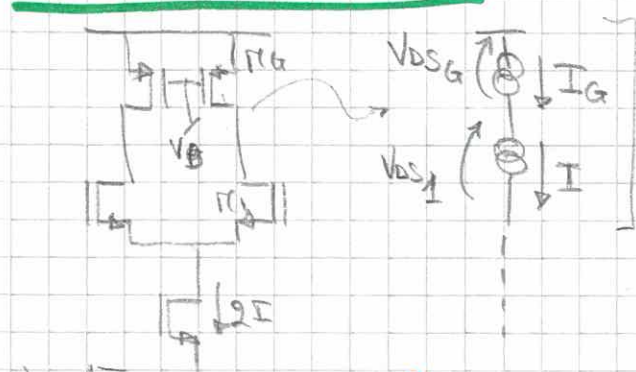


Since we have low voltage drop, we can use cascode structures in order to increase load impedance

We can do the same for tail generators so that G_{CM} can be decreased. at the expense of burning more power supply → High CMRR means more power consumption

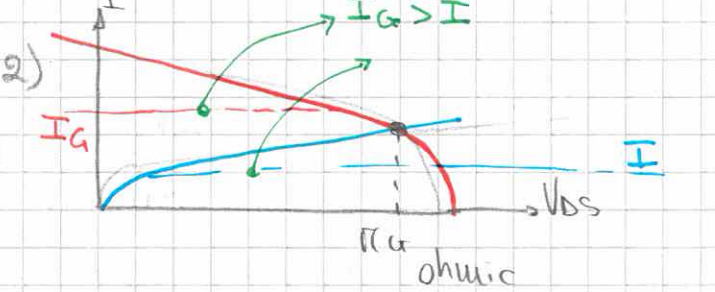
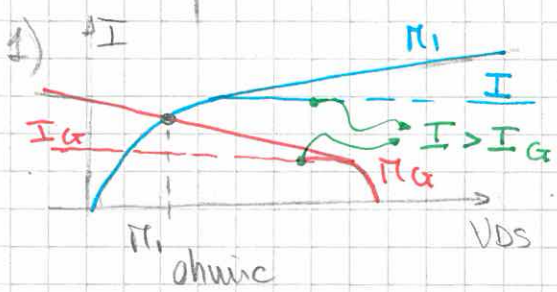
CMRR = Common Mode Rejection Ratio

Active bias issue



Currents need to be exactly matched, otherwise:

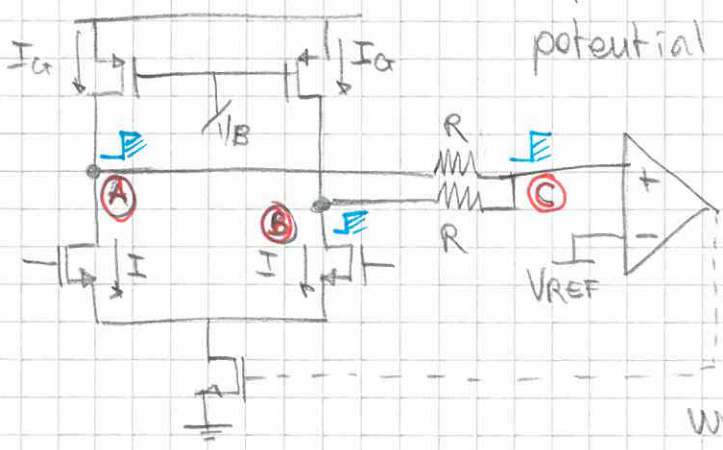
- 1 $I_G < I \rightarrow M_1$ ohmic
- 2 $I_G > I \rightarrow M_G$ ohmic



Common mode feedback

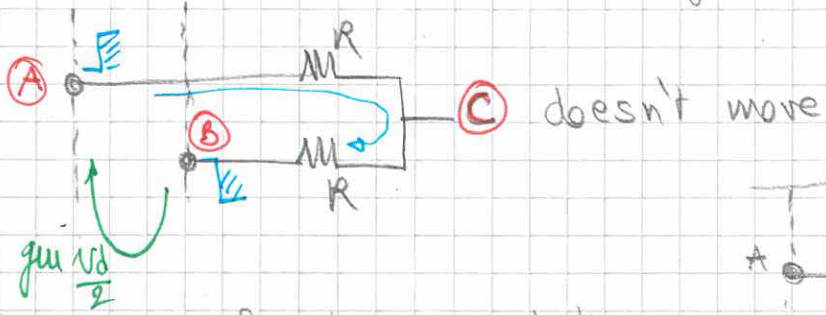
This helps rebalancing the currents so that they can be matched

If $I_G > I \rightarrow A, B$ will rise in potential, thus C will rise too

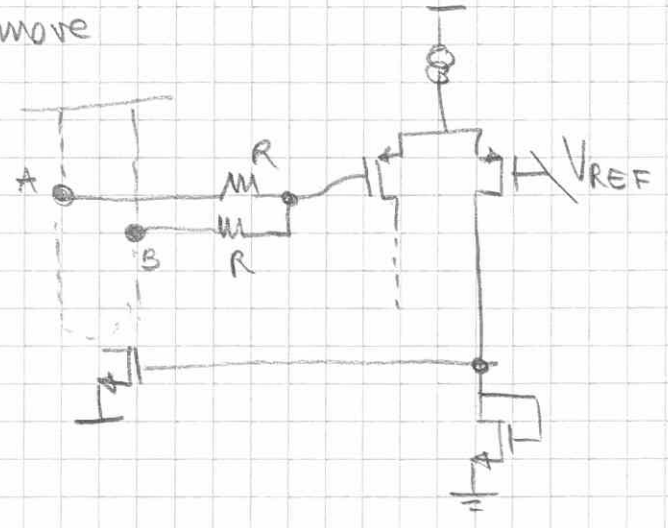


It's easy to see that two resistors are used to generate the average between A, B because we need not to change

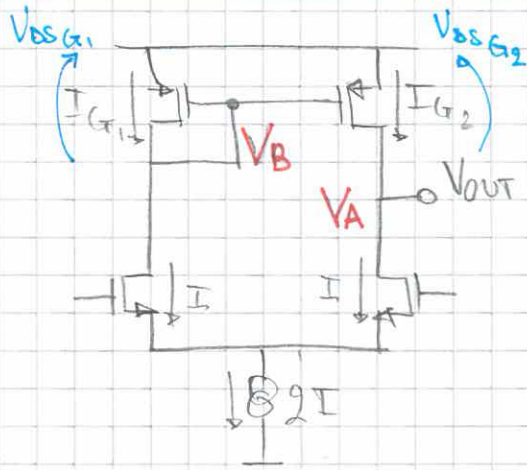
the common mode bias during differential stimuli:



Example of implementation



Another compensation: Single ended opamp



By using a current mirror we can eliminate the need for a CM feedback configuration. Cost:

- No fully differential stage
- Symmetry (\rightarrow CMRR deterministic contribution)

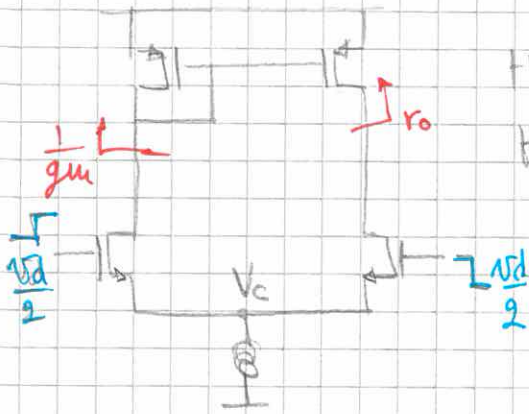
If $V_B = V_A \rightarrow I_{G1} = I_{G2} = I$ to first order ($V_{GS1} = V_{GS2}$)

When $2I$ is changed, V_B changes as well, the transdiode automatically adjusts its current in order to match

$$I_G = I$$

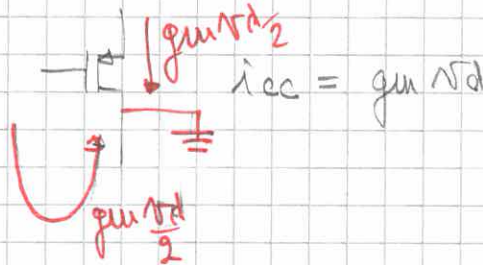
1) Single ended differential stage with mirrors: bias, dynamics, G_D, G_{Dn}

We see $\frac{1}{g_m}, r_o \rightarrow$ Stage is unbalanced

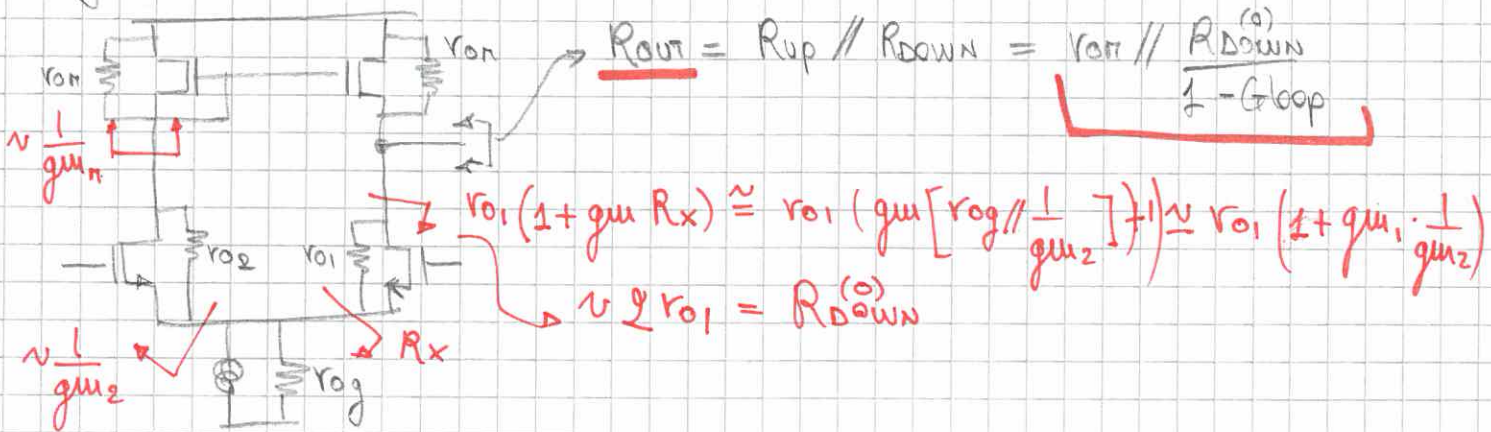


this means V_c is changing with respect to $V_d \rightarrow V_d$ isn't split symmetrically

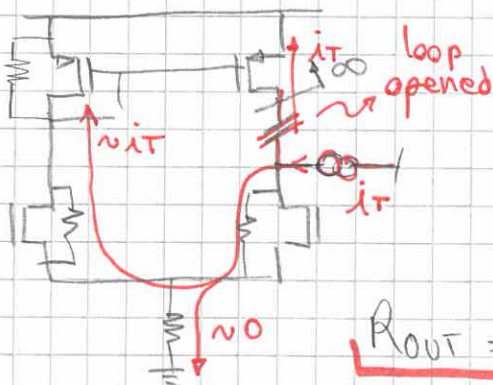
Turnaround: Norton theorem



By shorting V_{out} , stage will now see $\frac{1}{g_m}$ and $0_r \rightarrow$ we "recovered" symmetry, therefore we can easily compute G_D using Norton



We have a feedback loop:



$G_{loop} = -1 \rightarrow R_{down} = \frac{2 r_{o2}}{1 - G_{loop}} =$

$R_{down} = \frac{r_{o1} \cdot 2}{2}$

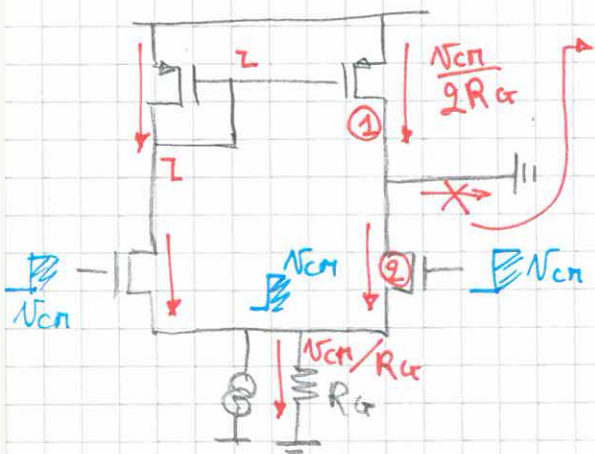
$R_{out} = r_{o1} \parallel r_{o1}$

$r_{on} = r_{o1}$

$G_D \Big|_{Norton} = i_{cc} \cdot R_{out} = g_m \cdot (r_{o1} \parallel r_{o1}) = \frac{g_m r_o}{2} = \frac{\mu}{2} \approx 7V$

We can simplify $G_D = \frac{2I}{V_{ov}} \cdot \frac{1}{2} \cdot \frac{V_A'}{V} = \frac{V_A^{(o)}}{V_{ov}} \cdot \frac{1}{L_{min}} \rightarrow 0,35 \mu$

Common mode gain



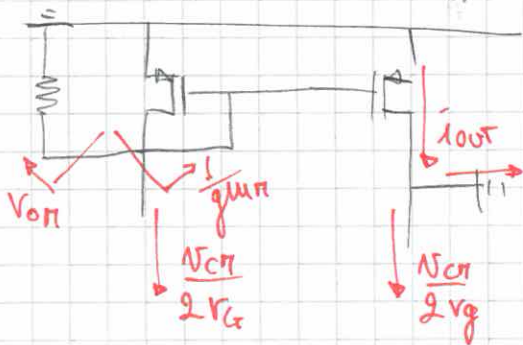
Ideally, current generated in ① would be sucked in ② so $i_{cc} = 0$

We can say that ① is slightly mismatched with respect to ② by an error ϵ , so:

$$i_{cc} = \epsilon \cdot \frac{N_{cm}}{2R_G} \rightarrow G_{cm} = i_{cc} R_{out} = \epsilon \frac{R_{out}}{2R_G}$$

$$CMRR = \frac{G_D}{G_{cm}} = \frac{g_m R_{out}}{\epsilon \frac{R_{out}}{2R_G}} = \frac{2g_m R_G}{\epsilon}$$

Let's briefly quantify one contribution of ϵ :



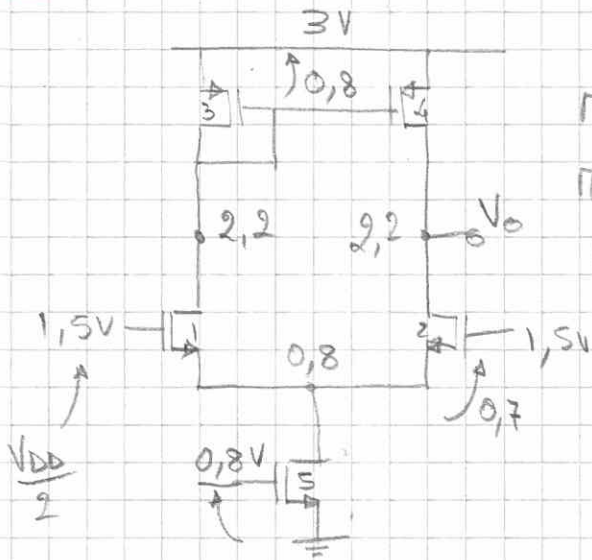
$$i_{out} = \frac{N_{cm}}{2R_G} \cdot \frac{V_{cm}}{V_{cm} + 1/g_{m11}} = \frac{N_{cm}}{2R_G} \cdot \frac{1}{1 + \frac{1}{g_{m11} V_{cm}}}$$

$$= \frac{N_{cm}}{2R_G} \left(1 - \frac{1}{g_{m11} V_{cm}}\right) = \frac{N_{cm}}{2R_G} (1 - \epsilon) \quad \frac{1}{1+x} \approx 1-x \text{ for } x \rightarrow 0$$

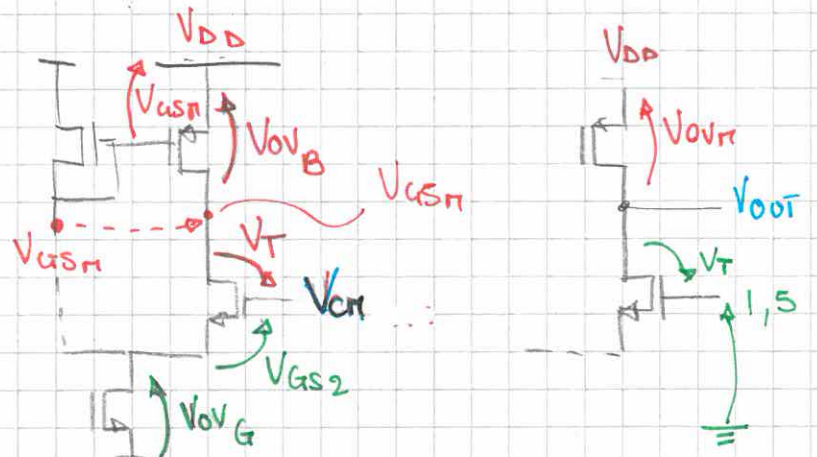
So $\epsilon \approx \frac{1}{\mu} \approx 1\%$ since $\mu = 100$

$$\text{Therefore } CMRR = \frac{2g_m R_G}{\epsilon} = \frac{2 \cdot 100}{1/100} \approx 86\text{dB}$$

Bias and dynamics (IN/OUT)



	I	V_{ov}	W/L	g_m	r_o
$\pi_{1,2}$	25μ	0,1	50	$0,5m$	560k
$\pi_{3,4}$	25μ	0,2	25	$0,25m$	560k
π_5	50μ	0,2	25	$0,25m$	140k



Input dynamics:

$$V_{IN}|_{max} = V_{DD} - V_{GS1} + V_T \approx 2,8$$

$$V_{IN}|_{min} = V_{ovG} + V_{GS2} \approx 0,9$$

$$V_{OUT}|_{max} = V_{DD} - V_{ovD} \approx 2,8$$

$$V_{OUT}|_{min} = V_{ovC} - V_T \approx 0,9$$

$$|\Delta V_{IN}^+| = 2,8 - 1,5 = 1,3V$$

$$|\Delta V_{IN}^-| = 1,5 - 0,9 = 0,6V$$

$$|\Delta V_{OUT}^+| = 2,8 - 2,2 = 0,6V$$

$$|\Delta V_{OUT}^-| = 2,2 - 0,9 = 1,3V$$

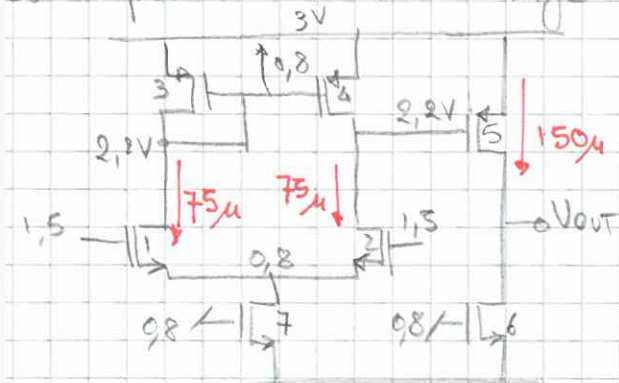
5) Two stages OTA: topology, freq response, Miller comp



$G_1 = \text{diff stage gain} \approx 40 \text{ dB}$,
 not enough for the $> 85 \text{ dB}$ spec
 $G_2 = \text{2nd stage gain}$

$$\underline{G_{TD}} = G_1 G_2 = \underbrace{g_{m1} (r_{o1} \parallel r_{o2})}_{140} \underbrace{g_{m5} (r_{o5} \parallel r_{o6})}_{140} \approx 86 \text{ dB}$$

We met the G_{TD} requirement. at the cost of more power dissipation \rightarrow 2nd stage requires bias:



μ_6 has to match μ_5 in order to get proper bias and V_{out} set at $V_{DD}/2$ (therefore $K_5 = K_6$)

$$\underline{G_{TD}} = \frac{g_{m1}}{V_{ov1}} \cdot \frac{V_{A1}}{2I_1} \cdot \frac{g_{m5}}{V_{ov5}} \cdot \frac{V_{A5}}{2I_5} = \frac{V_{A1}}{V_{ov1}} \cdot \frac{V_{A5}}{V_{ov5}}$$

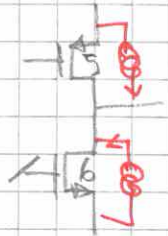
V_{ov5} has to be the same of $V_{ov3,4}$ because of symmetry

We can therefore choose proper L_5, L_6 to match gain by finding the $V_{A5} = \frac{L_5 \cdot 7V}{L_{min}}$

Note: when $\mu_5 \neq \mu_6$ we end up with a systematic offset

$$\text{that can be input translated to: } V_{os, in} = \left| \frac{V_{DD}/2 - V_{out}}{G_{TD}} \right|$$

Noise consideration on 2nd stage



$$S_{IN|TOT} = \frac{8kT\gamma}{g_{m1}} \left(1 + \frac{1}{2}\right) + 8kT\gamma g_{m5} \cdot \frac{1}{G_1^2 g_{m5}^2}$$

Division by G_1^2 makes 2nd stage noise negligible with respect to 1st stage

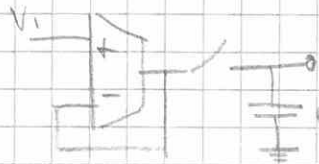
OTA applications



Resistive loads kill the loop gain

$$G_{loop} = g_{m1} \cdot \frac{R_{out}}{R_{out} + R_1 + R_2} \cdot R_1$$

We should use a buffer between G_{m1} stage and the resistive load \rightarrow OPA/OP

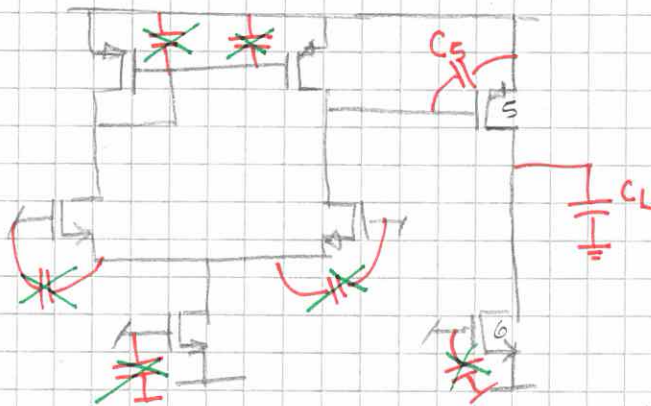


We could use the OTA as a S&H:

• No resistive load

• Capacitive load \rightarrow could generate instability

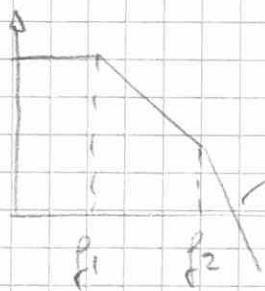
Frequency response



$C_{1,2,3,4}$ see low impedance \rightarrow pole is at high frequency

C_7, C_6 do not affect signal response.

C_{gs}, C_L affect frequency response by introducing two poles well beyond $G_{loop} = 1$:



$$f_1 = \frac{1}{2\pi C_L (r_{os1} \parallel r_{os2})}$$

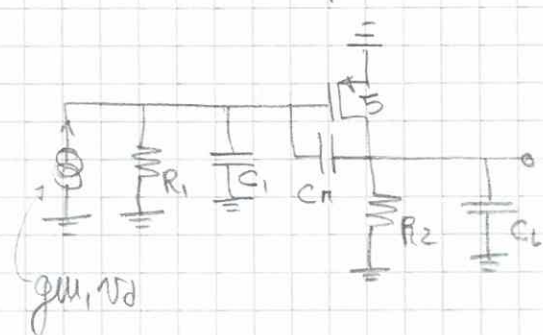
$$f_2 = \frac{1}{2\pi C_{gs} (r_{o1} \parallel r_{o2})}$$

We get 0dB with $-40dB \rightarrow$ UNSTABLE

This means that if we used the OTA as a buffer with a capacitive load, it would be unstable. or UNCOMPENSATED at unity gain

Miller compensation

Let's simplify the stage and let's add a comp. capacitor:



$R_1 =$ impedance of 1st stage

$R_2 =$ " " 2nd "

$C_1 = C_{gs}$

$C_M =$ Miller comp. capacitor

$$\underline{T(s)} = G_D(s) \cdot \frac{s^2 a_2 + s a_1 + 1}{s^3 b_3 + s^2 b_2 + s b_1 + 1} \quad \text{where}$$

$$b_1 = C_1 R_1^{(0)} + C_M R_M^{(0)} + C_L R_L^{(0)}$$

$$b_2 = C_M C_1 R_M^{(0)} R_1^{(1)} + C_1 C_L R_1^{(0)} R_L^{(1)} + C_M C_L R_M^{(0)} R_L^{(1)}$$

$$b_3 = C_M C_1 C_L R_M^{(0)} R_1^{(1)} R_L^{(1,2)}$$

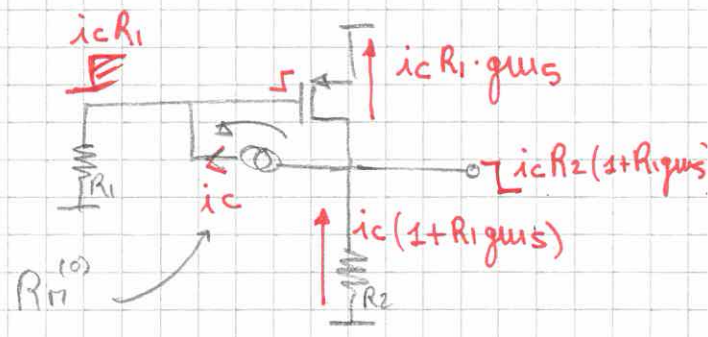
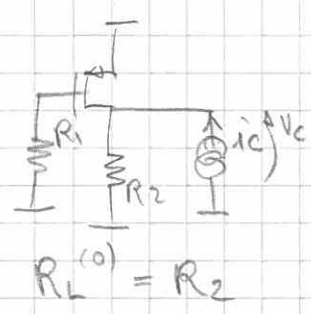
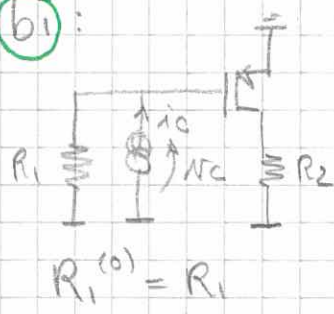
$$a_1 = C_1 R_{o1}^{(0)} + C_M R_{oM}^{(0)} + C_L R_{oL}^{(0)}$$

$$a_2 = C_M C_1 R_{oM}^{(0)} R_{o1}^{(1)} + C_M C_L R_{oM}^{(0)} R_{oL}^{(1)} + C_1 C_L R_{o1}^{(0)} R_{oL}^{(1)}$$

We can already say that there's no a_3 because C_L is directly tied to output $\rightarrow \infty$ zero

Since capacitors interact (they're dependent on each other), we can already say that $b_3 = 0$ for whatever combination
 So \rightarrow max 2 poles, max 2 zeros

(b1):

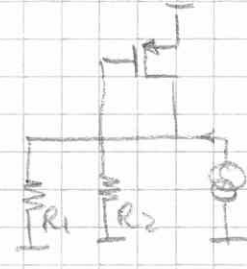
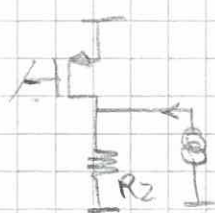
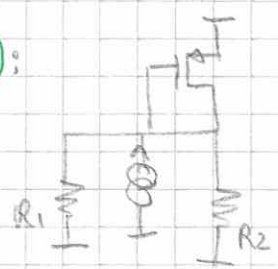


$R_{\pi}^{(0)}$: $V_c = ic R_1 - (-ic R_2 (1 + R_1 gm_S))$
 $R_{\pi}^{(0)} = \frac{V_c}{ic} = R_1 + R_2 (1 + R_1 gm_S) = R_2 + R_1 (1 + gm_S R_2)$

Remember that because of Miller effect $R_{\pi} = R_2 + R_1 (1 + G_2)$ where G_2 is the voltage gain between terminals of C_{π}

b_1 = $\underbrace{C_1 R_1}_{\sim 90n} + \underbrace{C_L R_2}_{\sim 450n} + \underbrace{C_{\pi} [R_2 + R_1 (1 + gm_S R_2)]}_{\sim 8\mu s} \approx \underline{\underline{C_{\pi} R_1 gm_S R_2}}$

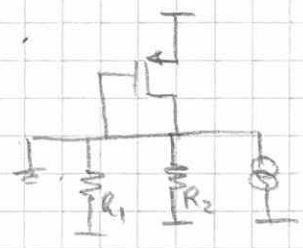
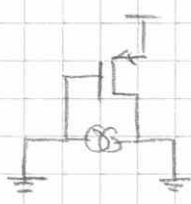
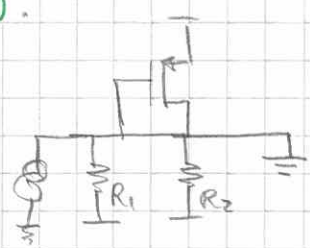
(b2):



$R_1^{(\pi)} = R_1 // R_2 // \frac{1}{gm_S}$
 $R_L^{(\pi)} = R_2$
 $R_1^{(\pi)} = R_1 // R_2 // \frac{1}{gm_S}$
 $\approx \frac{1}{gm_S}$

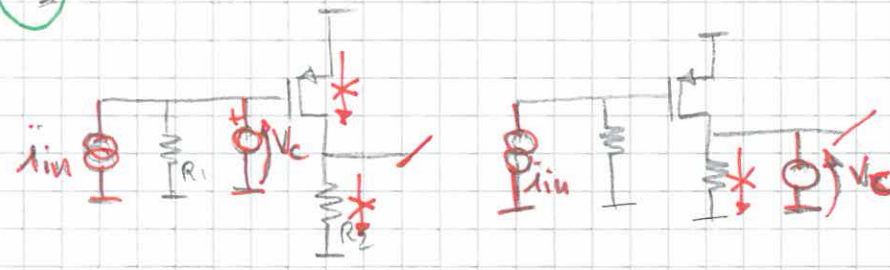
b_2 $\approx C_{\pi} C_1 R_1 R_2 gm_S \cdot \frac{1}{gm_S} + C_1 C_L R_1 R_2 + C_{\pi} C_L R_1 R_2 gm_S \cdot \frac{1}{gm_S}$
 $= R_1 R_2 [C_{\pi} C_1 + C_1 C_L + C_{\pi} C_L] = \underline{\underline{C_1 C_L R_1 R_2 + [C_1 + C_L] C_{\pi} R_1 R_2}}$

(b3):



$b_3 = 0$ for every possible combination

a₂:

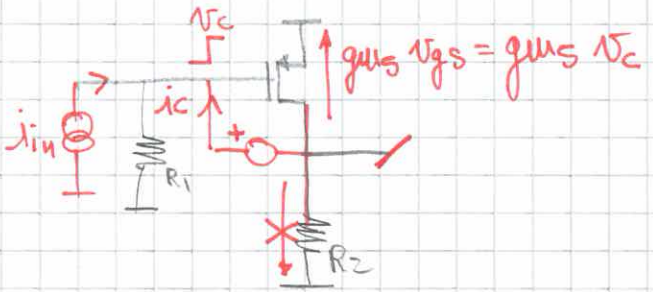


Setup for zeros:

- Vout can't move
- Input generator is ON
- Capacitor generator outputs i_c and v_c

R_{o1}⁽⁰⁾: Vout can't move → V_{R2} = 0 → I_{R2} = 0 → I_{cs} = 0 →
 → V_{gs} = 0 → v_c can't move] → R_{o1}⁽⁰⁾ = 0

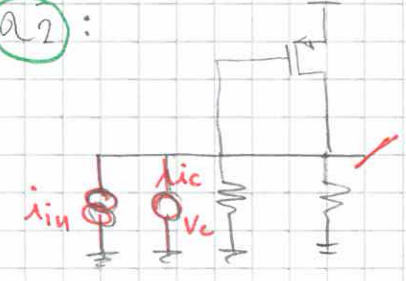
R_{o2}⁽⁰⁾: Vout can't move → v_c can't move] → R_{o2}⁽⁰⁾ = 0



Vout can't move → V_{R2}, I_{R2} = 0 → i_c + i_{cs} = 0 → i_c + g_mv_c = 0

$\frac{v_c}{i_c} = \underline{R_{o1}^{(0)}} = \underline{\frac{-1}{g_m}}$ while $i_{in} = \frac{v_c}{R_1} - i_c$

a₂:



a₂ = 0 it's easy to derive that
 R_{o1}^(∞), R_{o2}^(∞) = 0 because Vout can't move
 and both C₁, C₂ are tied to Vout
 when C_π is short

$G_D(s) = G_D(0) \cdot \frac{1 - s C_{\pi} / g_m}{s^2 b_2 + s b_1 + 1}$

Frequency results estimation

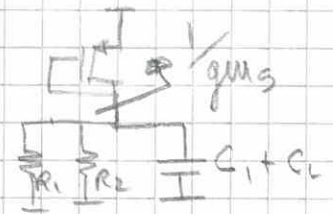
by seeing the large C_{π} , we can say that the resulting two poles will differ by more than a decade. Approx:

$$\underline{p_L} \approx -\frac{1}{b_1} = \frac{-1}{C_{\pi} R_1 g_{mS} R_2} \rightarrow C_{\pi} \text{ contributes the most}$$

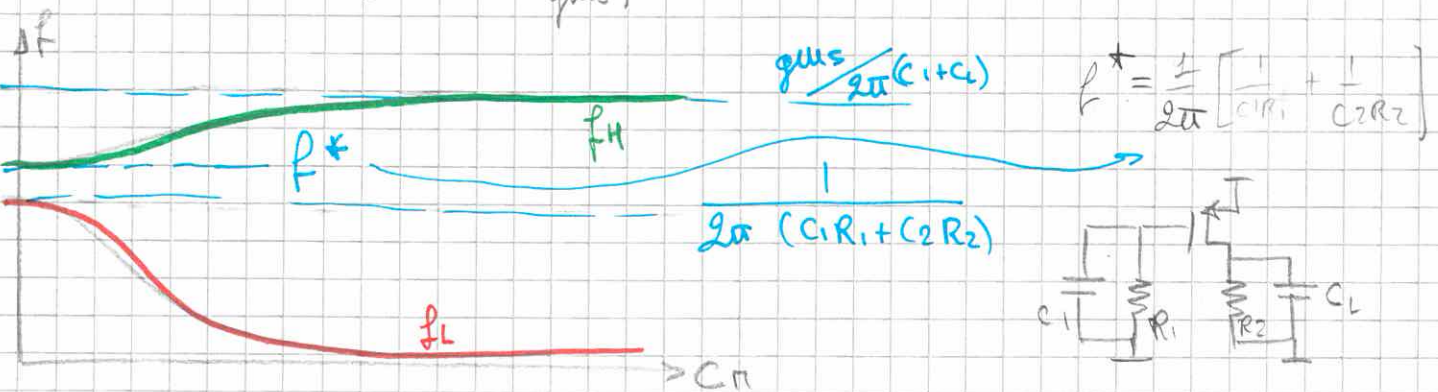
$$\underline{p_H} \approx -\frac{1}{b_2} = \frac{-C_{\pi} g_{mS} R_1 R_2}{[C_1(C_{\pi} + C_L) + C_{\pi} C_{\pi}] R_1 R_2} = \frac{-g_{mS} C_{\pi}}{C_{\pi}(C_1 + C_L) + C_1 C_L}$$

Note: for large $C_{\pi} \rightarrow p_H \approx -\frac{g_{mS}}{C_1 + C_L}$

p_H makes sense because if C_{π} is short:



$$f_H = \frac{1}{2\pi (C_1 + C_L) [R_1 // R_2 // \frac{1}{g_{mS}}]} \approx \frac{g_{mS}}{2\pi (C_1 + C_L)}$$

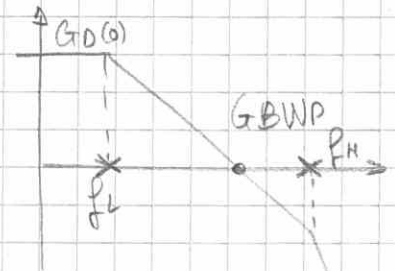


for $C_{\pi} \rightarrow 0$ only C_1, C_L take place in the circuit

for $C_{\pi} \rightarrow \infty$ $f_L \rightarrow 0$ and f_H saturates to

$$\frac{g_{mS}}{2\pi (C_1 + C_L)}$$

$$\underline{GBWP} = G_D(0) f_L = \frac{g_{m1} g_{mS} R_1 R_2}{2\pi C_{\pi} g_{mS} R_1 R_2} = \frac{g_{m1}}{2\pi C_{\pi}}$$



In order not to cut with -40 dB/dec :

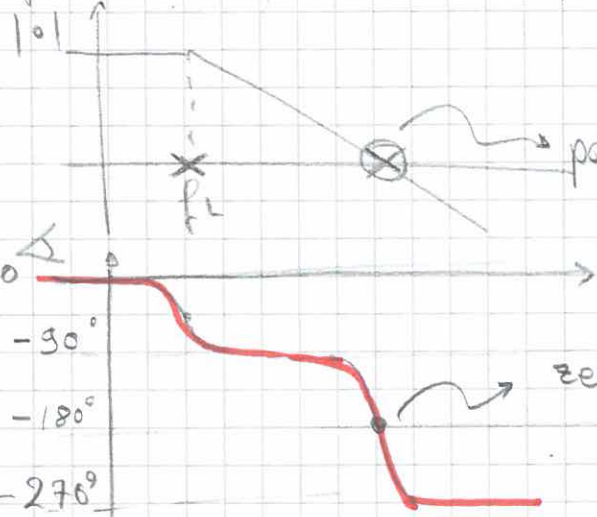
$$f_H \geq GBWP \quad \frac{g_{mS}}{2\pi (C_1 + C_L)} \geq \frac{g_{m1}}{2\pi C_{\pi}}$$

Therefore compensation should be $C_{\pi} \geq \frac{g_{m1}}{g_{mS}} (C_1 + C_L)$

If we need more phase margin, $f_H \geq 2 GBWP$

Phase margin analysis

f_L already contributed with 90°



$$f_H = \frac{g_{m5}}{2\pi(C_1 + C_2)}$$

$$GBWP = \frac{g_{m1}}{2\pi C_{in}}$$

$$f_z = \frac{g_{m5}}{2\pi C_{in}}$$

zero is RHP \rightarrow phase behaves like a pole $\rightarrow \phi_{\pi} = 0^\circ$

If this OTA is used like a buffer \rightarrow very close to instability

To recover some margin, $f_H, f_z = 2 GBWP$ so that

$$\phi_{\pi} = 180^\circ - 90^\circ - \underbrace{tg^{-1}\left(\frac{GBWP}{f_z}\right)}_{\frac{1}{2}} - \underbrace{tg^{-1}\left(\frac{GBWP}{f_H}\right)}_{\frac{1}{2}} = 90^\circ - 27^\circ - 27^\circ = \boxed{35^\circ}$$

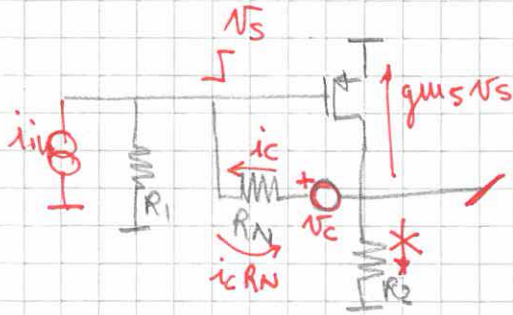
To obtain this we can: $\uparrow C_{in}$ or $\uparrow g_{m5}$:

- 1) Increasing C_{in} shifts GBWP to the left but also f_L is shifted \rightarrow BW
- 2) Increasing g_{m5} shifts f_H, f_z to the right at the cost of more power burned

6) Nulling resistor: compensation + implementation

We may think of a solution that changes the f_z independently or moves it to LHP ($+90^\circ$ contribution).

We need to place something in the path of the zero:



$$v_s = v_c - i_c R_N \quad g_m v_s = -i_c$$

$$R_{out}^{(0)} = \frac{v_c}{i_c} = R_N - \frac{1}{g_m} \rightarrow \text{new } \underline{f_z}$$

We can see that we can set

$R_N = \frac{1}{g_m}$ to have $f_z = \infty$, or we can move f_z

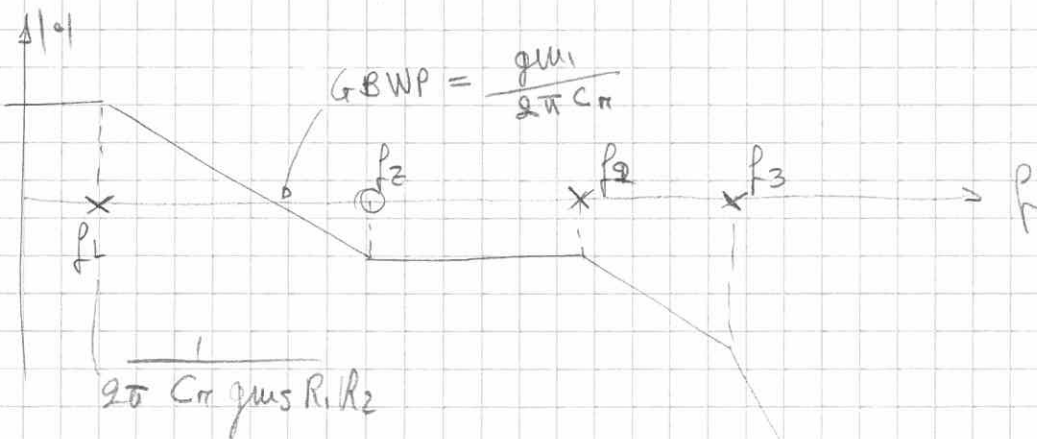
to the LHP so that we can have proper ϕ_m

Note: Caps are now independent \rightarrow additional pole.

$$\underline{P_{FL}} \sim \frac{1}{2\pi} \frac{1}{C_1 R_1 + C_L R_2 + C_{\pi} [R_1 + R_2 + g_m R_1 R_2 + R_N]}$$

It's easy to derive the new b_1 term

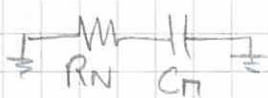
R_N is usually $\sim 2/g_m \rightarrow f_L$ has negligible change and so does GBWP



Let's compute f_3 :



C_L, C_{π} short



C_1, C_L short



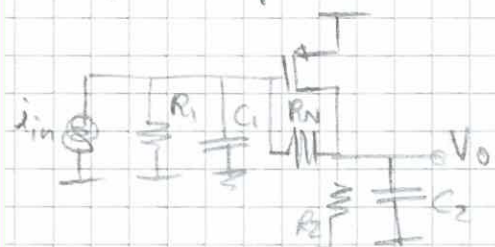
C_1, C_{π} short

$$f_3 = \sum_i \frac{1}{\tau_{i,\infty}} = \frac{1}{2\pi} \left[\frac{1}{C_1 (R_1 // R_N)} + \frac{1}{C_{\pi} R_N} + \frac{1}{C_L (R_2 // R_N)} \right]^{-1}$$

$$\sim \frac{1}{2\pi R_N (C_1 // C_{\pi} // C_L)} \sim \frac{1}{2\pi R_N C_1}$$

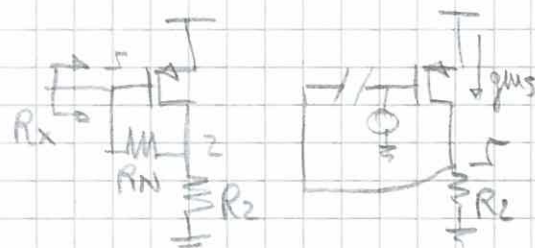
Note: it's not important to compute the precise value, hence this just gives us a hint on the order of magnitude

Compute f_2 : consider C_{π} as a short



$R_1^{(0)}: R_1 // R_x$

$R_x = \frac{R_2 + R_N}{1 - G_{loop}}$



$G_{loop} = -g_{m_s} R_2$ so $R_x \sim 1/g_{m_s}$ if $R_2 \gg R_N$ (which it is)

$R_L^{(0)}: R_L^{(0)} = R_2 // R_x$ where $R_x = \frac{1}{g_{m_s}}$

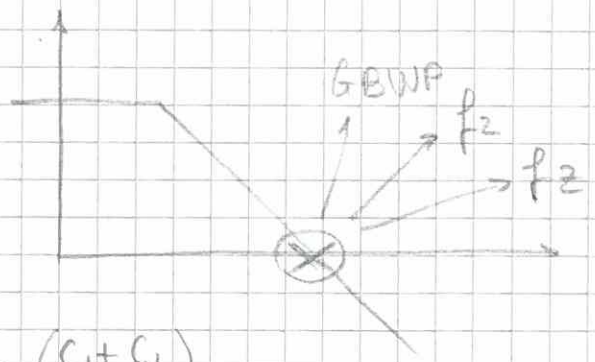
So $f_2 = \frac{1}{2\pi \left[\frac{C_1}{g_{m_s}} + \frac{C_L}{g_{m_s}} \right]} \sim \frac{g_{m_s}}{2\pi (C_1 + C_L)}$

f_2 is roughly in the same position as before

Zero-pole-compensation

First approach:

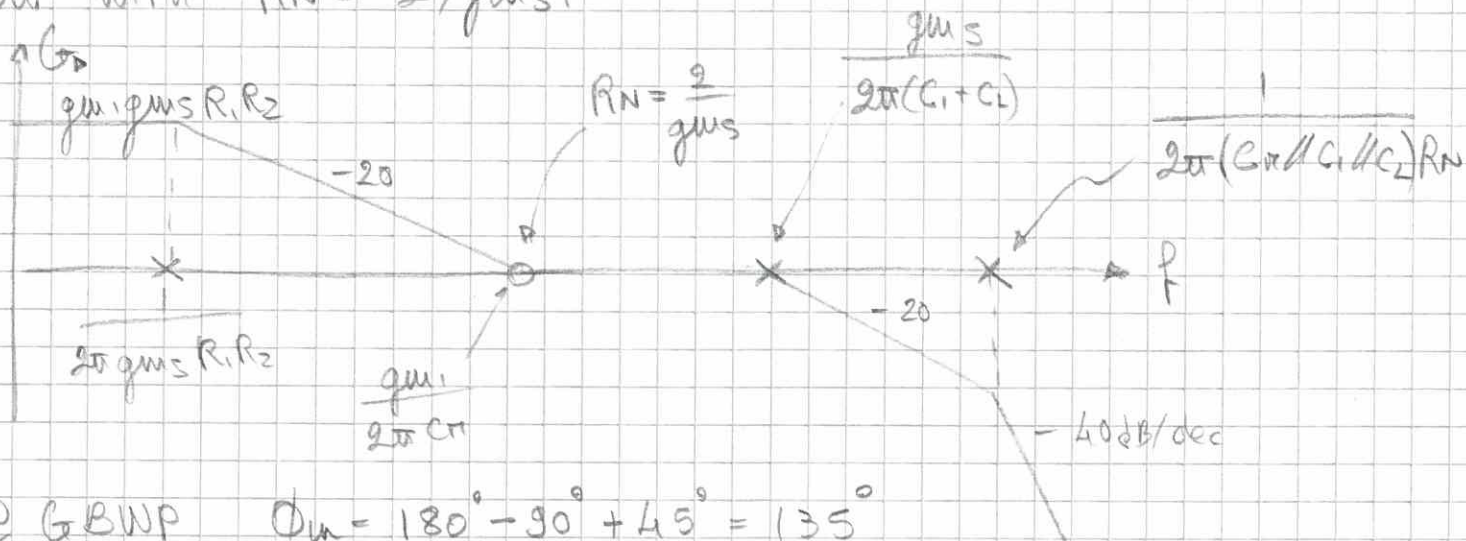
GBWP @ f_2 and $f_z = f_2$



$$\frac{g_{m1}}{2\pi C_{\pi}} = \frac{g_{m5}}{2\pi (C_1 + C_2)} \rightarrow C_{\pi} = \frac{g_{m1}}{g_{m5}} (C_1 + C_2)$$

$$\frac{1}{2\pi C_{\pi} (R_N - \frac{1}{g_{m5}})} = \frac{g_{m1}}{2\pi C_{\pi}} \rightarrow R_N \approx \frac{1}{g_{m5}} \left(1 + \frac{g_{m5}}{g_{m1}} \right)$$

Now with $R_N = 2/g_{m5}$:



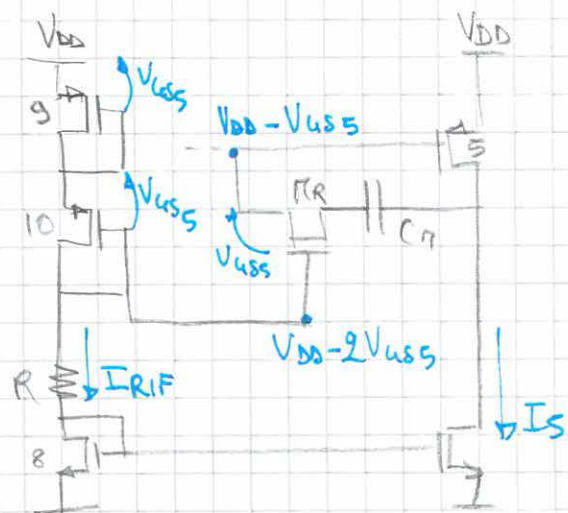
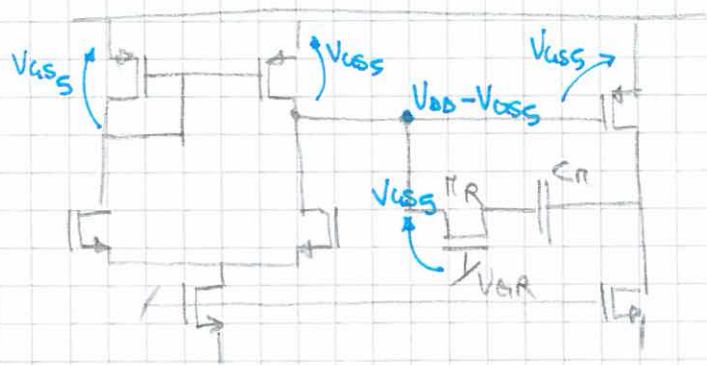
@ GBWP $\phi_m = 180^\circ - 90^\circ + 45^\circ = 135^\circ$

This way we can have the same bias, the same BW at the cost of implementing another resistor

Recap:

	GBWP	f_z	f_2	f_3
<u>Miller</u>	$\frac{g_{m1}}{2\pi C_{\pi}}$	$\frac{g_{m5}}{2\pi C_{\pi}}$ RHP	$\frac{g_{m5}}{2\pi (C_1 + C_2)}$	∞
<u>R_N</u>	$\frac{g_{m1}}{2\pi C_{\pi}}$	$\frac{1}{2\pi [R_N - \frac{1}{g_{m5}}] C_{\pi}}$	$\frac{g_{m5}}{2\pi (C_1 + C_2)}$	$> \frac{1}{2\pi C_1 R_N}$

RN implementation



M_R is a MOSFET in ohmic region

$$I_D = 2\mu' \left(\frac{W}{L}\right) \left(V_{ov} V_{ds} - \frac{V_{ds}^2}{2}\right) \quad g_{oR} = \left. \frac{\partial I_D}{\partial V_{ds}} \right|_{V_{ds}=0} = 2\mu' \left(\frac{W}{L}\right) (V_{ov} - V_{ds}) =$$

$$g_{oR} = 2\mu' \left(\frac{W}{L}\right)_R V_{ovR} = g_{mR}$$

Assume $R_N = \frac{1}{g_{mS}} (1 + \alpha)$ where α is a variability of the process

$$\text{Then } \frac{g_{mS}}{g_{mR}} = (1 + \alpha) \rightarrow \frac{\left(\frac{W}{L}\right)_S V_{ovS}}{\left(\frac{W}{L}\right)_R V_{ovR}} = (1 + \alpha)$$

We can have $V_{ovR} = V_{ovS}$ so we can better get $\frac{W}{L}$ ratios.

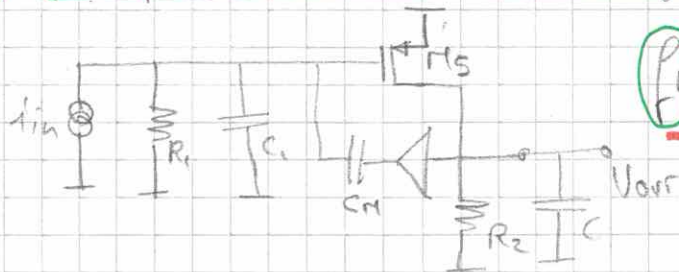
We can size $I_{RIF} = \frac{1}{m} I_S$ so we can have lower power dissip.

To have the V_{css} listed on 9, 10, we need $\left(\frac{W}{L}\right)_{9,10} = \frac{1}{m} \left(\frac{W}{L}\right)_S$

$$\text{While } \left(\frac{W}{L}\right)_8 = \frac{1}{m} \left(\frac{W}{L}\right)_6$$

Note: M_R is in ohmic region because at bias C is open and therefore $V_{ds} = 0$ because no current flows during bias

7) Compensation with voltage/current buffers (ideal/real)

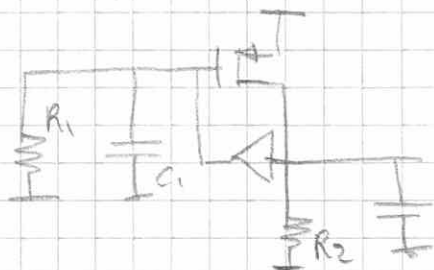


$$f_L = \frac{1}{2\pi C_{in} g_{m5} R_1 R_2} \quad \text{GBWP} = \frac{g_{m1}}{2\pi C_{in}}$$

IDEAL Voltage buffer

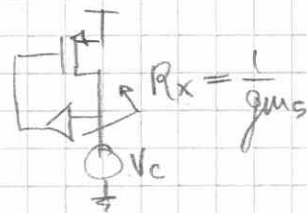
$f_3 = \infty$ with ideal voltage buffer (it can be verified by shorting two caps and seeing that $R_{eq}^{(\infty)} = 0$)

$f_2 = \infty$ say that C_{in} is basically short:



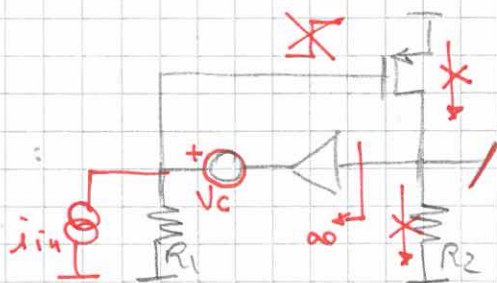
$R_1^{(0)} = 0$ because of the ideal buffer

$$R_L^{(0)} = R_2 \parallel R_x \approx \frac{1}{g_{m5}}$$



$$f_2 \approx \frac{g_{m5}}{2\pi C_L}$$

Let's see f_z :

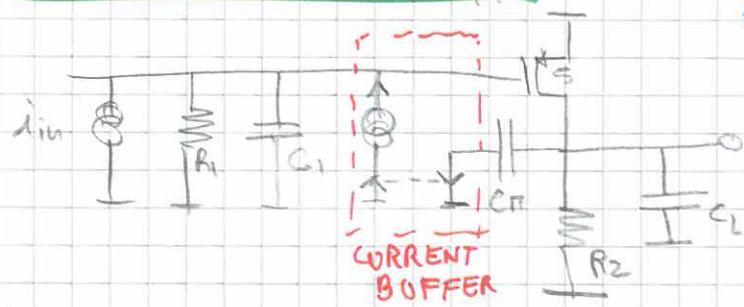


V_{out} can't move, $I_{R2} = 0$
 Therefore $I_{R1} = 0$
 This way $V_{ass} = 0$ so
 $V_c = 0 \rightarrow f_z = +\infty$

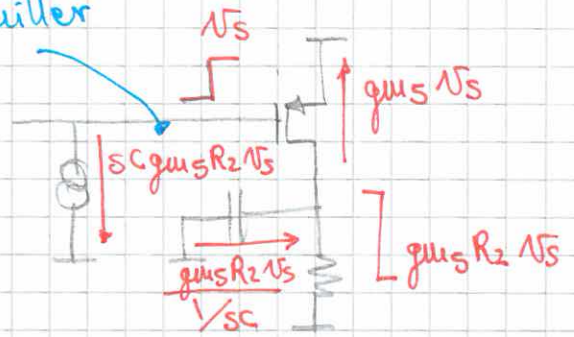
Recap

	GBWP	f_z	f_2	f_3
Voltage Buffer	$\frac{g_{m1}}{2\pi C_{in}}$	∞	$\frac{g_{m5}}{2\pi C_L}$	∞
IDEAL				

Ideal current buffer

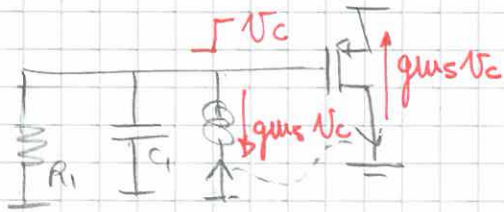


Usual miller effect



As always we can expect the same f_L , GBWP because we preserved the Miller effect

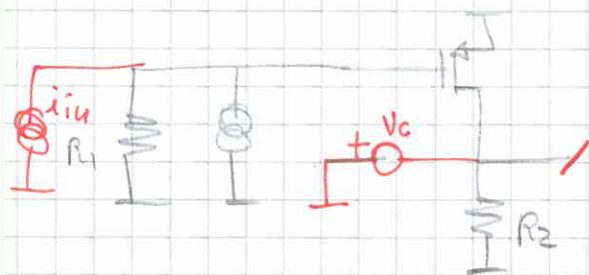
HF pole: C_{in} is short $\rightarrow R_2$ and C_L are shorted



$$R_1^{(b)} = R_1 \parallel \frac{1}{g_{m5}} \approx \frac{1}{g_{m5}}$$

$$f_2 = \frac{g_{m5}}{2\pi C_1}$$

Important: f_2 is independent from $C_L \rightarrow$ large loads do not affect frequency response!

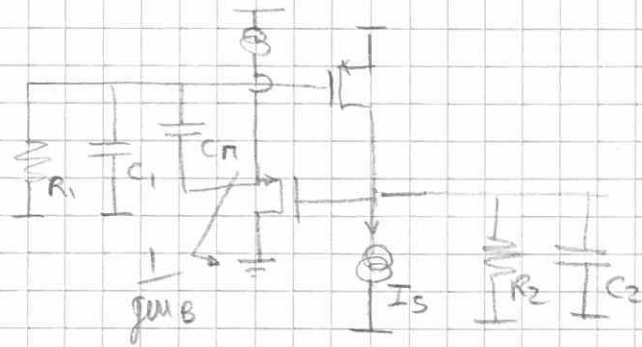


$f_z = \infty$ because v_c sees 0V on both its ends

	GBWP	f_z	f_2	f_3
Current Buffer	$\frac{g_{m1}}{2\pi C_{in}}$	∞	$\frac{g_{m5}}{2\pi C_1}$	∞
IDEAL				

f_3 can be easily derived by shorting the other two capacitors, we will see that result is the same of f_2

Real voltage buffer implementation

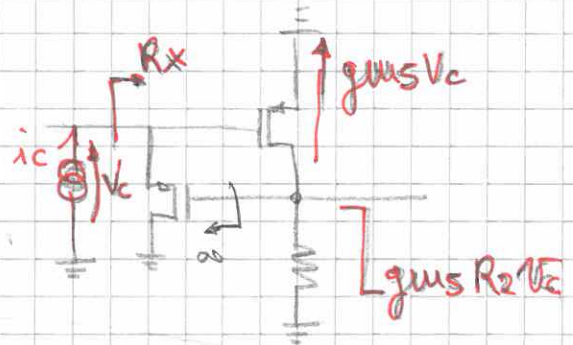


We now have a zero:

$$f_z = \frac{g_{mB}}{2\pi C_n}$$

f_2 : C_n is short

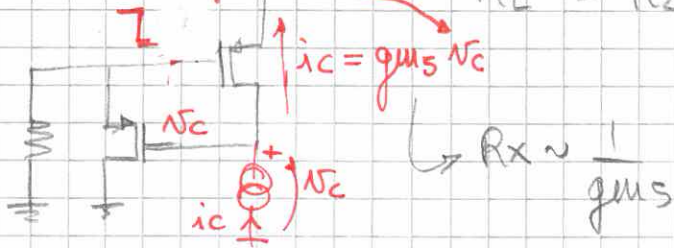
$$f_2 = \frac{1}{2\pi [C_1 R_1^{(0)} + C_L R_L^{(0)}]}$$



$$R_1^{(0)} = R_1 \parallel R_x \quad i_c = g_{mB} (V_c + g_{mS} R_2 V_c)$$

$$\frac{V_c}{i_c} = R_x = \frac{1}{g_{mB} (1 + g_{mS} R_2)} = \frac{1}{g_{mB} g_{mS} R_2} \approx R_1^{(0)} \quad (R_1 \text{ is neglig.})$$

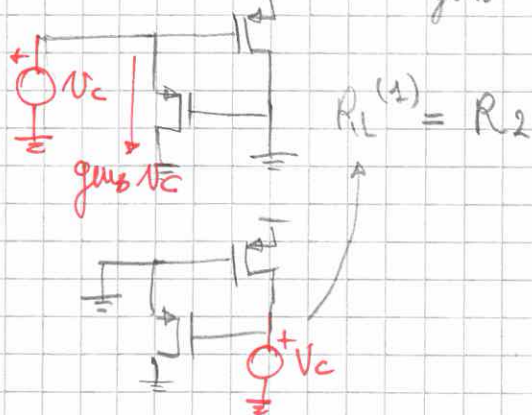
$$R_L^{(0)} = R_2 \parallel R_x \approx \frac{1}{g_{mS}}$$



$$f_z \approx \frac{g_{mS}}{2\pi C_L} \quad \text{since} \quad \frac{1}{g_{mB} g_{mS} R_2} \ll \frac{1}{g_{mS}}$$

$$f_3: \quad R_1^{(L)} = \frac{1}{g_{mB}} \parallel R_1 \approx \frac{1}{g_{mB}}$$

$$f_3 = \frac{1}{2\pi} \left[\frac{g_{mB}}{C_1} + \frac{1}{C_L R_2} \right] = \frac{g_{mB}}{2\pi C_1}$$

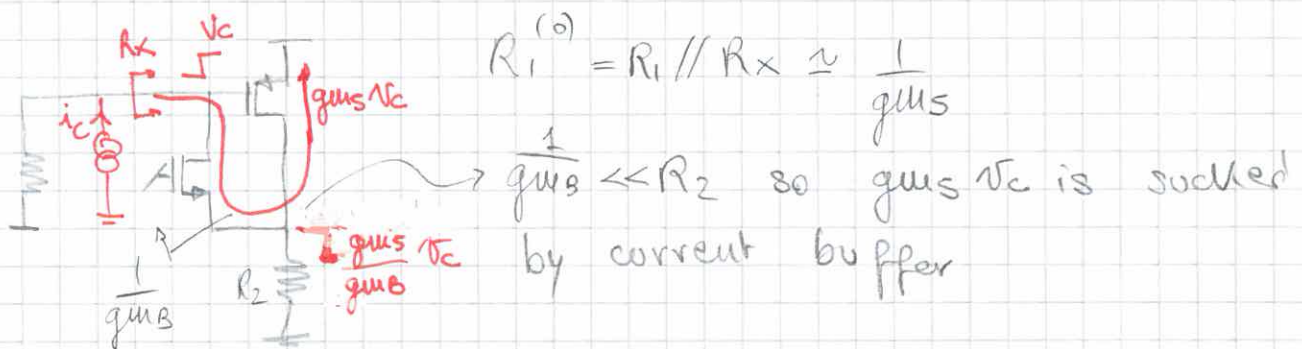
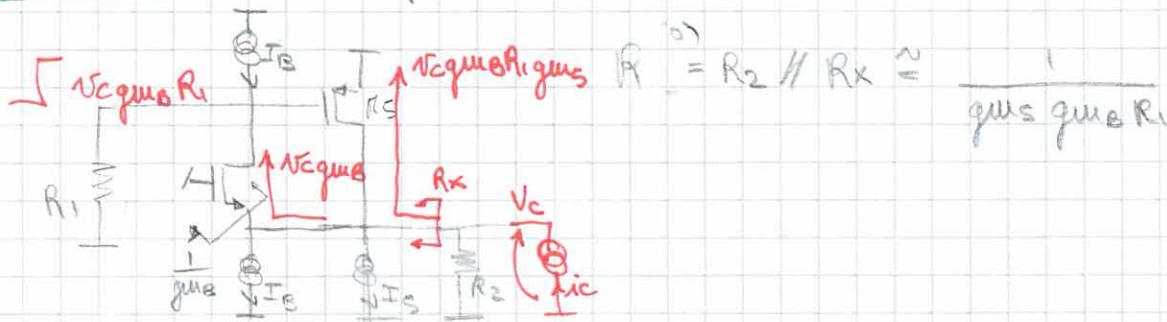


	GBWP	f_z	f_2	f_3
Voltage buffer	g_{m1}	$\frac{g_{mB}}{2\pi C_1}$	$\frac{g_{mS}}{2\pi C_L}$	$\frac{g_{mB}}{2\pi C_1}$
Real				

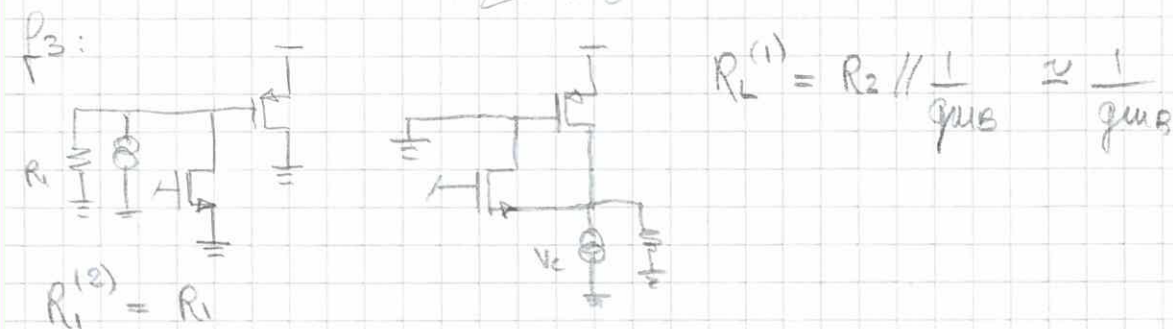
Since $C_L \gg C_1$, $f_3 > f_2$ (typically), better estimate would be using the full time constant method

We typically put $g_{mB} = g_{m1} \rightarrow$ We basically have the same result of a R_N comp but at the cost of more power consumption \rightarrow $F_{o11} = \frac{GBWP \cdot C_L}{I_{TOT}}$ decreases

Real current buffer

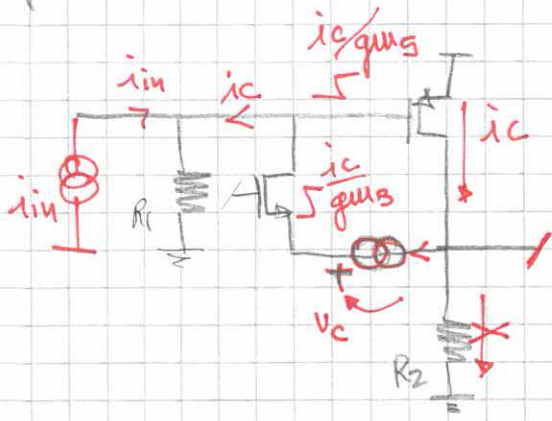


$$\underline{f_2} = \frac{1}{2\pi} \left[\frac{C_1}{g_{mS}} + \frac{C_L}{g_{mS} g_{mB} R_1} \right]^{-1} \approx \underline{\frac{g_{mS}}{2\pi C_1}}$$



$$\underline{f_3} = \frac{1}{2\pi} \left[\frac{1}{C_1 R_1} + \frac{g_{mB}}{C_2} \right] \approx \underline{\frac{g_{mB}}{2\pi C_2}}$$

f_z :



V_{out} can't move $\rightarrow V_{R2} = 0 \rightarrow I_{R2} = 0$
 $i_c \neq 0$ so it must be collected by M_5 (thus i_c / g_{m5} at the gate).

All i_c flows through g_{m5} so:

$$V_c = \frac{i_c}{g_{m5}} - 0 \rightarrow R_{out}^{(0)} = \frac{1}{g_{m5}}$$

$f_z = \frac{g_{m5}}{2\pi C_1}$

	GBWP	f_z	f_2	f_3
Current buffer	$\frac{g_{m1}}{2\pi C_1}$	$\frac{g_{m5}}{2\pi C_1}$	$\frac{g_{m5}}{2\pi C_1}$	$\frac{g_{m5}}{2\pi C_L}$
Real				

WARNING! If $g_{m5} \sim g_{m5}$, $C_L > C_1$, then $f_3 > f_2 \rightarrow$

This means that the method we used can't work \rightarrow
 We need the full Middlebrook theorem.

$$T(s) = G_D(0) \frac{1}{b_2 s^2 + b_1 s + 1} \rightarrow T(s) = 1$$

$$b_1 = C_1 R_1^{(0)} + C_L R_2^{(0)} = \frac{C_1}{g_{m5}} + \frac{C_L}{g_{m5} g_{m5} R_1}$$

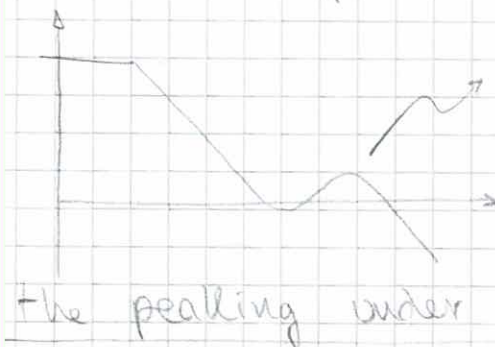
$$b_2 = C_1 C_L R_1^{(0)} R_2^{(1)} = \frac{C_1 \cdot C_L}{g_{m5} \cdot g_{m5}}$$

$$s^2 \frac{C_1 C_L}{g_{m5} g_{m5}} + s \left(\frac{C_1}{g_{m5}} + \frac{C_L}{g_{m5} g_{m5} R_1} \right) + 1 = 0$$

$\frac{1}{\omega_0^2}$ $\frac{1}{\omega_0 Q}$

$\omega_0 = \sqrt{\frac{g_{m5} g_{m5}}{C_1 C_L}}$ and since $\frac{C_1}{g_{m5}} \gg \frac{C_L}{g_{m5} g_{m5} R_1} \rightarrow Q \approx \sqrt{\frac{g_{m5} C_L}{g_{m5} C_1}}$

Note: be careful when placing the zeros

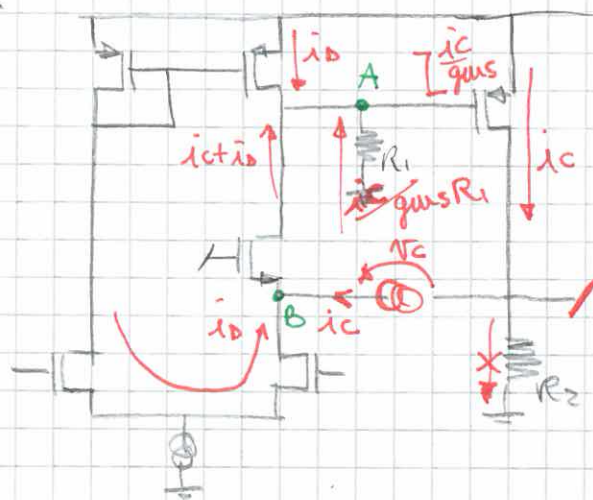


Peaking of the pole pair can go above 0dB. We may want to shift f_z right in order to keep the peaking under 0dB

Current buffer in the differential structure

Calculating the zeros

f_z :



Out circuit mode $\rightarrow V_{R2} = 0$ and
 $I_{RS} = i_c \rightarrow \sqrt{g_{mS}} = i_c / g_{mS}$

Balance current at nodes A, B:

$$\begin{cases} \frac{i_c}{g_{mS} R_1} = -i_c - 2i_d \\ \frac{V_c}{g_{mB}} = i_c + i_d \end{cases}$$

$$\frac{i_c}{g_{mS} R_1} + 2 \frac{V_c}{g_{mB}} = i_c \rightarrow \frac{V_c}{i_c} = R_{O1}^{(0)} = \frac{g_{mB}}{2} \left(1 - \frac{1}{g_{mS} R_1} \right)$$

$a_1 = C_{\pi} R_{O1}^{(0)} \rightarrow C_1, C_2$ do not introduce zeros \rightarrow

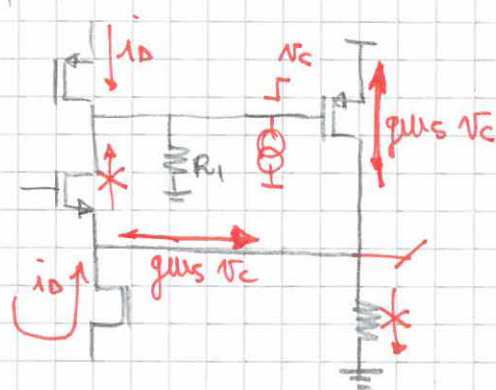
$a_2 = C_1 C_{\pi} R_{O1}^{(0)} R_{O1}^{(1)} : R_{O1}^{(0)} = 0$

while $R_{O3}^{(1)} = R_{O1}^{(0)} \Big|_{R_1=0} = \frac{g_{mB}}{2} \left(1 - \frac{1}{0} \right) = -\infty$

$a_2 = 0 \cdot (-\infty) \rightarrow$ indefinite form \rightarrow shuffle terms

$a_2 = C_{\pi} C_1 R_{O1}^{(0)} R_{O1}^{(1)} \rightarrow R_{O1}^{(1)}$

$g_{mS} V_c + i_d = 0 \rightarrow i_d = \frac{V_c}{R_1} - i_c$
 $g_{mS} V_c + \frac{V_c}{R_1} = i_c \rightarrow R_{O1}^{(1)} \approx \frac{-1}{g_{mS}}$



$a_2 = C_{\pi} C_1 \frac{1}{g_{mB}} \frac{-1}{g_{mS}} = - \frac{C_{\pi} C_1}{g_{mB} g_{mS}}$

Poles of cascaded Ahuja compensation

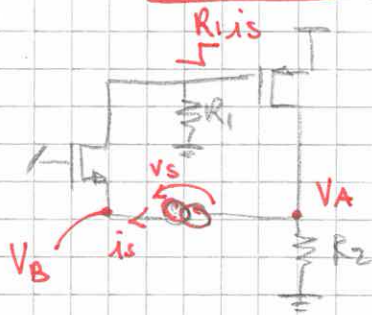
Since there are 3 independent capacitors, we have:

$$b_1 = C_1 R_1^{(0)} + C_L R_L^{(0)} + C_{\pi} R_{\pi}^{(0)}$$

$$b_2 = C_1 C_L R_1^{(0)} R_L^{(1)} + C_1 C_{\pi} R_1^{(0)} R_{\pi}^{(1)} + C_L C_{\pi} R_L^{(0)} R_{\pi}^{(1)}$$

$$b_3 = C_1 C_L C_{\pi} R_1^{(0)} R_L^{(1)} R_{\pi}^{(1)}$$

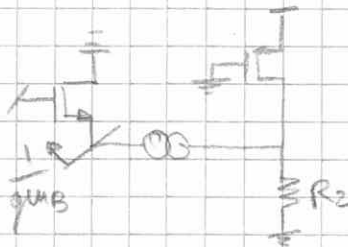
b_1 : $R_1^{(0)} = R_1$ $R_L^{(0)} = R_2$ as always, while $R_{\pi}^{(0)}$:



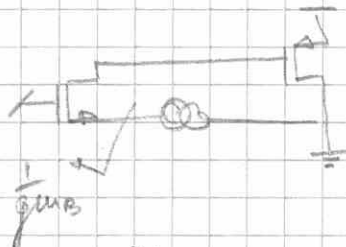
$$\underline{R_3^{(0)} = \frac{1}{g_{mB}} + R_2 + g_{mS} R_1 R_2}$$

$$\underline{b_1} = C_1 R_1 + C_L R_2 + C_{\pi} \left(\frac{1}{g_{mB}} + R_2 + g_{mS} R_1 R_2 \right) \approx \underline{C_{\pi} g_{mS} R_1 R_2}$$

b_2 :



$$R_{\pi}^{(1)} = \frac{1}{g_{mB}} + R_2$$



$$R_{\pi}^{(1)} = \frac{1}{g_{mB}}$$

$$\underline{b_2} \approx C_1 C_L R_1 R_2 + C_1 C_{\pi} R_1 \left(R_2 + \frac{1}{g_{mB}} \right) + C_L C_{\pi} \frac{R_2}{g_{mB}}$$

→ NEGLIGIBLE

$$= \underline{R_1 R_2 (C_1 C_L + C_1 C_{\pi}) + C_L C_{\pi} \frac{R_2}{g_{mB}}}$$

b_3 : $R_{\pi}^{(1,L)}$ is exactly like $R_{\pi}^{(1)} = \frac{1}{g_{mB}}$

$R_L^{(1)} = R_2$ intuitive

$$\underline{b_3} = \underline{C_1 C_L C_{\pi} \frac{R_1 R_2}{g_{mB}}}$$

Complete Ahuja transfer function

$$T(s) = g_{m1} g_{m2} R_1 R_2 \frac{a_2 s^2 + a_1 s + 1}{b_3 s^3 + b_2 s^2 + b_1 s + 1}$$

$$b_1 \approx C_{\pi} g_{m2} R_1 R_2$$

$$b_2 \approx R_1 R_2 (C_1 C_L + C_1 C_{\pi}) + C_L C_{\pi} \frac{R_2}{g_{mB}}$$

$$b_3 = C_1 C_L C_{\pi} \frac{R_1 R_2}{g_{mB}}$$

$$a_1 = C_{\pi} \frac{g_{mB}}{2} \left(1 - \frac{1}{g_{m2} R_1} \right) \approx \frac{g_{mB}}{2}$$

$$a_2 = -\frac{C_{\pi} C_1}{g_{m2} g_{mB}}$$

probably not asked at the oral

For large $g_{m2} R_2 C_{\pi}$ (read lectures pdf) + some algebra:

$$\text{Denominator: } \left[s^2 \frac{C_1 C_L}{g_{mB} g_{m2}} + s \frac{C_1}{g_{m2}} \left(\frac{C_{\pi} + C_L}{C_{\pi}} \right) + 1 \right] \left(\frac{s}{g_{m2} R_1 R_2 C_{\pi}} + 1 \right)$$

$$\text{So: } f_L = \frac{1}{2\pi C_{\pi} g_{m2} R_1 R_2}$$

$$\text{additional poles at } \omega_0 = \sqrt{\frac{g_{m2} g_{mB}}{C_1 C_L}} \quad Q = \frac{C_{\pi}}{C_{\pi} + C_L} \sqrt{\frac{g_{m2} C_2}{g_{mB} C_L}}$$

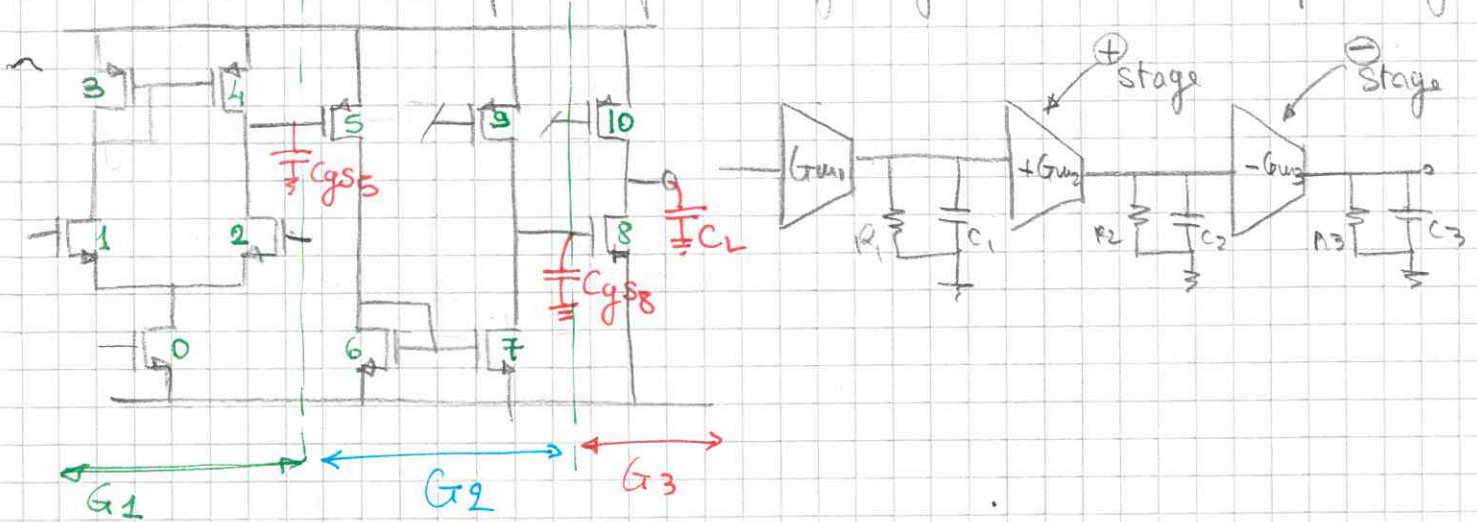
$$\text{Zeros: } s_{1,2} = \frac{g_{m2}}{2 C_1} \left[1 \pm \sqrt{1 + 8 \frac{g_{mB} C_1}{g_{m2} C_{\pi}}} \right]$$

$$\text{for large } C_L \quad z_1 = -2 \frac{g_{mB}}{C_{\pi}} \quad z_2 = \frac{g_{m2}}{C_1}$$

z_1 sits close to GBWP, z_2 is around a f_T because C_1 is typically a C_{gs}

8) Nested Miller compensation

As technology scales $\rightarrow V_{DD} \searrow \rightarrow$ We can't pile up transistors, Therefore for large gains \rightarrow multiple stages



$$|G_{DC}| = G_{m1} R_1 G_{m2} R_2 G_{m3} R_3 = (200)^3 = 138 \text{ dB}$$

C_L, C_{gs5}, C_{gs8} load the OTA the most (high impedance nodes)

low not to have high power burned

high to drive capacitive loads

critical for input noise

G_{m1}	100 μ A
R_1	2 $\text{M}\Omega$
C_1	10 pF

$$f_1 = 8 \text{ MHz}$$

G_{m2}	30 μ A/V
R_2	6.7 $\text{M}\Omega$
C_2	167 pF

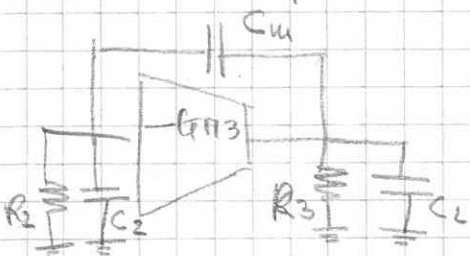
$$f_2 = 142 \text{ kHz}$$

G_{m3}	1 mA/V
R_3	200 $\mu\Omega$
C_1	100 pF

$$f_3 = 200 \text{ Hz}$$

With this configuration \rightarrow cut 0 dB with $-60 \text{ dB/dec} \rightarrow$ NOPE

Miller compensation for C_2, C_1 (classic)



$$GBWP_{23} = \frac{G_{m2} G_{m3} R_2 R_3}{2\pi C_m G_{m3} R_2 R_3} = \frac{G_{m2}}{2\pi C_m}$$

$$f_3' = \frac{G_{m3}}{2\pi (C_L + C_2)} \approx \frac{G_{m3}}{2\pi C_L}$$

$$f_z' = \frac{G_{m3}}{2\pi C_m} = \text{positive zero}$$

$$f_2' = \frac{1}{2\pi C_m G_{m3} R_2 R_3}$$

Neglect the zero $\rightarrow \phi_m = 60^\circ$ if $f_3' > 2 GBWP$:

$$\frac{G_{m3}}{2\pi C_L} = 2 \frac{G_{m2}}{2\pi C_m} \rightarrow C_m = 2 \frac{G_{m2}}{G_{m3}} C_L$$

At this point, the whole OTA response will be

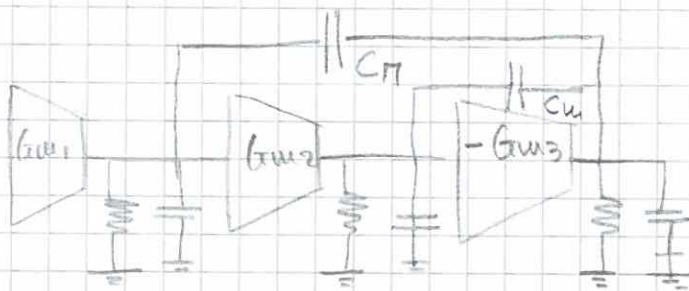
$$f_2' = \frac{1}{2\pi C_m G_{m3} R_2 R_3} = 20 \text{ Hz} \quad f_3' = 2 \text{ GBWP}_{23} = \frac{G_{m3}}{2\pi C_L} = 1,6 \text{ MHz}$$

$$f_1 = \frac{1}{2\pi C_1 R_1} = 8 \text{ MHz} \quad f_z' = 26,5 \text{ MHz}$$

$f_z' \gg \text{GBWP}$ we assumed its contribution is negligible.

We still cut with -40 dB/dec at $0 \text{ dB} \rightarrow \text{UNSTABLE}$

Nested Miller compensation

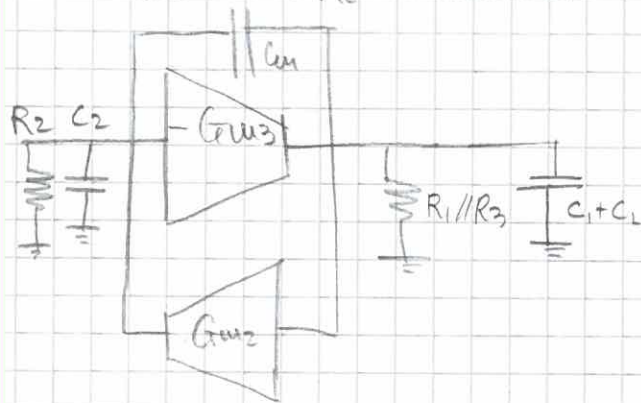
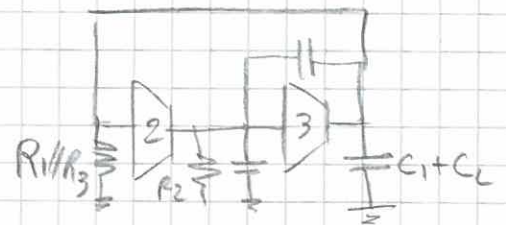


$\rightarrow G_{m2}$ has positive gain because we need overall negative gain and C_m in order to exploit Miller!

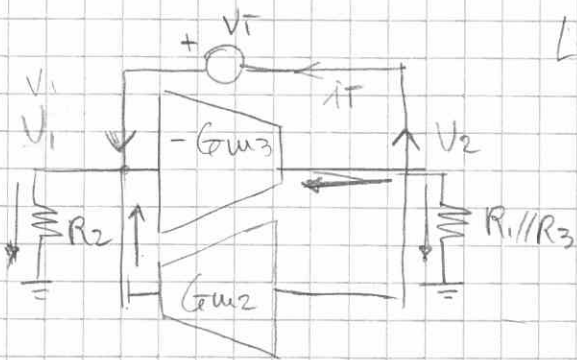
Since $G_2 = G_{m2} R_2$, $G_3 = G_{m3} R_3$ C_m sees $G_2 G_3$ while C_m only sees $G_3 \rightarrow$ lowest pole is given by C_m

$\text{GBWP}|_{\text{TOT}} = \frac{G_{m1}}{2\pi C_m} \rightarrow$ we can now short C_m (do not consider zeros for now)

$$f_2'' = \frac{1}{2\pi C_2 R_2^{(0)} + C_m R_m^{(0)} + (C_L + C_2) R_L^{(0)}}$$



Let's calculate $R_{in}^{(0)}$:



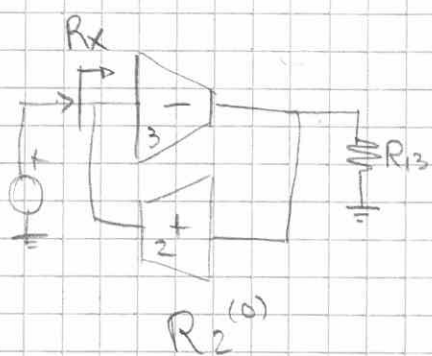
$$\frac{V_1}{R_2} = V_2 G_{m2} = I_T$$

$$\frac{V_2}{R_{13}} = V_1 G_{m3} = I_T$$

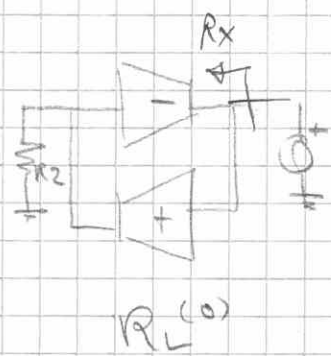
→ Neglect these

$$V_2 = -\frac{I_T}{G_{m2}} \quad V_1 = -\frac{I_T}{G_{m3}} \quad V_T = V_1 - V_2 = I_T \left(-\frac{1}{G_{m3}} + \frac{1}{G_{m2}} \right)$$

$$\underline{R_{in}^{(0)}} = \underline{\frac{G_{m3} - G_{m2}}{G_{m3} G_{m2}}}$$



$$\underline{R_z^{(0)}} = R_2 // R_x \approx \frac{1}{G_{m2} G_{m3} R_{13}}$$



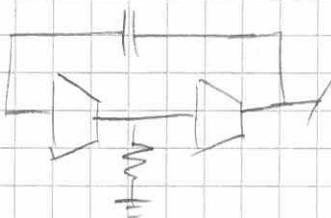
$$\underline{R_L^{(0)}} = R_{13} // R_x \approx \frac{1}{G_{m3} G_{m2} R_2}$$

$$f_2'' = \frac{1}{2\pi} \left[\frac{C_2}{G_{m2} G_{m3} R_{13}} + \frac{G_{m3} - G_{m2}}{G_{m2} G_{m3}} C_{in} + \frac{C_1 + C_L}{G_{m3} G_{m2} R_2} \right]^{-1} = 0,8 \text{ MHz}$$

As we did before, $f_2'' = 2 \text{ GBWP}$ so $\underline{C_{in}} = 2 \frac{G_{m1}}{2\pi f_2''} \approx 40 \text{ pF}$

$$\underline{\text{GBWP}} = \frac{G_{m1}}{2\pi C_{in}} = 400 \text{ kHz}$$

Estimate the zero:



$$f_z'' = \frac{G_{m2} R_2 G_{m3}}{2\pi C_z}$$

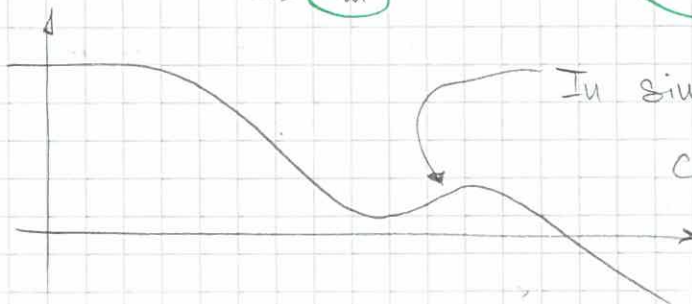
f_z'' is far away from GBWP. This though isn't totally correct, since C_{in} and C_{in} can contribute at most with two zeros (SPICE will tell)

Comments on TF

It is possible to derive the polynomial at the denominator:

$$s^2 b_2 + s b_1 + 1 = s^2 \frac{C_L C_M}{G_{M2} G_{M3}} + s \frac{G_{M3} - G_{M2}}{G_{M2} G_{M3}} + 1 = 0$$

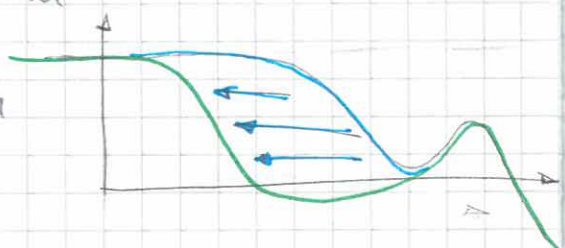
$$\omega_0 = \sqrt{\frac{G_{M2} G_{M3}}{C_L C_M}} \quad Q = \frac{1}{G_{M3} - G_{M2}} \sqrt{G_{M2} G_{M3} \frac{C_L}{C_M}}$$



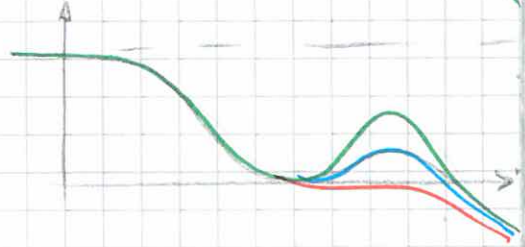
In simulations we see peaking. How can we solve this? (look at

G_{M3}, C_M)

• Increase $C_M \rightarrow$ BW is reduced but C_M plays no role in Q factor \rightarrow bad idea

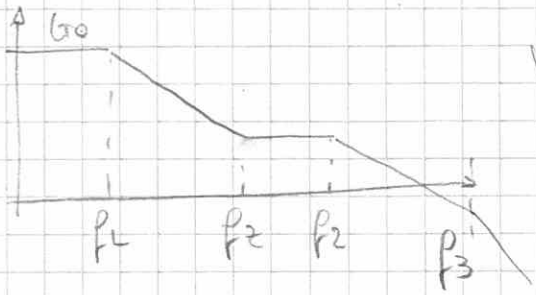


• Increase $G_{M3} \rightarrow$ peak is quenched down
Fine tuning is done on SPICE



9) OTA linear response. In band doublets + settling response

We can have something like: (Rn compensation)

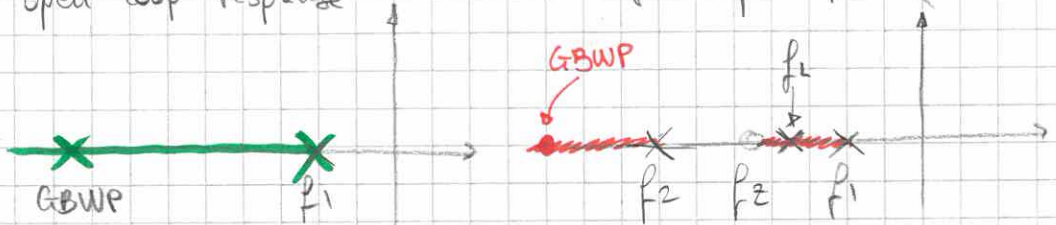
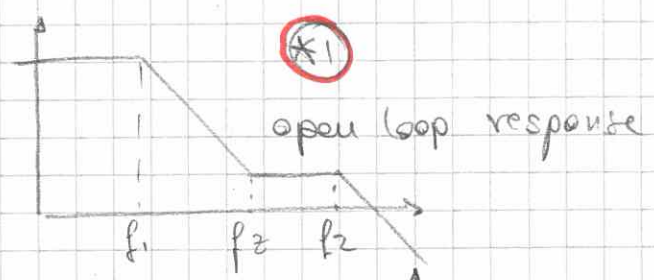
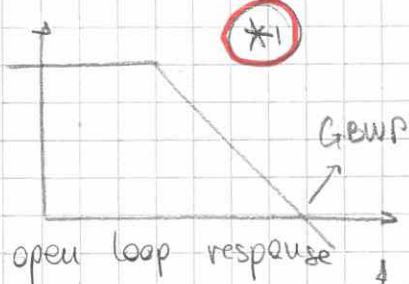
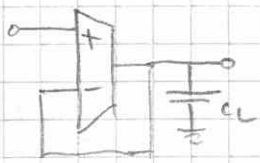


Where $f_z \approx \frac{g_{m5}}{2\pi C_L}$ $f_z = \frac{1}{2\pi (R_n - 1) C_n g_{m5}}$

$f_z - f_z'$ cancellation has negligible impact on GBWP and phase margin

This way, since $f_z \propto \frac{1}{C_L}$ we could think of driving large capacitive loads without degrading the amplifier stability.

There are however some drawbacks to this:



*1 Compensated plot where $f_z = f_z = \text{GBWP}$

*2 Same circuit used to drive larger C_L (f_z moves left) without major changes on compensation (f_z moves left too). If load isn't precise for zero-pole cancellation, we have *2

Analyze Gloop of *2:

$G_{loop}(s) = -G_0 \frac{1 + s\tilde{\tau}_z}{(1 + s\tilde{\tau}_1)(1 + s\tilde{\tau}_2)}$ by looking at the root locus:

f_L for a large G_0 approaches f_z

f_H will be $\sim \text{GBWP}$

Let's find the closed loop singularities:

$$G_{loop}(s) = 1 \quad G_0(1+s\hat{\tau}_z) = (1+s\hat{\tau}_1)(1+s\hat{\tau}_2)$$

$$s^2(\hat{\tau}_1\hat{\tau}_2) + s(\hat{\tau}_1 + \hat{\tau}_2 + G_0\hat{\tau}_z) + (G_0+1) = 0 \quad \left. \vphantom{s^2(\hat{\tau}_1\hat{\tau}_2)} \right\} \text{rough estimation:}$$

• for HF $s^2 \gg s \gg G_0+1$ $s^2(\hat{\tau}_1\hat{\tau}_2) + s(\hat{\tau}_1 + \hat{\tau}_2 + G_0\hat{\tau}_z) = 0$

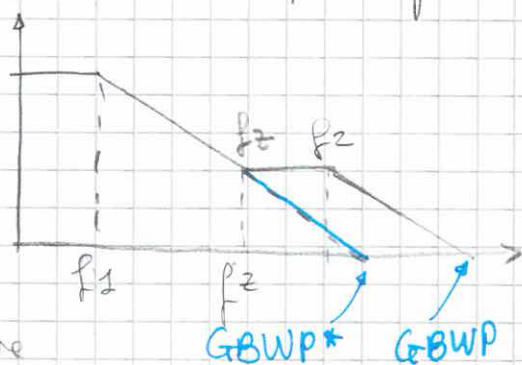
$$S_{HIGH} = - \frac{\hat{\tau}_1 + \hat{\tau}_2 + G_0\hat{\tau}_z}{\hat{\tau}_1\hat{\tau}_2} \approx \frac{G_0\hat{\tau}_z}{\hat{\tau}_1\hat{\tau}_2}$$

• for LF $s^2 \ll s \ll G_0+1$ $s(\hat{\tau}_1 + \hat{\tau}_2 + G_0\hat{\tau}_z) + G_0 + 1 = 0$

$$S_{LOW} = - \frac{G_0 + 1}{\hat{\tau}_1 + \hat{\tau}_2 + G_0\hat{\tau}_z} \rightarrow - \frac{1}{\hat{\tau}_z} \text{ flow approaches } f_z \text{ for high } G_0$$

Let's analyze this a little better:

$$p_{LOW} = \frac{\hat{\tau}_1 + \hat{\tau}_2 + G_0\hat{\tau}_z}{G_0 + 1} \approx \hat{\tau}_z + \frac{\hat{\tau}_1}{G_0}$$



It's interesting to note that $\frac{\hat{\tau}_z}{G_0}$ is the

GBWP*, the OdB cut if f_z wasn't there. In reality, the

$$GBWP \text{ is } GBWP = \frac{f_z}{\hat{\tau}_z} GBWP^*$$

We now ask ourselves what is the linear response:

$$\begin{array}{c} E/s \\ \downarrow \\ \boxed{T(s)} \\ \downarrow \\ V_{out} \end{array} \quad T(s) = \frac{1+s\hat{\tau}_z}{(1+s\hat{\tau}_L)(1+s\hat{\tau}_H)} \quad V_{out}(s) = \frac{E}{s} T(s)$$

$$\text{Use Heaviside } V_{out}(t) = \mathcal{L}^{-1} \left[\frac{E}{s} \left(\frac{A}{1+s\hat{\tau}_L} + \frac{B}{1+s\hat{\tau}_H} \right) \right]$$

Use limit theorem

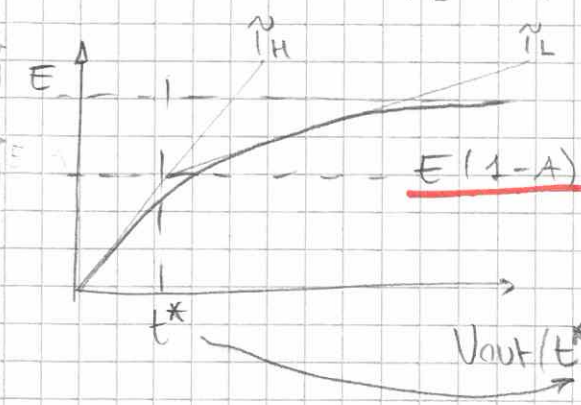
$$A = \lim_{s \rightarrow -\frac{1}{\hat{\tau}_L}} \frac{(1+s\hat{\tau}_z)(1+s\hat{\tau}_H)}{(1+s\hat{\tau}_L)(1+s\hat{\tau}_H)} = \frac{1 - \frac{1}{\hat{\tau}_L}(\hat{\tau}_z + \hat{\tau}_H)}{1 - \hat{\tau}_H/\hat{\tau}_L} = \frac{\hat{\tau}_L - \hat{\tau}_z - \hat{\tau}_H}{\hat{\tau}_L - \hat{\tau}_H}$$

$$B = \lim_{s \rightarrow -\frac{1}{\hat{\tau}_H}} \frac{(1+s\hat{\tau}_z)(1+s\hat{\tau}_L)}{(1+s\hat{\tau}_L)(1+s\hat{\tau}_H)} = \frac{\hat{\tau}_H - \hat{\tau}_z}{\hat{\tau}_H - \hat{\tau}_L} = \frac{\hat{\tau}_z - \hat{\tau}_H}{\hat{\tau}_L - \hat{\tau}_H}$$

Since $\hat{\tau}_z > \hat{\tau}_H$, we adjust to get $B > 0$

$$V_{out}(t) = E \left[A(1 - e^{-t/\tau_L}) + B(1 - e^{-t/\tau_H}) \right] = \underline{E[1 - Ae^{-t/\tau_L} - Be^{-t/\tau_H}]}$$

$$\underline{A+B} = \frac{\tau_L - \tau_z}{\tau_L - \tau_H} + \frac{\tau_z - \tau_H}{\tau_L - \tau_H} = \underline{1}$$



B term (τ_H) quickly vanishes, leading to $V_{out} \approx E[1 - Ae^{-t/\tau_L}]$

This means that the fast curve starts at a high

$$V_{out}(t^*) = E \left[1 - Ae^{-\frac{t^*}{\tau_L}} \right] = \underline{E(1-A)}$$

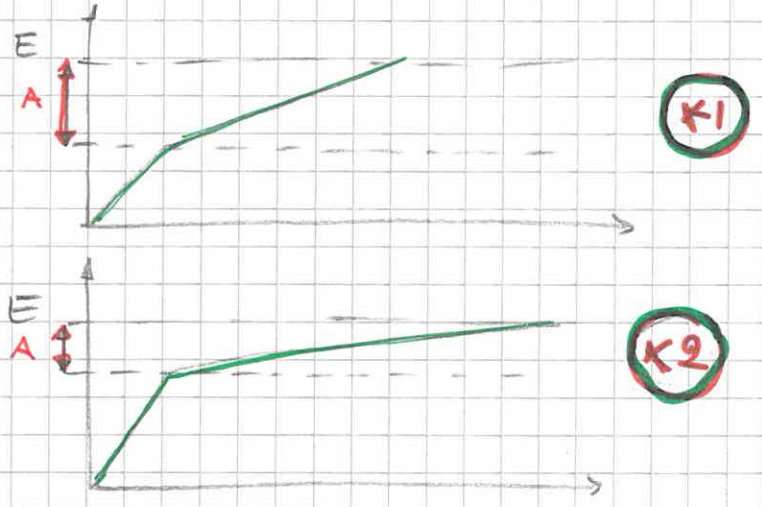
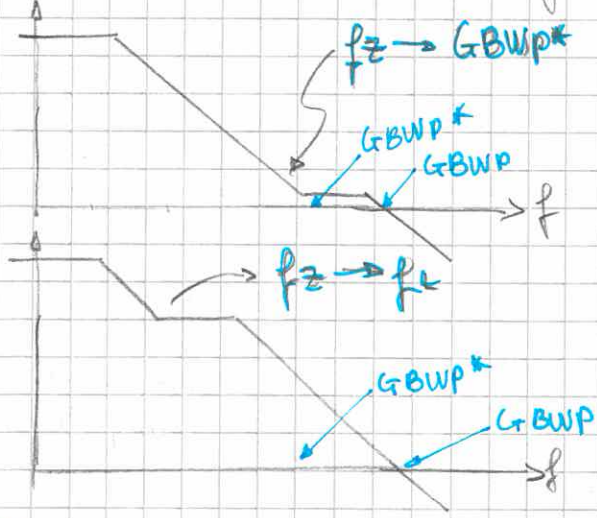
Where $A = \frac{\tau_L - \tau_z}{\tau_L - \tau_H} = \frac{\frac{\tau_z + \tau_1}{G_0} - \tau_z}{\frac{\tau_z + \tau_1}{G_0} - \tau_H} \rightarrow \underline{A} = \frac{\tau_1}{G_0 \tau_z} = \frac{f_z}{\underline{GBWP^*}}$ *3

Since $\tau_L \approx \tau_z + \frac{\tau_1}{G_0}$

Suppose $f_z = 5 \text{ MHz}$ $GBWP = 50 \text{ MHz}$ $f_z / GBWP^* = 10\%$

Therefore, we end up with $A \approx 0$ when $f_z \rightarrow f_L$
 $A \approx 1$ when $f_z \rightarrow GBWP^*$

These two limits generate two situations

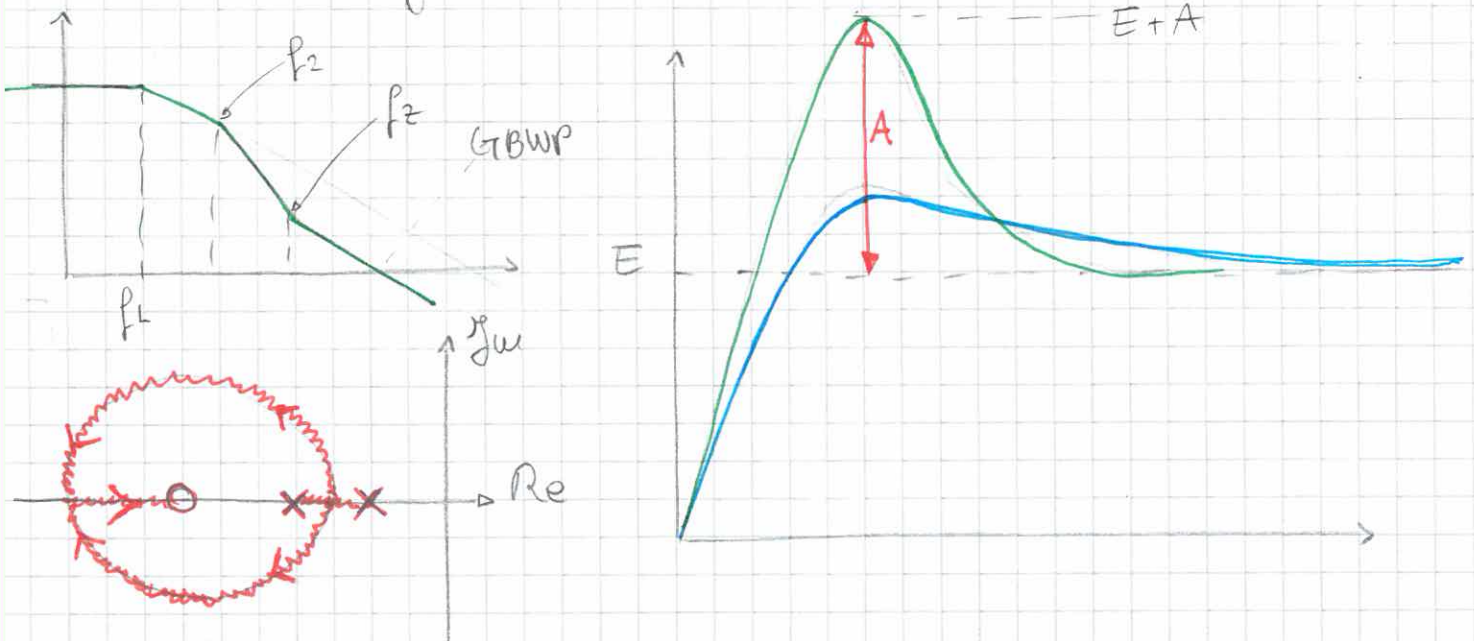


K1 f_z approaches $GBWP^* \rightarrow \tau_z \downarrow \rightarrow A \uparrow \uparrow$. $\tau_L = \tau_z + \frac{\tau_1}{G_0}$ is moderately slow

K2 f_z approaches $f_L \rightarrow \tau_z \uparrow \uparrow \rightarrow A \uparrow \uparrow$ and $\tau_L = \tau_z + \frac{\tau_1}{G_0}$ is

increased, therefore *2 settling time is higher than K1

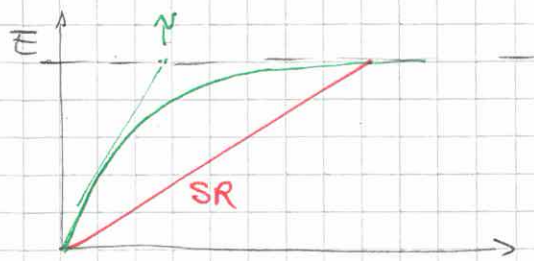
A similar thing happens when the zero is after f_2



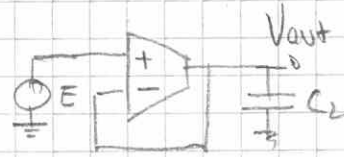
In this case, the issue will be $E+A$.

Design note: to get fast settling time we accept the trade off with large A to have low τ_L , therefore the design should foresee a $f_z \rightarrow GBWP^*$

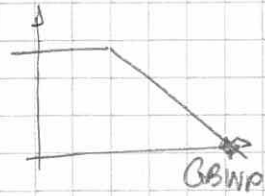
10) Slow Rate: impact on settling time, SR_{int} , SR_{ext} , class AB



Suppose we're driving a load using an OTA buffer:



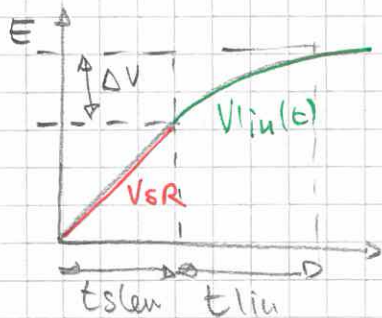
$$\tau = \frac{1}{GBWP}$$



OTA will be limited by τ (linear component) and the Slew Rate

$$V_{out}(t)|_{lin} = E(1 - e^{-\frac{t}{\tau}}) \rightarrow \frac{dV_{out}(t)}{dt} = \frac{E}{\tau}$$

If SR is slower, slopes will be $\frac{E}{\tau} > SR$. This means that we'll have a slow rate section + linear section:



There will be a condition where SR will become faster than the linear limitation set by the $GBWP \rightarrow$ slow rate becomes linear:

$$V_{lin}(t) = E - \Delta V e^{-\frac{t}{\tau}} \rightarrow \frac{dV_{lin}(t)}{dt} = \frac{\Delta V}{\tau}$$

Condition is $\frac{dV_{lin}(t)}{dt} = SR \rightarrow \frac{\Delta V}{\tau} = SR \rightarrow \underline{\Delta V = SR \cdot \tau}$

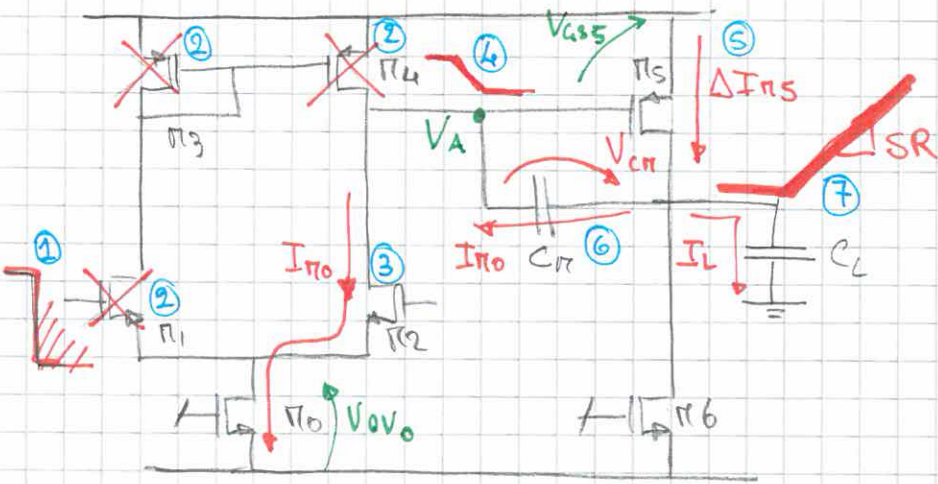
We can now compute $t_{slew} = \frac{E - \Delta V}{SR}$ and the t_{lin} will set when $V_{out}(t)$ reaches 99% of input step E :

$$V_{lin}(t) = E - \Delta V e^{-\frac{t_{lin}}{\tau}} = E - \frac{E}{100} \rightarrow \underline{t_{lin} = \tau \ln\left(\frac{\Delta V}{E} \cdot 100\right)}$$

Total settling time will be $t_{settling} = t_{slew} + t_{lin}$

Slew rate limitations of a two stage OTA

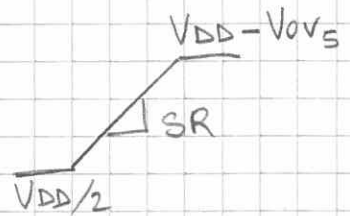
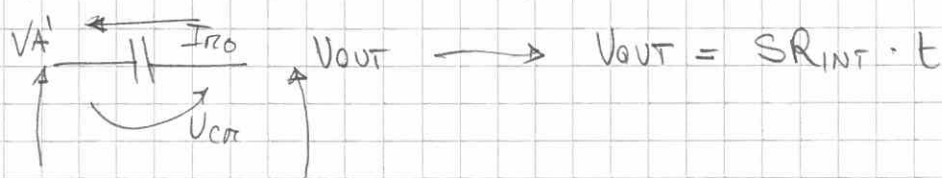
Positive SR



- 1) A large step is applied
 - 2) π_1, π_3, π_4 shut off because of ①
 - 3) All the generator current flows through ③ (suppose π_0 is still in saturation)
 - 4) The current I_{ro} is drained from node $V_A \rightarrow V_A$ decreases in voltage. \rightarrow neglect the presence of C_L
 - 5) (Don't consider C_L for the moment) V_A will generate a ΔI_S current. A stable condition would be when π_5 takes care of all the I_{ro} . This translates on its node by:

$$V_{ovs}' = \sqrt{\frac{I_{sbias} + I_{ro}}{K_5}}$$
 Note: $\Delta I_S = I_{ro}$ but $I_S|_{TOT} = I_{sbias} + \Delta I_S$
- Suppose $K = 100\mu$ $I_{ro} = 5\mu$ $I_{sbias} = 1\mu \rightarrow V_{ovs}' = 0,25V$
- Since $V_A|_{bias} \approx \frac{V_{DD}}{2} = 1,5V \rightarrow V_A' = 1,25V$
- This decrease poses no issue on the saturation of π_0, π_2 .
 Note: if π_2 goes ohmic, I_{ro} will still flow through node A. V_A' would have to decrease all the way down to V_{ovro} in order to change I_{ro} current. It does not happen
- 6) With point 5 C_{π} sees a constant (V_A) voltage on the left and a constant current I_{ro} . This will generate a ramp on its right pin that is $SR_{int} = \frac{I_{ro}}{C_{\pi}}$

Situation is now this one:



Since $V_{A'}$ is constant, the voltage ramp on C_{pi} defined as SR_{INT} will be the same exact ramp on V_{out} . Therefore, if we define SR_{EXT} as the ramp exhibited on the out, we can say $SR_{EXT} = SR_{INT}$

7) Now include C_L presence. V_{ovs} will increase more and $V_{A'}$ will decrease even more. At point 5 we will have a new (stable) condition that is $\Delta I_S = I_{ro} + I_L$

$V_{ovs} = \sqrt{\frac{I_{S5} + I_{ro} + I_L}{K_S}}$ The issue here will be when C_L is large enough to generate

a too high V_{ovs} that puts M_0 out of saturation.

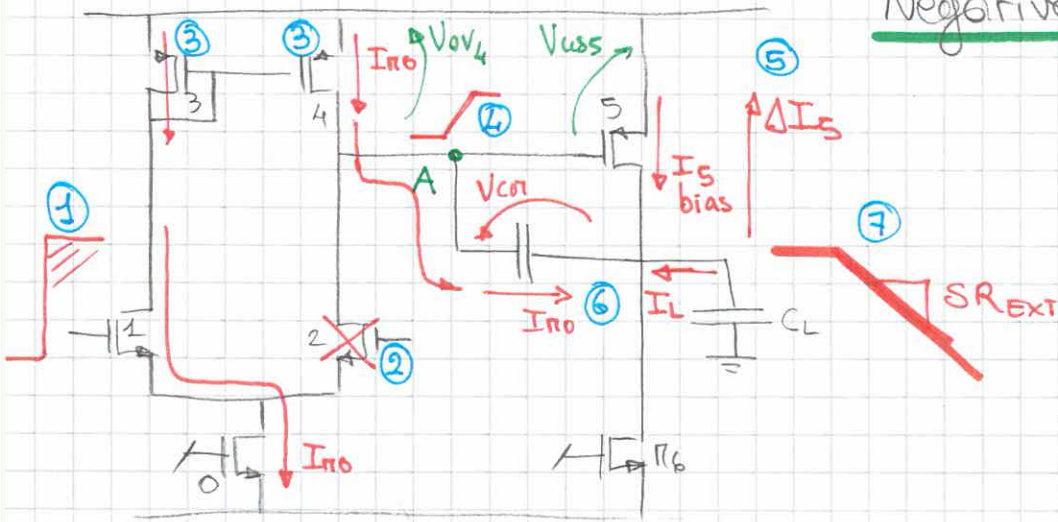
If this does not happen, M_0 can perfectly handle I_{ro} and I_L , therefore the voltage ramp will be set again by C_{pi} , leading to $SR_{EXT} = \frac{I_{ro}}{C_{pi}}$

The only thing is that we now have $I_L = SR_{EXT} \cdot C_L$ so $\Delta I_S = I_{ro} + SR_{EXT} C_L \stackrel{SR_{EXT} = SR_{INT}}{=} I_{ro} + SR_{INT} C_L$
 $SR_{EXT} = SR_{INT}$

Recap: SR_{EXT} | WITHOUT C_L = $SR_{INT} = \frac{I_{ro}}{C_{pi}}$

SR_{EXT} | WITH C_L = $SR_{INT} = \frac{I_{ro}}{C_{pi}}$ but keep an eye on M_0 saturation

Negative SR



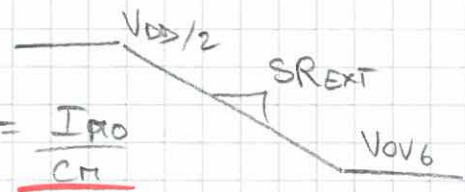
- 1) large positive is applied
- 2) M_2 shuts off and all I_{ro} flows through M_1
- 3) I_{ro} gets mirrored on M_4
- 4) I_{ro} flows into node (A) $\rightarrow V_A$ increases leading to a decrease on V_{ov5}

5) $V_{ov5} \rightarrow V_{ov5}$ decreases so it can accept ΔI_S .

Suppose now that $I_{sbias} > I_{ro}$ (Neglect C_L presence)

6) since $I_{sbias} > I_{ro}$, M_5 can handle I_{ro} and I_{ro} can flow through C_{cr}

7) same as before $SR_{EXT} = SR_{INT} = \frac{I_{ro}}{C_{cr}}$



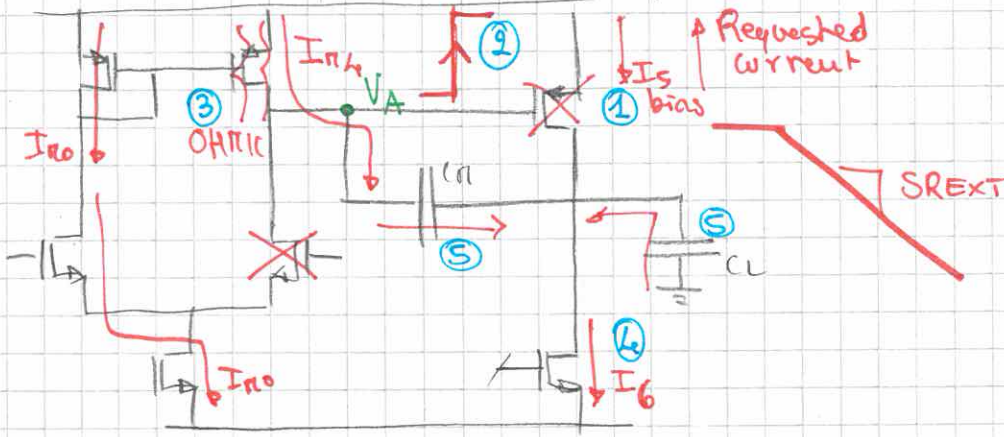
Now consider C_L presence $\rightarrow \Delta I_S = I_{ro} + I_L$ where

$$I_L = SR_{EXT} C_L = SR_{INT} C_L = \frac{I_{ro} C_L}{C_{cr}}$$

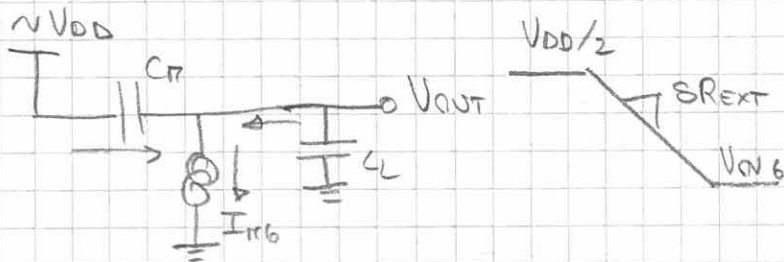
If $I_{sbias} > I_{ro} + I_L = I_{ro} \left(1 + \frac{C_L}{C_{cr}}\right)$ then M_5 can handle the additional current and nothing changes

$$SR_{EXT} = SR_{INT} = \frac{I_{ro}}{C_{cr}} \text{ but } \Delta I_S = I_{ro} + \frac{I_{ro} C_L}{C_{cr}}$$

Now let's analyze what happens with $I_{Sbias} < I_{ro} + I_L$:



- 1) Since the requested current $I_{ro} \left(1 + \frac{C_L}{C_n}\right) > I_{Sbias}$ is opposite and greater than I_{Sbias} , M_5 shuts off
- 2) The M_5 shut-off translates to a rapid reduction of $V_{SS} \rightarrow V_A$ rises to V_{DD}
- 3) Since $V_A \uparrow \uparrow \rightarrow V_{DS4} < V_{OV4}$ therefore M_4 goes ohmic and its current I_{M4} won't be I_{ro} anymore
- 4) The only current source remaining is M_6
- 5) The situation is now:



C_n and C_L have one end connected to a fixed voltage and the other tied to $V_{out} \rightarrow I_{IG}$ sees them "in parallel"
 this means that our ramp will be set at:

$$\underline{S_{REXT} = \frac{I_G}{C_n + C_L}}$$

Note: if C_L wasn't connected $S_{REXT} = S_{RINT} = \frac{I_G}{C_n}$

so $I_{M4} = I_G$
 \rightarrow OHMIC

Recap:

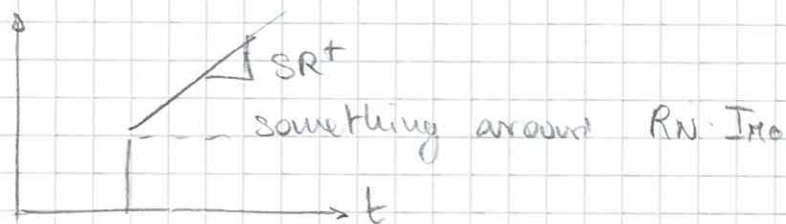
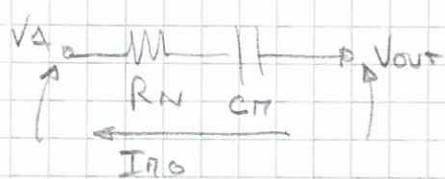
$$\left\{ \begin{array}{l} \underline{SR_{EXT}^+} = SR_{INT} = \underline{I_{ro}/C_{rr}} \end{array} \right.$$

$$\left\{ \begin{array}{l} \underline{SR_{EXT}^-} \rightarrow SR_{INT} = \underline{I_{ro}/C_{rr}} \end{array} \right.$$

$$\underline{\frac{I_6}{C_{rr} + C_L}} \quad \text{if } I_5 < I_{ro} \left(1 + \frac{C_L}{C_{rr}}\right)$$

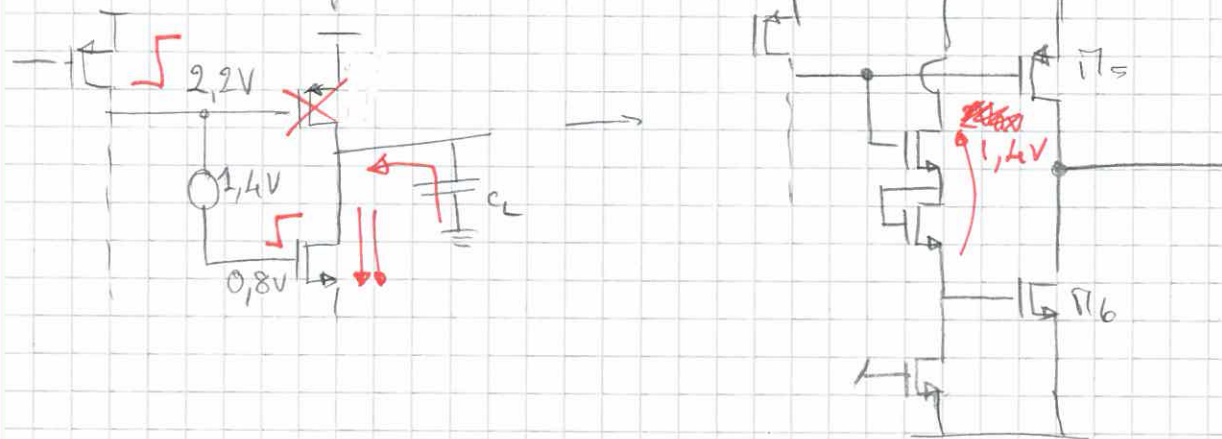
SR on a two stage compensated with R_N

Nothing changes except R_N will show an immediate voltage step on I_{ro} :



Class AB stage

On SR^- we need to make sure that $I_5 > I_{ro} \left(1 + \frac{C_L}{C_{rr}}\right)$ thus burning more power. The issue is that only I_5 can move. If we linked I_6 on the negative step only, we would recover the full SR:

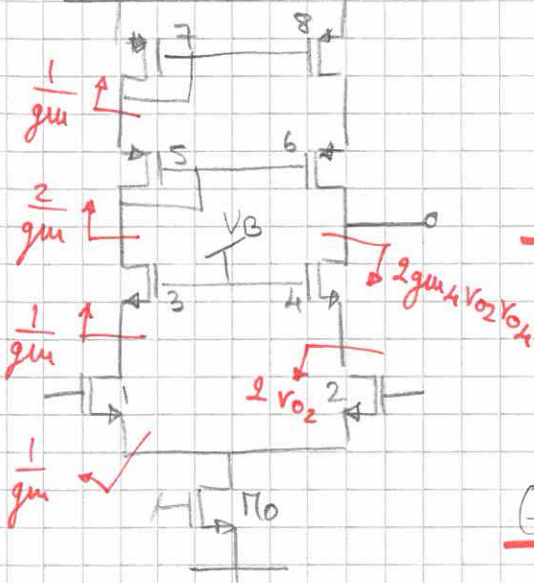


Therefore $SR_{EXT}^- = SR_{INT} \rightarrow F_{orr}$ improved because of the lower output current needed

Telescopic + folded cascode amplifiers

High gain \rightarrow 2 stage amp adopted \rightarrow 2 high impedance nodes that are critical to frequency response.

We can use a single stage:



By using cascode stages, we can increase the output impedance

$$R_{out} = gm_8 r_{o8} r_{o6} \parallel R_x \quad R_x = \frac{2gm_4 r_{o4} r_{o2}}{1 + \frac{1}{1}}$$

$$G_{loop} = -1 \text{ as always, so}$$

If $r_{o2} = r_{o4} = r_{o8} = r_{o6}$, $gm_8 = gm_4 = gm$

$$G_D = gm, \frac{gm r_o^2}{2} \rightarrow \text{We increased}$$

The usual gm, r_o of a factor μ .

The main limit to frequency response is C_L (other C_{gs} will show poles at $\sim \mu f_T$ so they're way higher)

$$GBWP = gm, \frac{gm r_o^2}{2} \frac{1}{2\pi C_L \cdot \frac{gm r_o^2}{2}} = \frac{gm \mu}{2\pi C_L}$$

Voltage dynamics:

$$V_{en,IN}^+ = V_B - V_{as3} + |V_T| \quad V_{en,IN}^- = V_{as1} + V_{ov0}$$

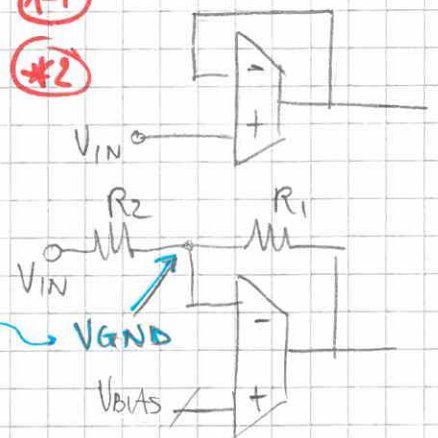
$$V_{out}^+ = V_{as7} + V_{ov6} \quad V_{out}^- = V_B - |V_T|$$

V_B serves as reference to get max $V_{en,IN}$ and min V_{out} .

- If V_B is high $\rightarrow \Delta V_{out} \downarrow$ but $\Delta V_{in,cm} \uparrow$ (*1)
- If V_B is low $\rightarrow \Delta V_{out} \uparrow$ but $\Delta V_{in,cm} \downarrow$ (*2)

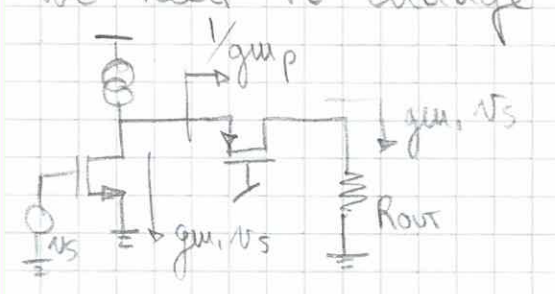
(*1) useful for OTAs used as buffers (large input swing needed)

(*2) Useful for OTAs with virtual ground (input swing is kept as low as possible)



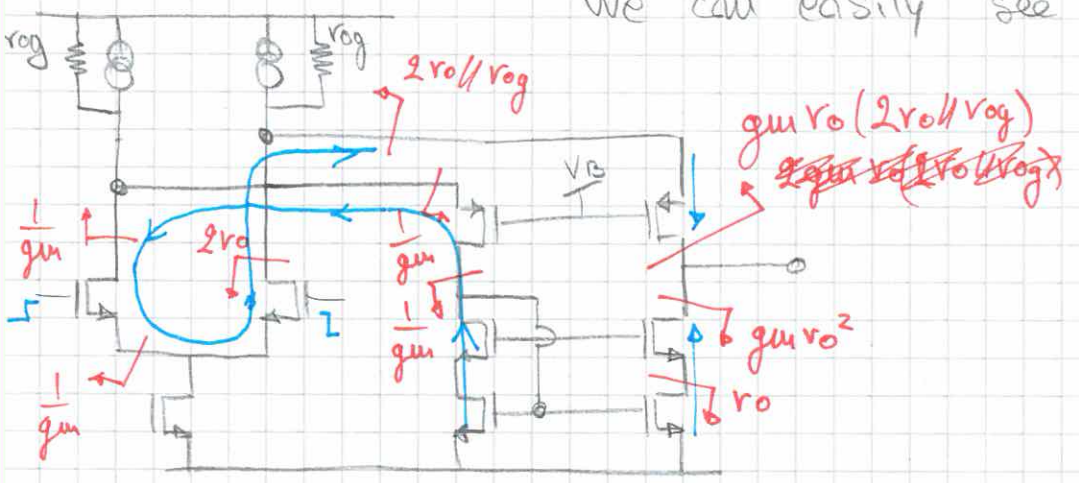
Folded cascode structure

Piling up transistors can't be done with low V_{DD} (V_{DD} decreases with technology scaling), this means we need to change the structure: use PMOS cascodes:



→ This is the concept we use when we talk about folding

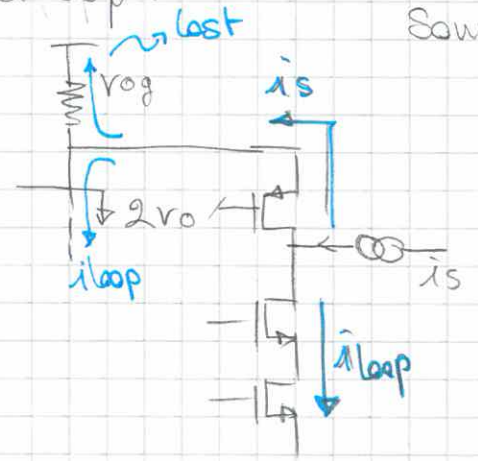
We can easily see that $i_{CC} = g_{m1} V_s$



$$R_{out} = R_{down} \parallel \frac{R_{up}^{(0)}}{1 - G_{loop}} \quad R_{down} = g_{m1} r_o^2 \quad R_{up} = g_{m1} r_o (2r_o \parallel r_{og})$$

G_{loop} :

Some current will be lost on r_{og} :



$$G_{loop} = - \frac{r_{og}}{r_{og} + 2r_o}$$

If $r_o \cong r_{og} \rightarrow G_{loop} = -\frac{1}{3}$

$$R_{up} = \frac{g_{m1} r_o \cdot \frac{2}{3} r_o}{1 + \frac{1}{3}} = g_{m1} \frac{r_o^2}{2}$$

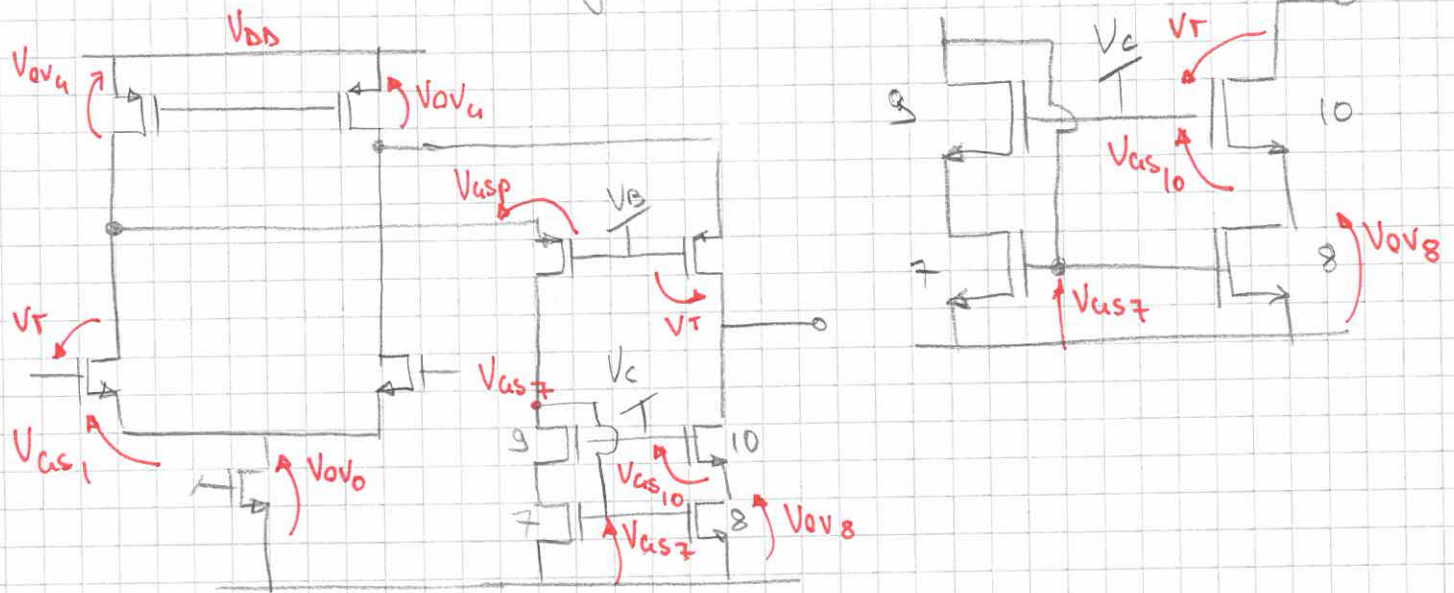
Therefore $R_{out} = g_{m1} \frac{r_o^2}{3}$ comparable value with a telescopic OTA

$R_{out}|_{FOLDER}$ can be matched to a $R_{out}|_{TELE}$ by changing

MOSET'S LENGTH

Folded cascode dynamics

Let's see how the cascode improved voltage dynamics (issue with V_{DD} scaling)



$V_{IN}^+ = V_{DD} - V_{OV4} + |V_T|$ or $V_B + |V_{ASp}| + |V_T|$ the more stringent of the two ($V_{Bmax} = V_{DD} - V_{OV4} - |V_{ASp}|$). If $V_B = V_{Bmax}$ the two conditions are the same

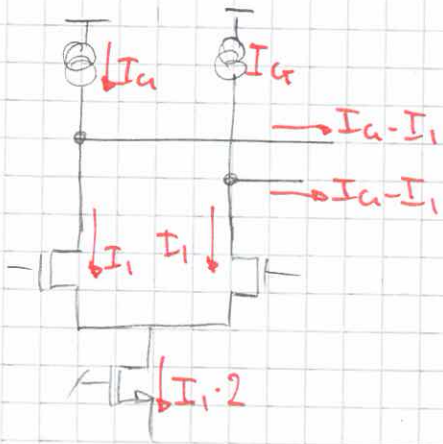
$$\underline{V_{IN}^-} = V_{AS1} + V_{OV0}$$

V_{OUT} , without the enhanced mirror, would be limited to $V_{AS7} + V_{OV10}$, but here we can choose $V_C = V_{OV8} + V_{AS10}$ so that:

$\underline{V_{OUT}^-} = V_C - V_T$ or $V_{OV10} + V_{OV8}$ (check and see that the two conditions are the same!)

$$\underline{V_{OUT}^+} = V_B + |V_T|$$

Drawback of folded cascode: current consumption



Condition to have a working circuit is $I_G > 2I_1$ because on bias the two branches will see $I_G - I_1$, while on large steps (SR):

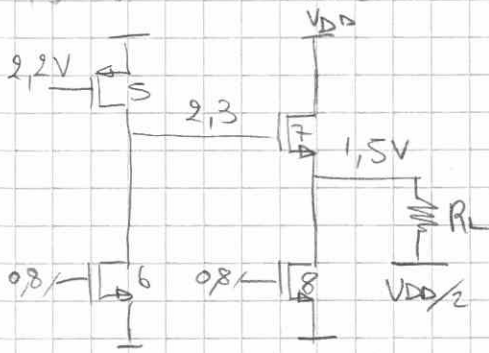
- One branch will have I_G
- The other will have $I_G - 2I_1$ (*)

Since with $I_G < 2I_1$ (*) can push MOSFETS in ohmic region, we need to burn more power.

This means we need to burn at least twice the current of a equal telescopic OTA

12) Opamp output stages: Class A, Class B, Efficiency + HD

We need to decouple OTA's high out impedance in order to drive resistive loads \rightarrow Design a buffer stage



Set $V_{ov} = 0,2$ for all mosfet.

Previously V_{G7} was set to $1,5V$

but now we need $1,5V$ on V_{out} so

that $I_{R_L}|_{bias} = 0$

$$V_{G7} = V_{out} + V_{as7} = 1,5 + 0,8 = 2,3V$$

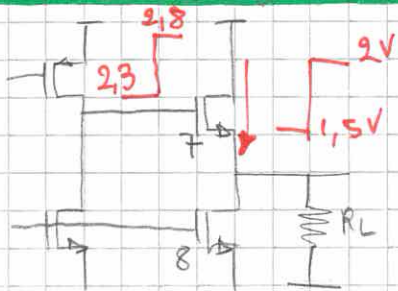
$$\frac{1}{2} \mu_p' C_{ox} \frac{W_5}{L_5} \left(1 + \frac{V_{DSS} - V_{DSSAT5}}{V_{AS}} \right) = \frac{1}{2} \mu_n' C_{ox} \frac{W_6}{L_6} \left(1 + \frac{V_{DSS6} - V_{DSSAT6}}{V_{A6}} \right)$$

$$V_{DSS} = 3V - 2,3 \quad V_{DSSAT5} = 0,2V \quad V_{DSS6} = 2,3V \quad V_{DSSAT6} = 0,2V$$

L_5, L_6 are set because of differential gain: $L_5 = L_6$

$$\text{So } \underline{W_6 = W_5} \left(\frac{1 + \frac{0,5}{V_{AS}}}{1 + \frac{2,1}{V_{A6}}} \right)$$

Positive/negative swings



Positive swing: V_{G7} can go up to $2,8V$

$(V_{DD} - V_{ov5}) \rightarrow V_{out}$ goes up to $2V$ if

$\text{Gain}|_{BUF} = 1 \rightarrow \text{ideal}$

This does not happen for two reasons:

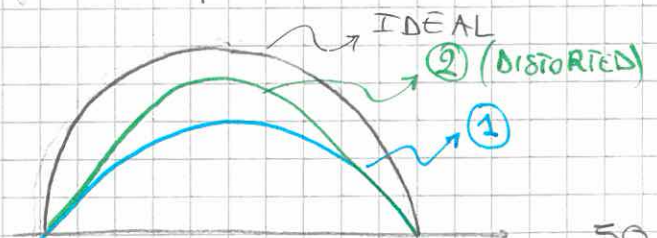
$$\textcircled{1} \quad G_{BUF} = \frac{R_L // r_{os}}{R_L // r_{os} + 1/g_{m7}} \approx \frac{R_L}{R_L + 1/g_{m7}} < 1$$

$\textcircled{2} \quad 1/g_{m7}$ is not linear, when $V_{as} \uparrow \uparrow$, $1/g_{m7} \rightarrow \rightarrow$ leading to

an output distortion of (distortion of a MOS buffer):

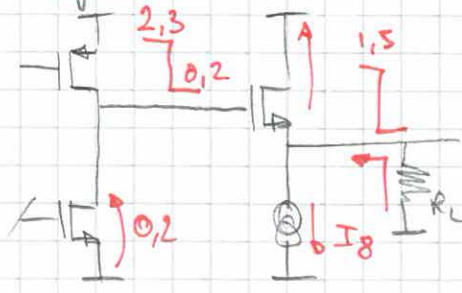
$$HD_2 = \frac{V_{as}}{2V_{ov}} \frac{1}{1 + g_m R_L}$$

(Not explained in this course!!)



On the positive swing, peak current $I_s = I_8 + \frac{\Delta V_{out}}{R_L}$
 where $\Delta V_{out} = 5V$ and R_L (arbitrary) = 500Ω

Negative swing



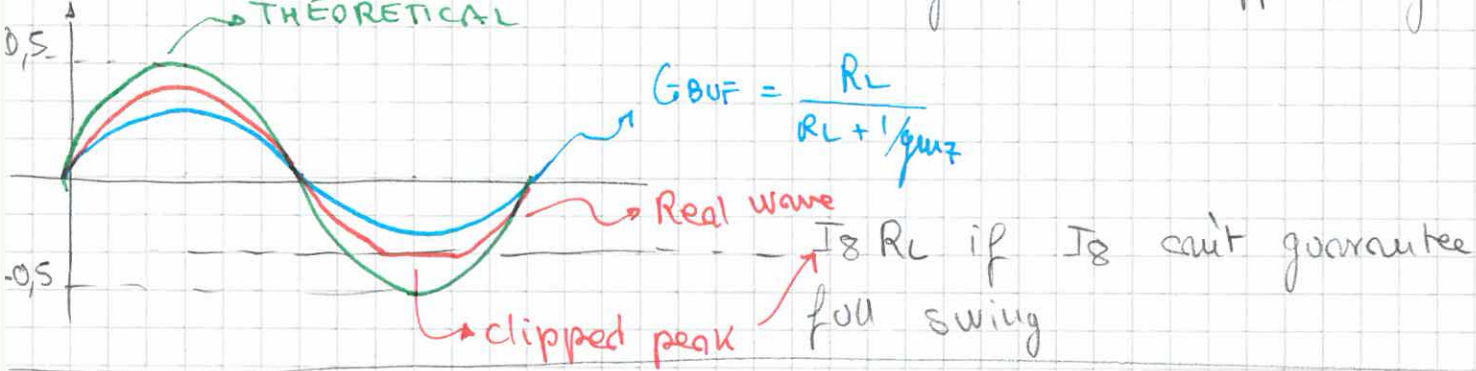
Even though $V_{a7} \rightarrow V_{a6}$, M_7 will move towards off state. If $I_8 = I_7 = 0,5mA$, when M_7 shuts off there's only I_8 providing current \rightarrow

$\Delta V_{out}^- = I_8 \cdot R_L = 0,5mA \cdot 500\Omega = 0,25V \rightarrow V_{out} = 1,5 - 0,25$
 Therefore $V_{a7} = V_{out}^- + V_T = 1,25 + 0,6 = 1,85V$

Since on the positive side we would have 0,5V theoretical swing, we would like to also have that on the negative \rightarrow

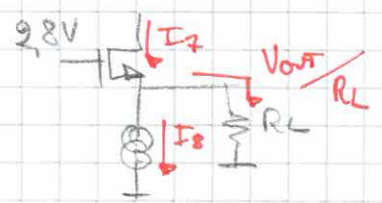
$\Delta V_{out}^- = I_8 R_L = 0,5V \rightarrow I_8 = \frac{0,5V}{500\Omega} = 1mA$

$I_7 = I_8 = 1mA$ current doubled to guarantee $1V_{pp}$ swing:



To verify the peaks: consider max positive peak:

$I_{7,MAX} = K_7 (V_{a7} - V_{DD} - \Delta V - V_T)^2 = \frac{\Delta V}{R_L} + I_8$
 Values: $12,5 \frac{\mu A}{V^2}$, $2,8$, $1,5$, $0,6$, $R_L = 500\Omega$, $I_8 = 1mA$

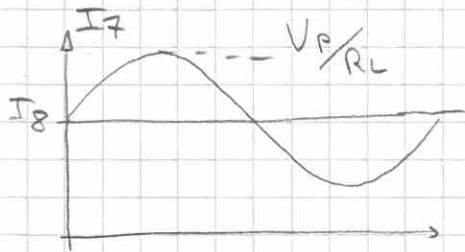


$\Delta V = 0,33V \rightarrow$ definitely lower than 0,5 (ideal)

Note: formula includes a squared variable.

M_7 is in large signal operation \rightarrow we can't consider it as a linear stage anymore. Positive peaks will be slightly higher and negative peaks smaller

Class A buffer power efficiency



$$\eta = \frac{P_{load}}{P_{DISSIPATED}}$$

consider a sinusoidal waveform

$$\eta = \frac{\frac{V_p^2}{2RL}}{V_{DD} \cdot I_8} = \frac{V_p^2}{2RL V_{DD} I_8} \quad \rightarrow \text{estimate this}$$

Since $V_p < \frac{V_{DD}}{2}$ at most and we know that $I_8 RL = V_p$

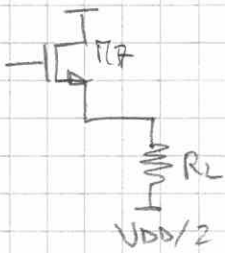
$$\eta = \frac{V_p^2}{2 V_{DD} \cdot V_p} = \frac{V_{DD}/2}{2 \cdot V_{DD}} = \frac{1}{4} \quad \text{max theoretical out efficiency}$$

I_7, I_8

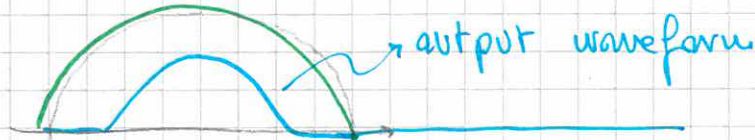
25% is very low because of the continuous current consumption even when no signal is applied

Class B stages - push-pull

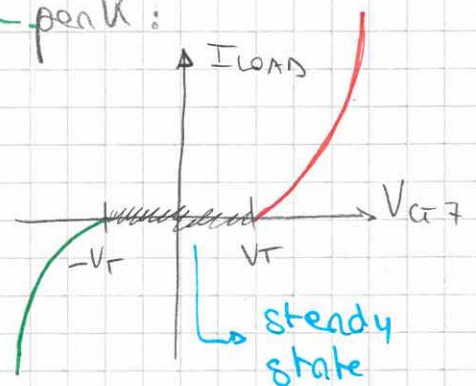
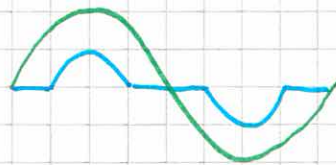
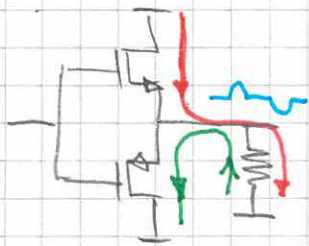
Concept: deliver current only when needed \rightarrow remove I_8



\rightarrow V_{G7} can deliver current only when $V_{G7} > \frac{V_{DD}}{2} + V_T$ \rightarrow heavy distortion

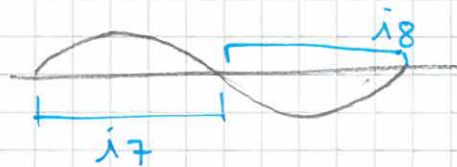


Solution: take care of the negative-penik:



We can already see that out wave is symmetric \rightarrow no HD2, only third harmonics

Check distortion



$$i_{OUT} = i_7 - i_8$$

$$i_7 = A_0 + A_1 \sin(\omega_0 t + \phi_1) + A_2 \sin(2\omega_0 t + \phi_2) + A_3 \sin(3\omega_0 t + \phi_3)$$

$i_8 = -i_7(t - \frac{T}{2}) \rightarrow i_8$ is the shifted version of i_7 by half a period:

$$i_8 = A_0 + A_1 \sin(\omega_0(t - \frac{T}{2}) + \phi_1) + A_2 \sin(2\omega_0(t - \frac{T}{2}) + \phi_2) + \dots$$

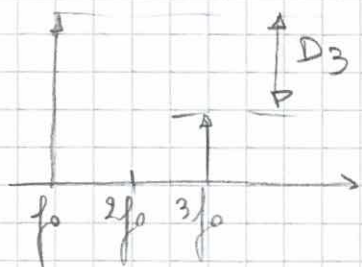
We can see that $\omega_0 \frac{T}{2} = \frac{2\pi}{T} \cdot \frac{T}{2} = \pi$ $2\omega_0 \frac{T}{2} = 2\pi \dots$ so

$$i_8 = A_0 - A_1 \sin(\omega_0 t + \phi_1) + A_2 \sin(2\omega_0 t + \phi_2) - A_3 \sin(\omega_0 t + \phi_3)$$

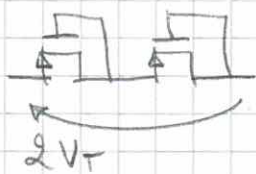
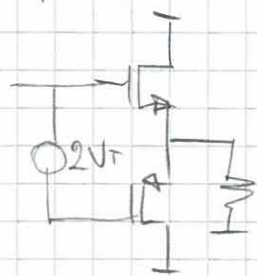
Since $i_{OUT} = i_7 - i_8 =$ even terms cancel out =

$$i_{OUT} = 2A_1 \sin(\omega_0 t + \phi_1) + 2A_3 \sin(3\omega_0 t + \phi_3)$$

No HD2 if π_7, π_8 have exact voltage-current curves



Distortion is produced by the dead zone. To eliminate it, use a voltage shifter of $2V_T \rightarrow$ Class AB

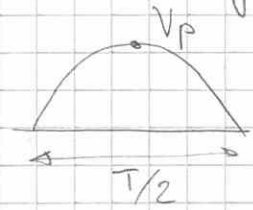


\rightarrow set transistors V_{GS} in order to have $2V_T$ shift

Class B efficiency

$$P_L = \frac{V_p^2}{2R_L} \quad \text{as always} \quad P_{\text{Diss}}^+ = \frac{V_{DD}}{2} \cdot \overline{I_L} \quad \text{where } \overline{I_L} \text{ is}$$

the average dissipated current over a period



Neglect signal distortion

$$\omega = \frac{2\pi}{T}$$

$$\overline{I_L} = \frac{1}{T} \cdot \frac{2}{T} \cdot \frac{V_p}{R_L} \int_0^{T/2} \sin(\omega t) dt \cdot \frac{2\pi}{T} =$$

$$= \frac{1}{\pi} \frac{V_p}{R_L} \int_0^{\pi} \sin \theta d\theta = \frac{2}{\pi} \frac{V_p}{R_L} \quad \rightarrow \quad P_{\text{Diss}}^+ = \frac{V_{DD}}{2} \cdot \frac{2}{\pi} \frac{V_p}{R_L}$$

Same happens for P_{Diss}^- , so $\overline{I_L}$ is the same for the entire period.

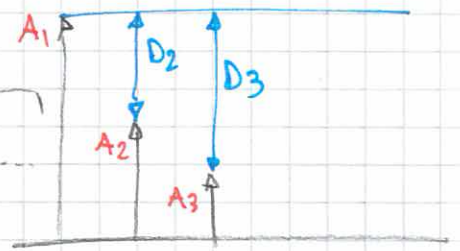
$$\eta = \frac{P_L}{P_{\text{Diss}}} = \frac{\frac{V_p^2}{2R_L}}{\frac{2}{\pi} \cdot \frac{V_{DD}}{2} \cdot \frac{V_p}{R_L}} = \frac{\pi}{4} \frac{2V_p^2}{V_p V_{DD}} = \frac{\pi}{4} \frac{2V_p}{V_{DD}}$$

We set $V_p < \frac{V_{DD}}{2}$ therefore $\eta_{\text{MAX}} = \frac{\pi}{4} \approx 78\%$

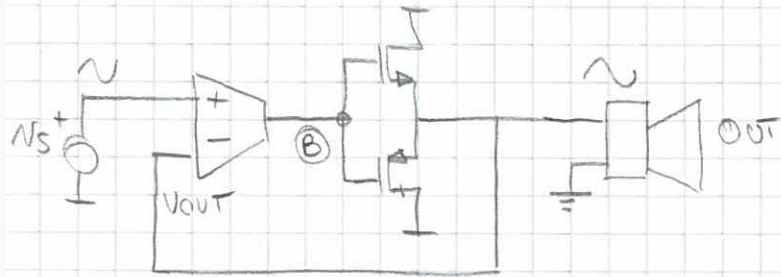
We improved efficiency by a lot

3) Distortion and feedback

$$THD = \sqrt{\frac{A_2^2}{A_1^2} + \frac{A_3^2}{A_1^2} + \dots} = \sqrt{D_2^2 + D_3^2 + \dots}$$

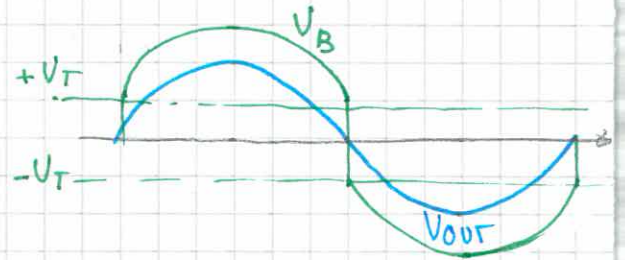


Loop can reduce distortion:

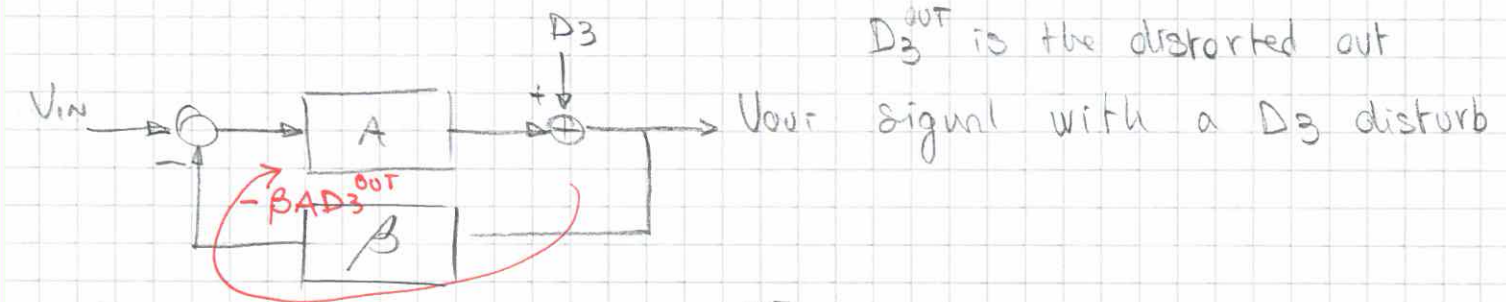


Suppose $G_{loop} = -\infty$
 With this configuration
 $V_{out} = V_{in}$

The only way to have $V_{out} = V_{in}$ is that on V_B the voltage is pre distorted:



This can be also seen through feedback theory by adding a distortion:

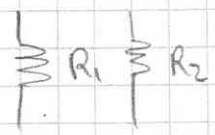


D_3^{out} is the distorted out
 signal with a D_3 disturb

$$D_3^{out} = D_3 - \beta A D_3^{out} \rightarrow D_3^{out} = D_3 \frac{1}{1 + A\beta}$$

We see that out distortion is reduced by $A\beta$ and on input there will be a $-\beta A D_3^{out}$ signal (pre-distort)

(4) Variability and matching of resistors → Pelgrom



Suppose $R_1 = R_2 = R$ In reality, processes will have a variability on these values



$$R = \rho \cdot \frac{L}{\Delta \cdot W} = \frac{\rho}{\Delta} \cdot \frac{L}{W} = R_0 \cdot \frac{L}{W}$$

$$\frac{\Delta R}{R} = \frac{\Delta R_0}{R_0} + \frac{\Delta L}{L} + \frac{\Delta W}{W}$$

→ worst case

process is precise = Negligible contribution

R_0 = unit square resistance, fixed by technology

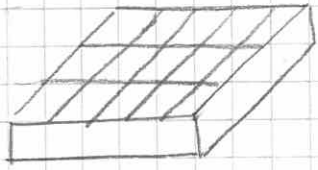
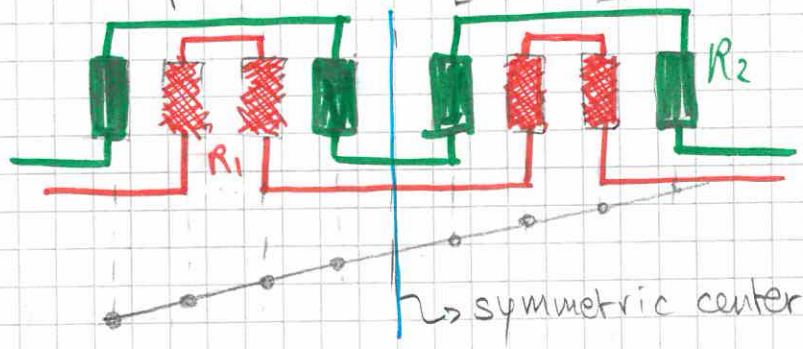
$R_0 = 110 \Omega$ unsilicided n+ polysilicon

$= 10 \Omega$ silicided (doped with metals) n+ polysilicon

Along R_0 , manufacturers will give a $\% \cdot \mu m$ coefficient used in the variability formula:

$$\frac{\Delta R}{R} = \frac{K \Delta R/R}{\sqrt{WL}} \rightsquigarrow \text{statistical spread, } K = [\% \cdot \mu m]$$

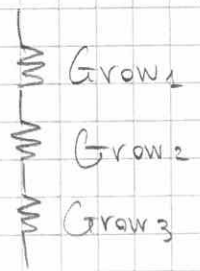
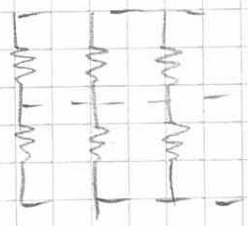
Deterministic spread can be reduced



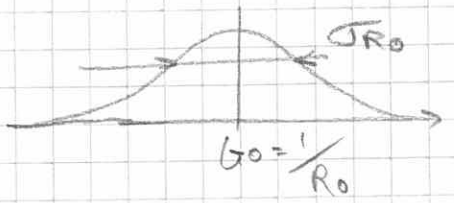
Each R_0 has a different value (statistical)

We can divide the resistor in rows and columns of single resistors.

$$G_0 = 1/R_0 \rightsquigarrow \text{average resistance}$$



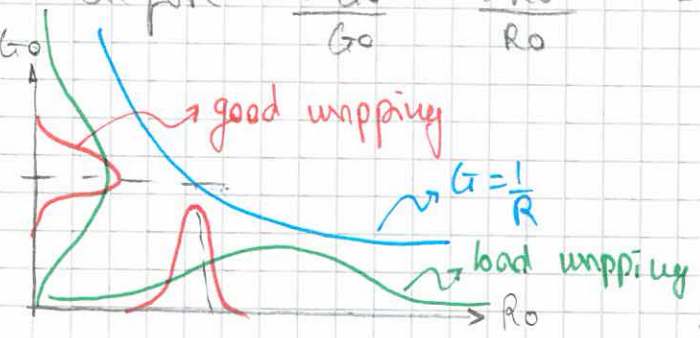
$$G_{row} = \sum G_i = N G_0$$



$\sigma^2_{G_{row}} = N \sigma^2_{G_0}$ → Now we want to link the conductance σ to resistance σ → relation is not linear because $G = \frac{1}{R}$

Linearize $dG_0 = -\frac{dR_0}{R_0^2} R_0 = -\frac{dR_0}{R_0} \leftarrow (*)$

Therefore $\frac{\sigma_{G_0}^2}{G_0} = \frac{\sigma_{R_0}^2}{R_0} \rightarrow$ be careful! This holds for small σ , large σ^2 will be influenced by the $1/R$ relation



Now that we have computed the variability for a row,

we can analyze the columns

$R_T = \sum_{\text{TOTAL}} \pi R_r = \pi R_r = \frac{\pi}{N} R_0 \sim$ intuitive $(*)$

We know that $\sigma_{R_T}^2 = \pi \sigma_{R_r}^2 \rightarrow \frac{\sigma_{R_T}^2}{R_T^2} = \frac{\pi \sigma_{R_r}^2}{\pi^2 R_r^2} = \frac{1}{\pi} \frac{\sigma_{R_r}^2}{R_r^2}$

$\frac{1}{\pi} \frac{\sigma_{G_r}^2}{G_r^2} = \frac{1}{\pi} \frac{N \sigma_{G_0}^2}{N^2 G_0^2} = \frac{1}{\pi N} \frac{\sigma_{G_0}^2}{G_0^2} = \frac{1}{\pi N} \frac{\sigma_{R_0}^2}{R_0^2}$

So $\frac{\sigma_{R_T}^2}{R_T^2} = \frac{\sigma_{R_0}^2}{R_0^2} \cdot \frac{1}{\pi N}$ small relative spread

Now, suppose that each square has length Δ , then $N = \frac{L}{\Delta}$ $M = \frac{W}{\Delta}$ we can express the relative spread

$\frac{\sigma_{R_T}^2}{R_T^2} = \frac{\Delta^2}{WL} \frac{\sigma_{R_0}^2}{R_0^2} = \frac{K^2}{WL}$ $\frac{\Delta R_T}{R_T} = \sqrt{\frac{\sigma_{R_T}^2}{R_T^2}} = \frac{K_R}{\sqrt{WL}}$

Apply this to R_1, R_2 : $\Delta R = R_1 - R_2$ $\frac{\sigma_{R_1}}{R_1} = \frac{K}{\sqrt{WL_1}}$ $\frac{\sigma_{R_2}}{R_2} = \frac{K}{\sqrt{WL_2}}$

Suppose $WL_1 = WL_2$ then

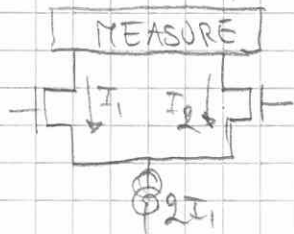
$\sigma^2(\Delta R) = \sigma_{R_1}^2 + \sigma_{R_2}^2 = 2 \sigma_{R_1}^2 = 2 \frac{K^2}{WL} R_1^2$

So $\frac{\Delta R}{R} = \sqrt{\frac{\sigma_{\Delta R}^2}{R^2}} = \sqrt{2} \frac{K_R}{\sqrt{WL}}$

The $\frac{K_{\Delta R}}{R}$ the manufacturer gives us is exactly

$\frac{K_{\Delta R}}{R} = \sqrt{2} K \left[\frac{1}{\mu m} \right]$

15) Variability and matching: V_T relative matching

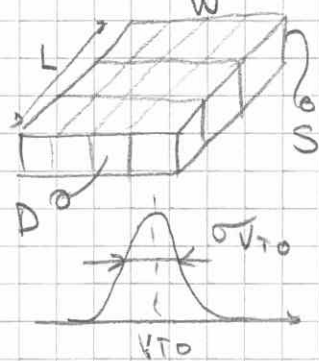


$I_1 \neq I_2$ always because of mosfet V_T variability. We can change the deterministic contribution by using the common centroid technique \rightarrow layout issue



We lay out MOSFets in // on a center axis

Suppose we divide the mosfet in square cells. Each cell has a V_{T0} square spread plus a $\sigma_{V_{T0}}$ contribution (statistical)



$$\bar{V_T} = \frac{\sum V_{Ti}}{N} = \frac{N V_{T0}}{N} = V_{T0}$$

Since $\sigma^2(\sum_N X) = N \sigma_X^2$ and $\sigma^2\left(\frac{X}{b}\right) = \frac{\sigma_X^2}{b^2}$

We can say that $\sigma_{V_T}^2 = \frac{\sigma^2(\sum_N V_{Ti})}{N^2} = \frac{N \sigma_{V_{T0}}^2}{N^2} = \frac{\sigma_{V_{T0}}^2}{N}$

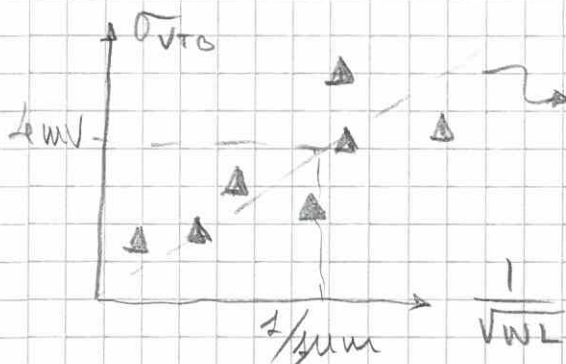
Let's say $A_0 = \frac{WL}{N}$ area of a single cell, then

$$\sigma^2(\bar{V_T}) = \frac{\sigma_{V_{T0}}^2 A_0}{WL} \approx \frac{K_{VT}^2}{KL} \rightarrow \sigma_{V_T} = \frac{K_{VT}}{\sqrt{WL}}$$

Since we're interested on the mismatch between M_1, M_2 we have:

- mean square $E(\Delta V_T) = V_{T1} - V_{T2} = V_{T0} - V_{T0} = 0$
- $\sigma^2(\Delta V_T) = \sigma_{V_{T1}}^2 + \sigma_{V_{T2}}^2 = 2 \sigma_{V_{T0}}^2$

Therefore $\sigma^2(\Delta V_T) = 2 \frac{K^2}{WL} \rightarrow \sigma_{\Delta V_T} = \sqrt{2} \frac{K_{\Delta V_T}}{\sqrt{WL}}$



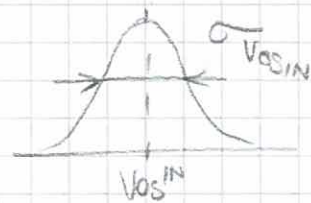
\rightarrow slope set by manufacturing process K

16) Offset : deterministic + statistical contribution

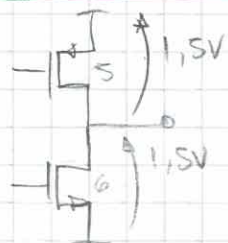
We can always model an output offset to an input referred one

$$V_{os}^{IN} = \frac{V_{os}^{out}}{A_d}$$

differential gain



Deterministic offset



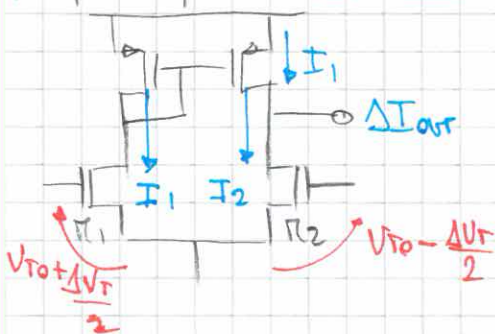
M_5 and M_6 should be properly sized to have the same V_{DS} , thus the proper $\frac{V_{DS}}{2}$ bias at the output \rightarrow NO V_{os}

Statistical offset

This kind of offset is introduced by the following:

- Input pair $\rightarrow V_T$ mismatch + K mismatch
- Diff mirror $\rightarrow V_T$ mismatch + K mismatch

Input pair statistical V_T mismatch



Consider a symmetrical mismatch $+ \frac{\Delta V_T}{2}$ between M_1, M_2 . Then currents will be:

$$I_1 = K_{IN} \left(V_{as1} - V_{T0} - \frac{\Delta V_T}{2} \right)^2 \quad I_2 = K_{IN} \left(V_{as2} - V_{T0} + \frac{\Delta V_T}{2} \right)^2$$

$\Delta I = I_2 - I_1$ and $V_{as} - V_{T0} = V_{ov}$, so

$$\Delta I = K_{IN} \left(V_{ov} - \frac{\Delta V_T}{2} \right)^2 - K_{IN} \left(V_{ov} + \frac{\Delta V_T}{2} \right)^2 =$$

$$= K_{IN} \left(V_{ov}^2 - V_{ov} \Delta V_T + \frac{\Delta V_T^2}{4} \right) - K_{IN} \left(V_{ov}^2 + V_{ov} \Delta V_T + \frac{\Delta V_T^2}{4} \right)$$

$\Delta I = 2 K_{IN} V_{ov} \Delta V_T$ since $2 K V_{ov} = g_{m,IN}$



$i_{CC} = \frac{g_{m1}}{2} V_{os}^{IN}$ then

$$g_{m1} V_{os}^{IN} = \frac{2 K_{IN} V_{ov}}{g_{m1}} \Delta V_T \quad \underline{V_{os}^{IN} = \Delta V_T}$$

Input pair statistical K mismatch

Same reasoning $I_1 = \left(k + \frac{\Delta k}{2}\right) V_{ov}^2$ $I_2 = \left(k - \frac{\Delta k}{2}\right) V_{ov}^2$

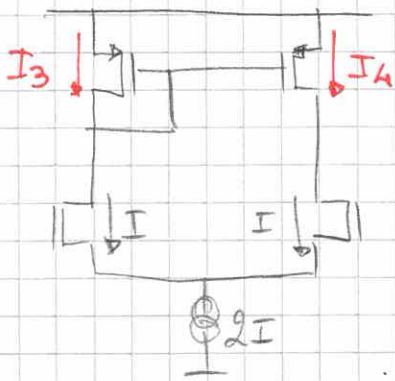
$$\Delta I = I_1 - I_2 = \Delta k V_{ov}^2 \rightarrow g_{m1} V_{os}^{IN} = \Delta k V_{ov}^2$$

$$V_{os}^{IN} = \frac{\Delta k V_{ov}^2}{g_{m1}} \cdot \frac{k}{k} = \frac{\Delta k}{k} \frac{V_{ov}^2 k}{g_{m1}} = \frac{\Delta k}{k} \frac{I}{2I} V_{ov} = \frac{\Delta k}{k} \frac{V_{ov}}{2}$$

\xrightarrow{I}
 $\xrightarrow{\frac{gI}{V_{ov}}}$

So $V_{os}^{IN} = \frac{\Delta k}{k} \cdot \frac{V_{ov}}{2}$

Mirror statistical V_T mismatch



$$I_3 = k_n \left(V_{os} - V_{T0} + \frac{\Delta V_T}{2} \right)^2 = k_n \left(V_{ov} + \frac{\Delta V_T}{2} \right)^2$$

$$I_4 = k_n \left(V_{ov} - \frac{\Delta V_T}{2} \right)^2$$

$$I_3 = k_n \left(V_{ov}^2 + V_{ov} \Delta V_T + \frac{\Delta V_T^2}{4} \right)$$

$$I_4 = k_n \left(V_{ov}^2 - V_{ov} \Delta V_T + \frac{\Delta V_T^2}{4} \right)$$

$$\Delta I = I_4 - I_3 = \underbrace{k_n \cdot 2 \cdot V_{ov} \Delta V_T}_{g_m} \quad \text{So}$$

$$g_{m1} V_{os}^{IN} = g_{m_n} \Delta V_{Tn} \rightarrow V_{os}^{IN} = \frac{g_{m_n} \Delta V_{Tn}}{g_{m1}}$$

$$V_{os}^{IN} = \frac{V_{ov} \Delta V_{Tn}}{V_{ov}}$$

Mirror statistical K mismatch

$$I_3 = \left(k_n - \frac{\Delta k}{2}\right) V_{ov_n}^2 \quad I_4 = \left(k_n + \frac{\Delta k}{2}\right) V_{ov_n}^2 \rightarrow \Delta I = I_4 - I_3 = \Delta k V_{ov_n}^2$$

$$V_{os}^{IN} g_{m1} = \Delta k V_{ov_n}^2 \rightarrow V_{os}^{IN} = \frac{\Delta k}{k} \cdot \frac{k V_{ov}^2}{g_{m1}} = \frac{\Delta k_n}{k_n} \frac{V_{ov_n}}{2}$$

$$V_{os}^{IN} = \frac{\Delta k_n}{k_n} \frac{V_{ov_n}}{2}$$

Summary of mismatches for input referred offset

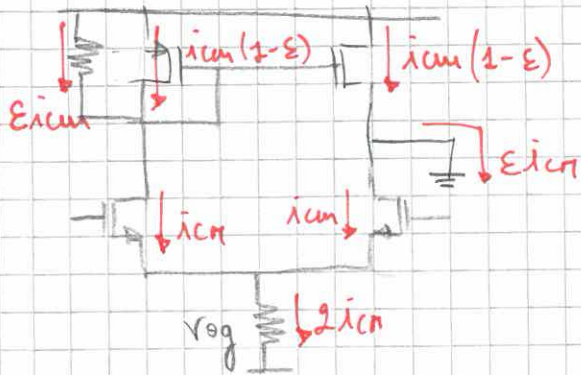
$$V_{OS} \Big|_{\text{INPUT REF}} = \underbrace{\Delta V_{TIN} + \frac{\Delta K_{IN}}{K_{IN}} \frac{V_{OVIN}}{2}}_{\text{input pair}} + \underbrace{\Delta V_{TP} + \frac{\Delta K_{TP}}{K_{TP}} \frac{V_{OVIN}}{2}}_{\text{mirror pair}}$$

$$\sigma_{V_{OS}} = \sqrt{\sigma_{V_{TIN}}^2 + \sigma_{V_{TP}}^2 \left(\frac{V_{OV2}}{V_{OV1}} \right)^2 + \left[\sigma_{\left(\frac{\Delta K_{IN}}{K_{IN}} \right)}^2 + \sigma_{\left(\frac{\Delta K_{TP}}{K_{TP}} \right)}^2 \right] \left(\frac{V_{OVIN}}{2} \right)^2}$$

17) CMRR: deterministic and statistical limits

CMRR deterministic contribution is purely related to the circuit topology. In our case, the only asymmetry we can find is the mirror and input pair impedance

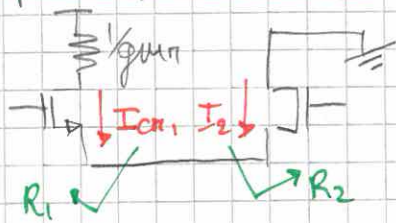
Mirror deterministic contribution



It's easy to find that some portion of i_{icm} is lost through r_{on}

$$\epsilon = \frac{1/g_{mM}}{\frac{1}{g_{mM}} + r_{on}} = \frac{1}{1 + g_{mM} r_{on}}$$

Input pair impedance mismatch



We supposed that $2i_{icm}$ always perfectly divides through R_1, R_2 . This is not true as $R_1 \neq R_2$

$$R_1 = \frac{1/g_{mM} + r_{o1}}{1 + g_{m2} r_{o1}}$$

$$R_2 = \frac{r_{o1}}{1 + g_{m1} r_{o1}}$$



We can say with minimal error that $v_g \sim v_{cm}$ $v_s \sim v_{cm}$ so

$$i_1 = 2i_{icm} \frac{R_2}{R_1 + R_2} = \frac{v_s}{r_{eq}} \frac{R_2}{R_1 + R_2} \quad \text{and} \quad i_2 = \frac{v_s}{r_{eq}} \frac{R_1}{R_1 + R_2}$$

With a perfect mirror, $i_{ccout} = i_2 - i_1 = \frac{v_s}{r_{eq}} \frac{R_1 - R_2}{R_1 + R_2}$

$$\underline{R_1 - R_2} = \frac{1/g_{mM}}{1 + g_{m1} r_{o1}} \quad \underline{R_1 + R_2} = \frac{2r_{o1} + 1/g_{mM}}{1 + g_{m1} r_{o1}}$$

Therefore
$$\varepsilon \frac{\sqrt{v_{ov1}}}{2r_{o1}} = \frac{V_{S1}}{r_{o1}} \frac{1/g_{m1}}{1+g_{m1}r_{o1}} \frac{1+g_{m1}r_{o1}}{2r_{o1} + 1/g_{m1}}$$

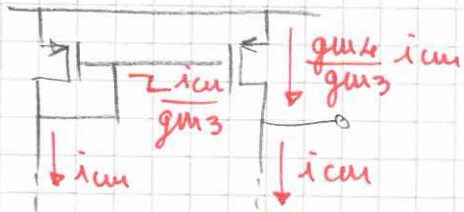
$$\varepsilon \frac{\sqrt{v_{ov1}}}{2r_{o1}} = \frac{V_{S1}}{r_{o1}} \cdot \frac{1}{2} \frac{1}{g_{m1}r_{o1}} \quad \text{if } \alpha = \frac{V_{S1}}{V_{ov1}} \approx 1 \text{ then}$$

$$\varepsilon \approx \frac{1}{g_{m1}r_{o1}}$$

If we also consider mirror mismatch:

$$\varepsilon_{DET} \approx \frac{1}{g_{m1}r_{o1}} + \frac{1}{g_{m1}r_{o1}}$$

Mirror statistical gm mismatch



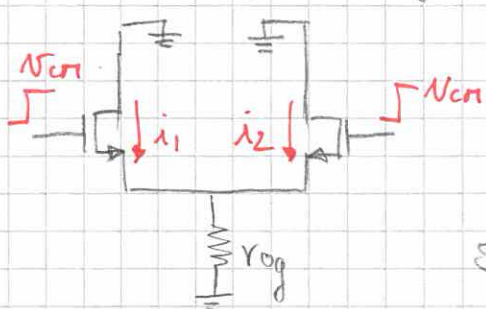
We now consider a difference in gm (caused by V_T or k statistical variability).

$$i_{out} = i_{cm} \left(\frac{g_{m1}}{g_{m3}} - 1 \right) = i_{cm} \frac{g_{m1} - g_{m3}}{g_{m3}} \triangleq i_{cm} \frac{\Delta g_{m1}}{g_{m1}}$$

Since $\varepsilon i_{cm} = i_{cm} \frac{\Delta g_{m1}}{g_{m1}} \rightarrow \underline{\varepsilon_{STAT MIRROR} = \frac{\Delta g_{m1}}{g_{m1}}}$

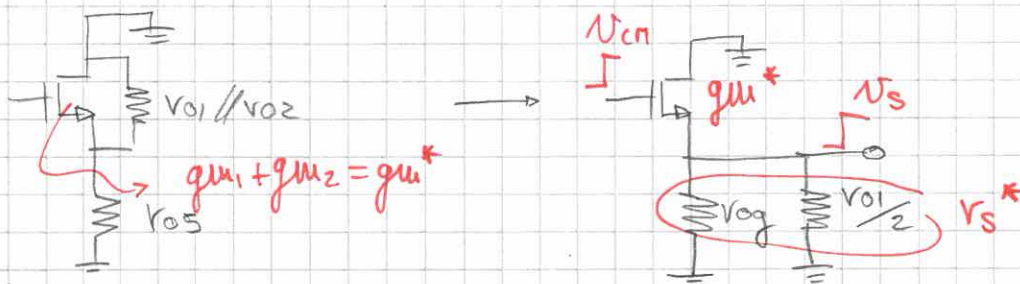
Input pair statistical mismatch for CTRR

Since we already estimated the mirror mismatch, put everything to ground:



As usual $i_{cc\ out} = i_2 - i_1$

To estimate gm_2, gm_2 mismatches we could think of folding the structure in half (since V_{cm} is applied on both gates)



We can compute $V_s = V_{cm} \frac{r_s^*}{r_s^* + 1/gm^*}$, now, since V_{cm} it's the same, it's easy to see that:

$$i_1 = gm_1 (V_{cm} - V_s) \quad i_2 = gm_2 (V_{cm} - V_s)$$

$$i_2 - i_1 = (gm_2 - gm_1) (V_{cm} - V_s) = \Delta gm V_{cm} \left(1 - \frac{r_s^*}{r_s^* + 1/gm^*} \right)$$

Let's compare it with typical CTRR current

$$\epsilon \frac{V_{cm}}{2r_{og}} = \Delta gm V_{cm} \frac{1/gm^*}{r_s^* + 1/gm^*} \quad \text{where } gm^* = gm_1 + gm_2 = 2gm$$

negligible

$$r_s^* = r_{og} // r_{o1/2}$$

$$\epsilon \frac{V_{cm}}{2r_{og}} = \frac{\Delta gm}{2gm} \cdot \frac{r_{og} + r_{o1/2}}{r_{og} r_{o1/2}} = \frac{V_{cm}}{2r_{og}} \cdot \frac{\Delta gm}{gm} \left(1 + \frac{2r_{og}}{r_{o1}} \right)$$

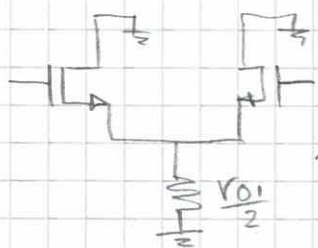
$$\epsilon = \frac{\Delta gm_{IN}}{gm_{IN}} \left(1 + \frac{2r_{og}}{r_{o1}} \right)$$

What's the meaning of this?

$r_{oq} \rightarrow \infty$

$$CMRR = \frac{2 g_{m1} r_{oq}}{\epsilon} = \frac{2 g_{m1} r_{oq}}{\frac{\Delta g_{m1}}{g_{m1}} \left(1 + \frac{2 r_{oq}}{r_{o1}}\right)} = \frac{g_{m1}^2 r_{o1}}{\Delta g_{m1}}$$

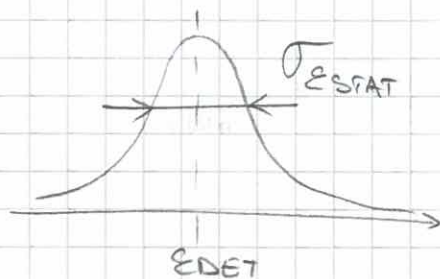
We see that even though we have a perfect tail generator ($r_{oq} \rightarrow \infty$) we still have a higher limit for the CMRR set by r_{o1} . This means:



This tells us that having a great r_{oq} may be not enough to have a CMRR high enough

Recap

$$\epsilon_{STAT} = \frac{\Delta g_m}{g_m} \Big|_{\pi} + \frac{\Delta g_m}{g_m} \Big|_{IN} \left(1 + \frac{2 r_{oq}}{r_{o1}}\right)$$



$$\epsilon_{TOT} = \epsilon_{DET} + \epsilon_{STAT}$$

$$\sigma_{STAT} = \sqrt{\frac{\sigma_{\Delta g_m \pi}^2}{g_{m\pi}^2} + \frac{\sigma_{\Delta g_m IN}^2}{g_{m IN}^2} \left(1 + \frac{2 r_{oq}}{r_{o1}}\right)^2}$$

estimate $\frac{\sigma_{\Delta g_m}}{g_m}$

$$\frac{\Delta g_m \pi}{g_m} = \frac{1}{g_m} \frac{\partial g_m}{\partial V_T} + \frac{1}{g_m} \frac{\partial g_m}{\partial K} =$$

$$dV_T = d(V_{ov} - V_T) = -\Delta V_T$$

$$= \frac{1}{2\mu V_{ov}} \cdot 2\mu V_{ov} dK + \frac{1}{2\mu V_{ov}} \cdot 2\mu dV_T$$

$$= \frac{\Delta K}{K} - \frac{\Delta V_T}{V_{ov}}$$

$$\sigma^2 \left(\frac{\Delta g_m}{g_m} \right) = \sigma^2 \left(\frac{\Delta K}{K} \right) + \frac{\sigma^2(\Delta V_T)}{V_{ov}^2}$$

statistically independent but in reality they are correlated

(considered uncorrelated for simplicity)

18) Noise: PSD, thermal noise on resistors, MOSFETs

We found out that gain/BW tradeoff is independent from the current ($g_{mV_{ov}} = \frac{2VA'}{V_{ov}}$ and $f_T = \frac{\mu V_{ov}}{2\pi L^2}$). What sets

the current? Noise. For this discussion, consider gaussian noise:

$$p(\bar{v}) = \frac{1}{\sqrt{2\pi}\sigma} e^{-\frac{\bar{v}^2}{2\sigma^2}} \quad p(\bar{v}) \text{ is the probability of the amplitude}$$

$\sim 68\%$ of samples will fall within $\pm \sigma$.

We can describe noise as a superposition of orthogonal harmonics (like signals). Consider just two:

$$X(t)|_{\text{noise}} = A \sin(\omega_1 t + \phi_1) + B \sin(\omega_2 t + \phi_2)$$

- Mean value is zero
- Mean square value is

$$\langle X(t)^2 \rangle = \langle A^2 \sin^2(\omega_1 t + \phi_1) + B^2 \sin^2(\omega_2 t + \phi_2) + 2AB \sin(\omega_1 t + \phi_1) \sin(\omega_2 t + \phi_2) \rangle$$

Therefore:

$$\begin{aligned} \langle X(t)^2 \rangle &= \langle A^2 \sin^2(\omega_1 t + \phi_1) \rangle + \langle B^2 \sin^2(\omega_2 t + \phi_2) \rangle + \langle 2AB \sin(\omega_1 t + \phi_1) \sin(\omega_2 t + \phi_2) \rangle \\ &= \frac{A^2}{2} + \frac{B^2}{2} \quad \rightarrow \text{Recall } A^2 \cdot \frac{1}{T} \int_0^T \sin^2\left(\frac{2\pi t}{T}\right) dt = \frac{A^2}{2} \end{aligned}$$

orthogonal

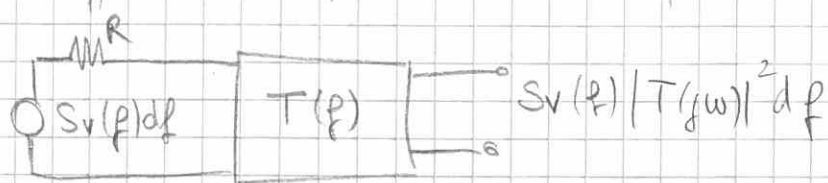
Since mean value is zero: $\langle X(t)^2 \rangle = \sigma_{x_n}^2$

If we consider a set of sinusoids (different freq/amplitude)

$$\sigma^2 = \sum_i \sigma_i^2 = \int_0^{+\infty} S_n(f) df \quad \text{where } S_n(f) \text{ is the PSD}$$

For a small frequency BW Δf we can say $S_n(f_0) \Delta f = \Delta \sigma_{f_0}^2$

Suppose that a resistor is generating noise through a filtering network, the mean square value contribution will be



We can say the same for every component inside a linear network, leading to:

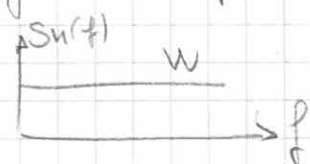
$$S_{out}(f) df = S_{n1}(f) |T_1(j\omega)|^2 df + S_{n2}(f) |T_2(j\omega)|^2 df + \dots$$

$$\sigma_{out} = \sqrt{\int_0^{+\infty} S_{out}(f) df}$$

Thermal noise in resistors

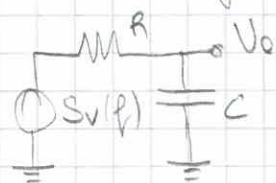
Charge movement and collision with ions is fast ($< 1ps$).

This means that the spikes generated are short \rightarrow short signal (impulse) means broad spectrum \rightarrow approx to flat



$$S_n(f) = W \quad W \text{ stands for white}$$

To estimate W value we can use an energetic argument by filtering the noise with a capacitor



$$V_o(s) = V_i(s) \cdot \frac{1}{1 + sRC}$$

Since $S_v(f) = W$ and we said $\langle V_o^2 \rangle = \sigma_o^2 = W \int_0^{+\infty} |T(j\omega)|^2 d\omega$

$$\int_0^{+\infty} |T(f)|^2 df = \int_0^{+\infty} \frac{1}{1 + (2\pi fRC)^2} df \cdot \frac{2\pi R}{2\pi R} = \frac{1}{2\pi R} \left[\arctan(2\pi fRC) \right]_0^{+\infty} =$$

$$\frac{1}{2\pi R} \cdot \frac{\pi}{2} = \frac{1}{4R} \quad \text{Therefore } \sigma_o^2 = \frac{1}{4R} \cdot W$$

If we express this in frequency w/c off $\frac{1}{4R} = \frac{\pi}{2} BW_{-3dB}$

So the ENBW = equivalent noise BW is $\frac{\pi}{2}$ higher the pole w/c off $f_{cut} = \frac{1}{2\pi R}$

It's higher because it accounts for the additional integrated noise over f_{cut}

Energy stored in the capacitor is

$$\frac{1}{2} C \langle v_0^2 \rangle = \frac{1}{2} k_B T \rightarrow \frac{1}{2} \text{ for a single degree of freedom}$$

according to Boltzmann:

$$\frac{1}{2} C \cdot \frac{W}{kT} = \frac{1}{2} k_B T \quad \frac{Q \cdot W}{4 R k} = k_B T \rightarrow \underline{W = 4 k_B T R}$$

$$W = S_v(f) |R = \left(\frac{V}{\sqrt{Hz}}\right)^2 \text{ or } \left(\frac{A}{\sqrt{Hz}}\right)^2 \text{ if current referred}$$

$$S_i(f) |R : \text{ [Circuit: voltage source } S_v \text{ in series with resistor } R \text{ and load } S_i \text{]} \rightarrow S_i(f) = 4 k_B T R \cdot \left(\frac{1}{R^2}\right)^2 \left|T(j\omega)\right|^2$$

$S_i(f) |R = \frac{4 k_B T}{R}$

Same happens for the resistive of MOSFETs in ohmic region:

$$R_{ch} = \frac{1}{g_{m0}} \text{ where } g_{m0} = k (V_{as} - V_T) = 2 k V_{ov}$$

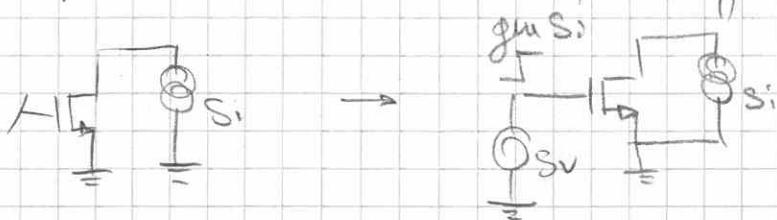
$$\text{So } S_v(f) |_{MOS} = 4 k_B T g_m \rightarrow \text{ohmic}$$

For MOSFETs in saturation region, the resistive channel is shorter and less uniform \rightarrow compensate with factor f

$$\underline{S_i(f) |_{MOS}} \begin{cases} \underline{4 k_B T f g_m} \rightarrow \underline{OHMIC} \\ \underline{4 k_B T \frac{2}{3} g_m} \rightarrow \underline{SAT} \\ \underline{4 k_B T 2 g_m} \rightarrow \underline{SAT (short channel L)} \end{cases}$$

The factor is expressed using $f = 1, \frac{2}{3}, 2$

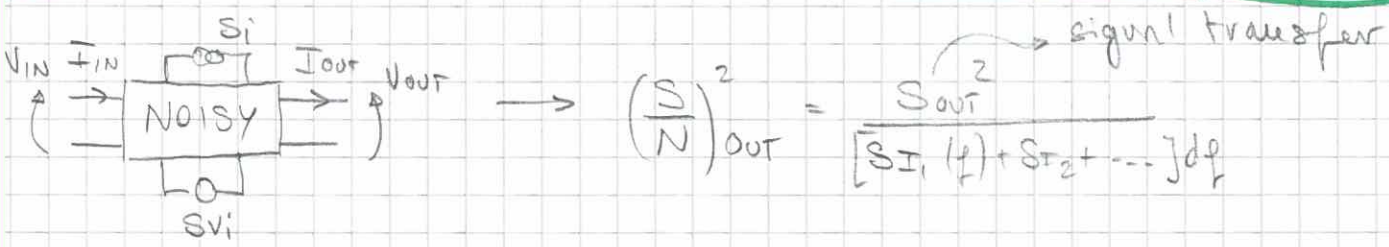
$S_v(f)$ is also available by using the following:



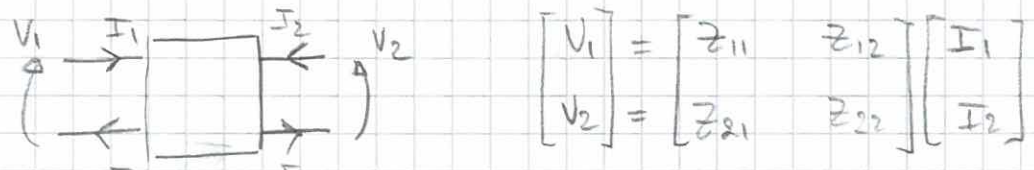
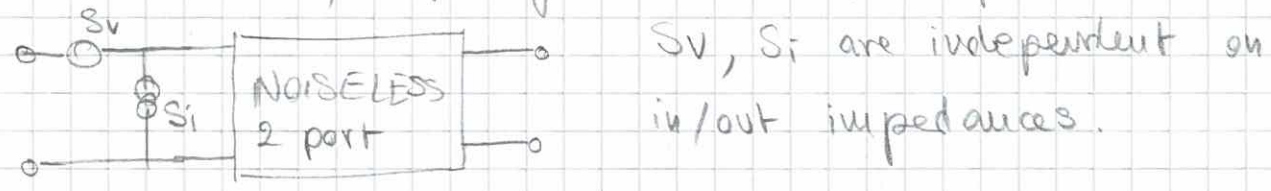
$S_v(f) |_{MOSFET} = \frac{4 k_B T f}{g_m}$

Note: see next question for a better procedure for S_v

9) Input referred noise sources (2 part) + Diff stage noise



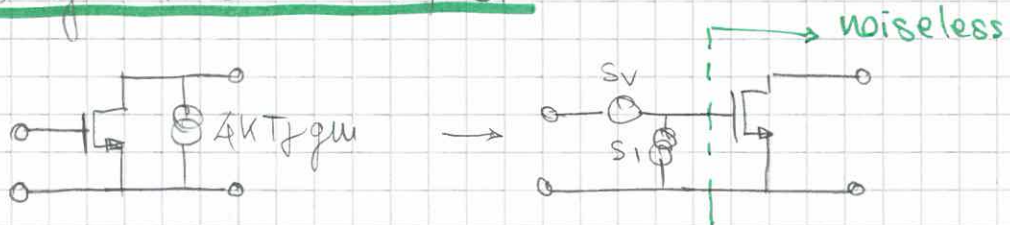
We can model the noisy network as a two port noiseless network + noisy input generators (input referred noise)



Definition of two part network

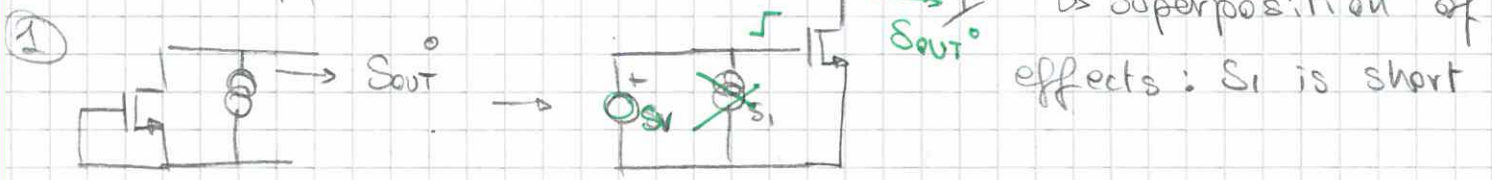
A two part network can be represented with the Z matrix

Single MOSFET S_v, S_i

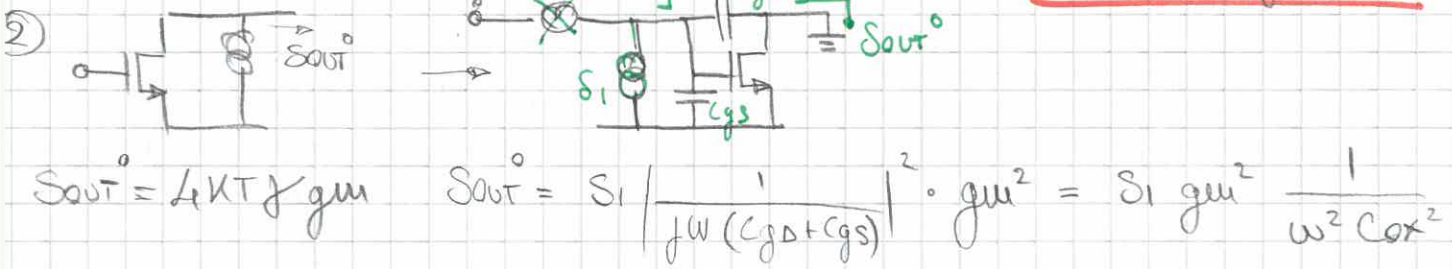


Its easy to see that
 short input $\rightarrow S_v \rightarrow$ ①
 open input $\rightarrow S_i \rightarrow$ ②

Let's clarify:



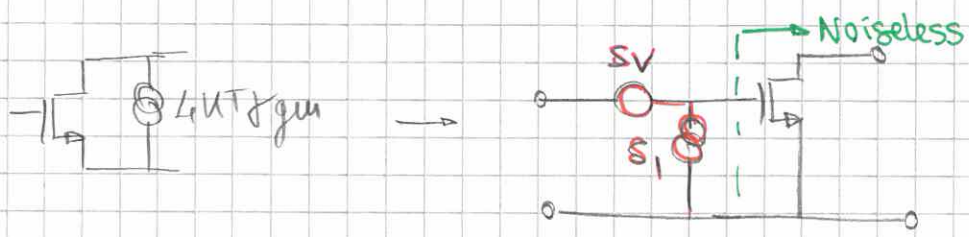
$S_{OUT}^0 = 4kT/g_m$ $S_{OUT}^0 = g_m^2 S_{VIN}$ $\rightarrow S_{VIN} = \underline{\underline{\frac{4kT/g_m}{g_m^2}}}$



$$4kTf gm = S_1 gm \frac{1}{\omega^2 C_{ox}^2} \rightarrow S_1 = 4kTf gm \left(\frac{\omega}{\omega_T} \right)^2$$

Where ω_T is the cutoff $\omega_T = \frac{gm}{C_{ox}}$

Recap on input referred noise for a MOSFET



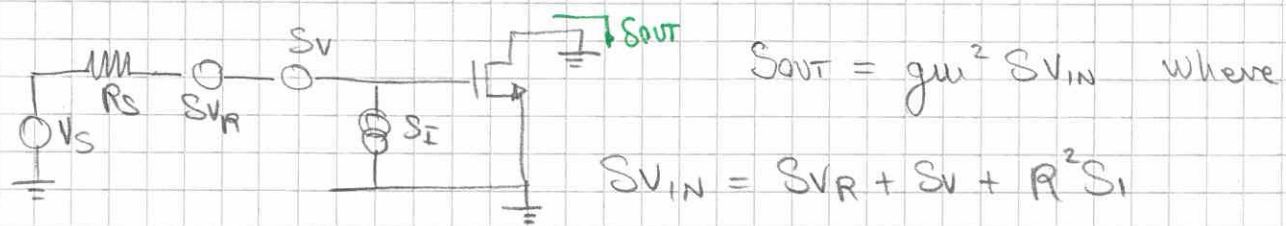
$$S_{V|IN} = \frac{4kTR}{gm} \quad S_{I|IN} = 4kTf gm \left(\frac{\omega}{\omega_T} \right)^2$$

Note: $\left(\frac{\omega}{\omega_T} \right)$ is a rising factor because since $\frac{1}{s C_{ox}}$

$Z_{IN} = \frac{1}{s C_{ox}}$ is decreasing with frequency, to get a flat

output spectrum ($4kT gm$), S_1 has to compensate by increasing its power amplitude.

Example of application

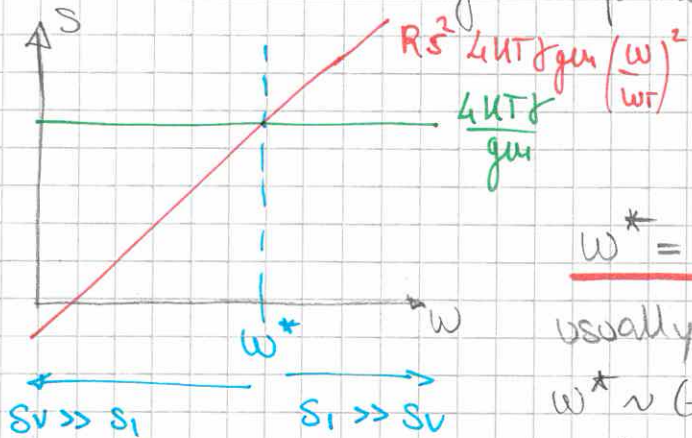


$$S_{out} = gm^2 S_{VIN} \text{ where}$$

$$S_{VIN} = S_{VR} + S_V + R_s^2 S_I$$

$$S_{VR} = 4kTR_s \quad S_V = \frac{4kTf}{gm} \quad S_I = 4kTf gm \left(\frac{\omega}{\omega_T} \right)^2$$

Be careful:
 -> low impedance $R_s \rightarrow S_V$ is relevant
 -> high impedance $R_s \rightarrow S_I$ is relevant



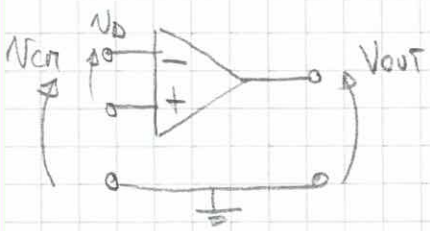
$$\frac{4kTf}{gm} = R_s^2 4kTf gm \left(\frac{\omega}{\omega_T} \right)^2$$

$$\omega^* = \omega_T / gm R_s \text{ but since } \omega_T \text{ is}$$

usually very high and R_s very low $\omega^* \sim \text{GHz} \rightarrow$ we can almost always neglect S_I contribution (for $\sim \text{MHz}$) 70

Input referred noise for a differential stage

Important: is the differential stage a 2 port network?



It's not, but it can be if we don't consider V_{CM} . This means that $C_{RR} \rightarrow \infty$. Let's be more specific

$$\underline{V_{OUT} = A^+ V^+ - A^- V^-} \quad \underline{V_D = V^+ - V^-} \quad \underline{V_{CM} = \frac{V^+ + V^-}{2}} \quad \text{so}$$

$$V^+ = V_{CM} + \frac{V_D}{2} \quad V^- = V_{CM} - \frac{V_D}{2}$$

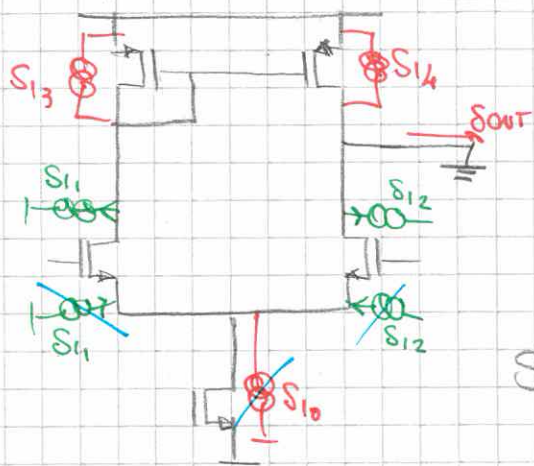
$$V_{OUT} = A^+ \left(V_{CM} + \frac{V_D}{2} \right) - A^- \left(V_{CM} - \frac{V_D}{2} \right) = V_{CM} (A^+ - A^-) + V_D \left(\frac{A^+ + A^-}{2} \right)$$

• If $A^+ = A^-$ $V_{CM} (A^+ - A^-) = 0$ therefore:

$$\underline{V_{OUT} = V_D \left(\frac{A^+ + A^-}{2} \right) = A \cdot V_D} \rightarrow \text{it's a two port network}$$

In reality $C_{RR} \sim 100\text{dB}$ to make those assumptions

Output noise current of a differential stage



Important! CMRR $\rightarrow \infty$

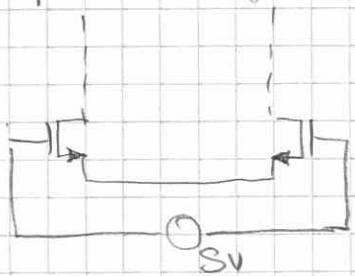
Note: all the generators connected to the source give a common mode contribution \rightarrow No output noise current

$$S_{out}^0 = S_{11} + S_{12} + S_{13} + S_{14}$$

$$= 8kT \gamma (g_{m_{in}} + g_{m_{II}})$$

$$\underline{S_{out}^0 = 8kT \gamma (g_{m_{in}} + g_{m_{II}})}$$

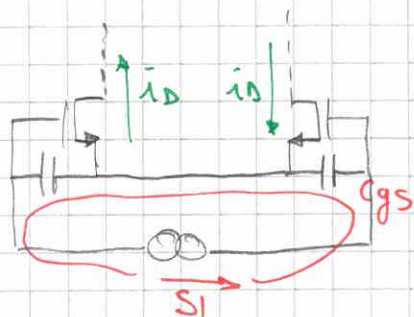
Input voltage referred noise / current referred noise



$S_{out}^0 = S_v \cdot g_{m_{in}}^2 \rightarrow$ differential transfer function

$$S_v |_{nos} = \left(\frac{1}{g_{m_{in}}} \right)^2 8kT \gamma (g_{m_{in}} + g_{m_{II}}) =$$

$$\underline{S_v |_{nos} = \frac{8kT \gamma}{g_{m_{in}}} \left(1 + \frac{V_{ov_{in}}}{V_{ov_{II}}} \right)}$$



We can see that $i_D = g_m V_{gs} = g_{m_{in}} \frac{i_{in}}{s C_{gs}}$
 V_{gs} is the voltage on C_{gs}

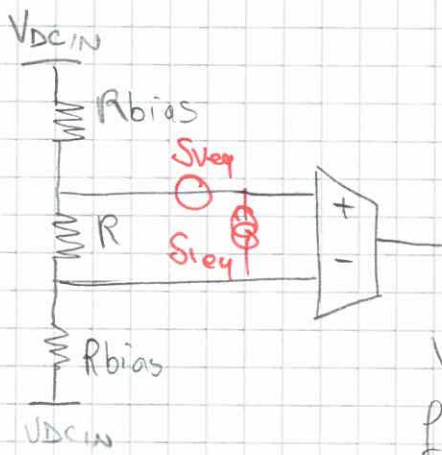
$$i_{in, out} = 2 g_m \frac{i_{in}}{s C_{gs}} \rightarrow S_{out}^0 = \frac{4 g_m^2}{\omega^2 C_{gs}^2} S_1$$

by setting $\omega_T = \frac{g_m}{C_{ox}} \sim \frac{g_m}{C_{gs}}$ with minimal error we

can say $S_{out}^0 = 4 S_1 \left(\frac{\omega}{\omega_T} \right)^2 \rightarrow$ if we compare S_1, S_v :

$$S_{v_{in}} \cdot g_{m_{in}}^2 = 4 S_1 \left(\frac{\omega}{\omega_T} \right)^2 \rightarrow \underline{S_{1eq} = S_{veq} \frac{g_{m_{in}}^2}{4} \left(\frac{\omega}{\omega_T} \right)^2}$$

Current, voltage noise comparison



$$S = S_{veg} + S_{req} R_s^2$$

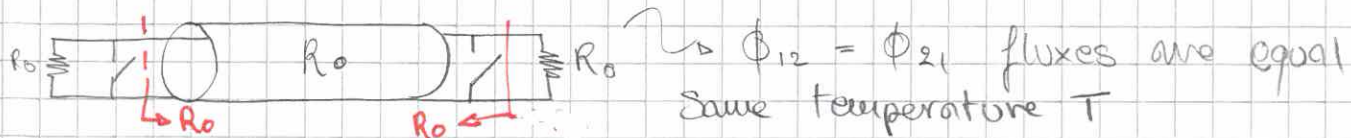
$$\rightarrow S_{veg} \frac{g_{mIN}^2}{4} \left(\frac{\omega}{\omega_T} \right)^2$$

Voltage noise is dominant up to

$$f = \frac{2 f_T}{g_{mIN} R_s} \text{ so, unless for very}$$

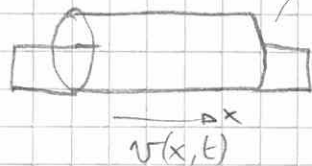
high frequency (or very high impedances) \rightarrow S_I negligible

18) Noise models: Nyquist experiment for PSD



Use a matched coax cable. For $t = -\infty$ system will be at equilibrium. We set up matched impedances \rightarrow no reflections take place therefore the same energy (electric/magnetic field) is stored in the cable.

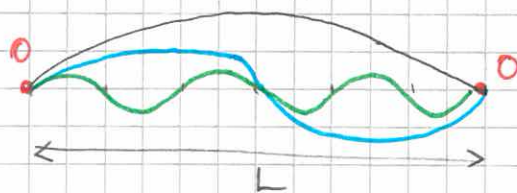
At $t = 0^+$ we short the two ends so we separate the R_0 from the coax. Some e.m. waves will be trapped inside. Also, the energy E_0 of the resistor will be twice inside the cable.



\rightarrow e.m. system with the following equations

$$\frac{\partial V(x,t)}{\partial x^2} = \frac{1}{c^2} \frac{\partial^2 V(x,t)}{\partial t^2} \quad \frac{\partial i(x,t)}{\partial x^2} = \frac{1}{c^2} \frac{\partial^2 i(x,t)}{\partial t^2}$$

Since the system has boundaries, surviving modes will be:

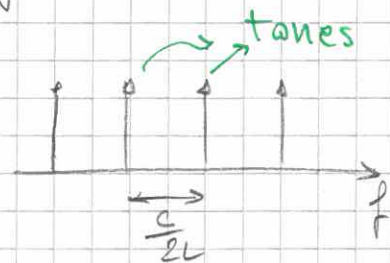


first mode is $L = \frac{\lambda_1}{2}$ using the dispersion relationship $f_1 \cdot \lambda_1 = c$

$$f_1 = \frac{c}{2L}$$

We find ∞ modes trapped with a wavelength that is a multiple of the length of the coax:

$$f_k = \frac{c}{2L} \cdot k \quad \text{every tone is spaced by } \frac{c}{2L}$$



If we select a Δf bandwidth, can count

the number of modes falling inside $\# \text{ modes} = \frac{\Delta f}{c/2L}$

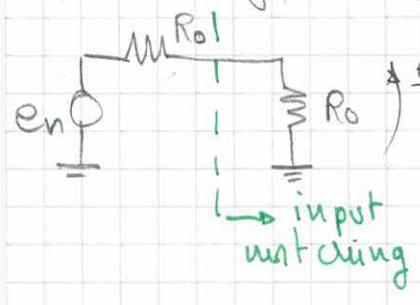
Each mode has electrical/magnetic degrees of freedom

$\rightarrow \frac{KT}{2} \cdot 2 = KT$ We can say that the average energy

trapped in a single interval is $E_{\Delta f} = KT \cdot \frac{\Delta f}{c} \cdot 2L$

$$E_{\Delta f} = kT \cdot \frac{\Delta f}{c} \cdot 2L \quad \left[\begin{array}{l} \rightarrow \text{number of modes} \\ \rightarrow \text{energy per single mode} \end{array} \right]$$

We can now connect the trapped energy with the input energy supplied. Consider a sinusoidal generator e_n :



Power delivered to coax is

$$P = \left(\frac{e_n}{2} \cdot \frac{1}{\sqrt{2}} \right)^2 \cdot \frac{1}{R_0} = \left(\frac{e_n}{2} \right)^2 \cdot \frac{1}{2R_0} \quad \left[\frac{\text{double}}{\text{seconds}} \right]$$

e_n is the peak voltage of gen

We now need to compute the energy, therefore we need to compute how much time it takes to travel:

$T = \frac{L}{c}$ where $T = \text{transit time}$ $L = \text{length of coax}$
 $c = \text{speed of light}$

$$E_{\text{TOT}} |_{\text{supplied}} = 2 \cdot T \cdot P = 2 \cdot \frac{L}{c} \cdot \left(\frac{e_n}{2} \right)^2 \cdot \frac{1}{2R_0}$$

↓ double contribution
↓ transit
↓ power

Since e_n is the voltage noise generator, we link it to the mean square value

$$\frac{e_n^2}{2} = S_V \Delta f$$

$$\langle e_n(t) \rangle^2 = \frac{e_n^2}{2}$$

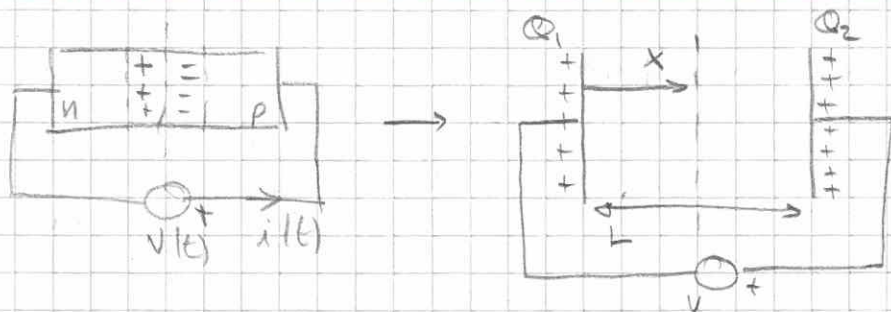
Therefore we "linked" the power of a Δf in a single sinusoid. We can now write

$$E_{\text{TOT}} |_{\Delta f} = \frac{L}{c} \left(\frac{e_n}{2} \right)^2 \cdot \frac{1}{R_0} = \frac{L}{c} \frac{e_n^2}{2} \cdot \frac{1}{2} \cdot \frac{1}{R_0} = \frac{S_V \Delta f}{2} \cdot \frac{L}{c} \cdot \frac{1}{R_0}$$

This must be equal to the stored $E_{\Delta f}$:

$$\frac{1}{2} \cdot \frac{1}{R_0} \cdot \frac{S_V \cdot \Delta f}{2} = kT \cdot \frac{\Delta f}{c} \cdot 2L \rightarrow \underline{S_V = 4kTR_0}$$

20) Shot noise in PN junctions + weak inversion MOSFETS



The electron moving inside the PN junction behaves like in a capacitor with plane plates:

$$Q_2 = \frac{x}{L} q \quad Q_1 = \left(\frac{L-x}{L}\right) q \quad Q_1 + Q_2 = q \approx \text{charge of the electron}$$

$\downarrow > 0$ $\downarrow > 0$

The transit of the electron generates a current

$$\left| \frac{dQ_2}{dt} \right| = \left| \frac{dQ_1}{dt} \right| = \frac{q}{L} \frac{dx}{dt} = \frac{q}{L} v(t) = I_{\text{CARRIER}}(t) \quad \left[\frac{C}{s} \right]$$

\downarrow carrier velocity

The average flow of electrons is $\lambda = \frac{\text{electrons}}{s} = I/q$

Therefore in a time interval dt the probability that an electron starts to flow is λdt



Suppose the carriers travel through the junction with saturation velocity. The current contribution will have a rectangular shape. This means that, for a single e^- :

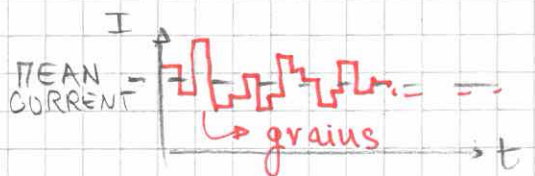
$$i(t) = q h(t) \quad \text{where} \quad \int_0^{+\infty} h(t) dt = 1$$

We can say that total current will be a sum of elementary grains (pulses):

$$i(t) = qh(x_1) + qh(x_2) + \dots$$

\hookrightarrow arbitrary axis to take

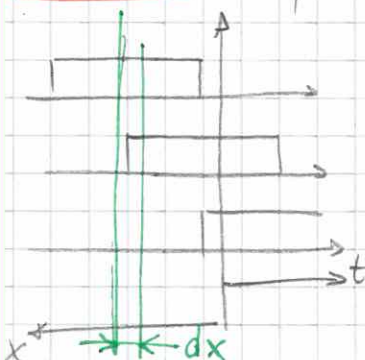
into account the shape during the integration.



We now compute the mean value and the mean square

$$\langle i(t) \rangle = \langle qh(x_1) + qh(x_2) + \dots \rangle = \int_0^{+\infty} qh(x) \cdot \lambda dx = q\lambda \int_0^{+\infty} h(x) dx = q\lambda$$

↑ continuous
↳ probability of a pulse to start in a dx



It makes sense, the average current is given by the rate per unit time multiplied by the carrier charge $\langle i(t) \rangle = q\lambda$

$$\langle i(t)^2 \rangle = \langle (qh_1 + qh_2 + \dots)^2 \rangle$$

$$= \langle (qh_1)^2 + (qh_2)^2 + (qh_3)^2 + 2q^2 h_1 h_2 + 2q^2 h_2 h_3 + 2q^2 h_1 h_3 + \dots \rangle$$

↳ single squared contributions
↳ biproducts

Pass from discrete to continuous

$$\langle i^2(t) \rangle = q^2 \int_0^{+\infty} h^2(x) \lambda dx + q^2 \iint_0^{+\infty} h_1(x) h_2(y) \lambda^2 dx dy$$

$$= q^2 \int_0^{+\infty} h^2(x) \lambda dx + q^2 \int_0^{+\infty} h_1(x) \lambda dx \int_0^{+\infty} h_2(y) \lambda dy$$

$\int_0^{+\infty} h(x) dx = 1$

$$= q^2 \lambda \int_0^{+\infty} h^2(x) dx + q^2 \lambda^2 = \langle i^2(t) \rangle \quad \text{from statistical theory:}$$

$$\sigma_i^2 = \langle i^2(t) \rangle - \langle i(t) \rangle^2 = q^2 \lambda \int_0^{+\infty} h^2(x) dx + q^2 \lambda^2 - (q\lambda)^2$$

Therefore $\sigma_i^2 = q^2 \lambda \int_0^{+\infty} h^2(x) dx = qI \int_{-\infty}^{+\infty} |H(f)|^2 df$

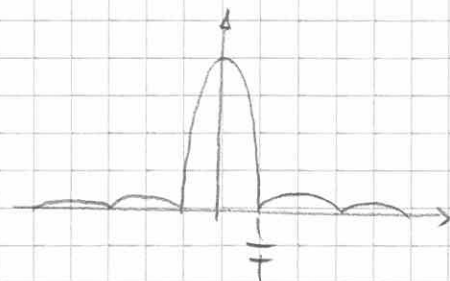
↑ parseval
↳ $h^2(x) = 0$ for $x < 0$

Since $\sigma_i^2 = \int_{-\infty}^{+\infty} S_I(f) df \rightarrow S_I(f) = 2qI |H(f)|^2$

Factor 2 is given by $\textcircled{*2}$ and $\textcircled{*3}$ because $H(f)$ is symmetric thus giving $\int_{-\infty}^{+\infty} = 2 \int_0^{+\infty}$

$$S_I(f) = 2q I |H(f)|^2$$

where, for a rect it is $|H(f)| = \left| \frac{\sin 2\pi f/T}{2\pi f/T} \right|$



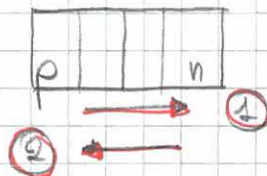
Since T is very short (few ps) we can say that $\frac{1}{T}$ will be placed at ~ 100 GHz so we can basically say that S_I is \sim white unless some special cases

Shot noise on a PN junction

$$I = I_s e^{\frac{qV}{kT}} - I_s$$

\rightarrow p \rightarrow n contribution (diffusion) ①

\rightarrow n \rightarrow p contribution (minority carriers) ②



$I_{DIFFUSION} = I + I_s$ $I_{REVERSE} = -I_s$ Since I_D and I_R are independent on each other, they sum up in spectrum

$$S_I = 2q (I + I_s) + 2q I_s \quad \text{where}$$

- Reverse bias $I_{DIFF} = 0 \rightarrow S_I = 2q I_s$
- Forward bias $I_{DIFF} \gg I_{REV} \rightarrow S_I = 2q I$
- Zero bias $I = 0 \rightarrow S_I = 4q I_s$

$$\frac{I_s}{V_{TH}}$$

Note: at zero bias $g_{D0} = \left. \frac{\partial I}{\partial V} \right|_{V=0} = \left[\frac{q I_s}{kT} e^{\frac{qV}{kT}} \right]_{V=0} = \frac{q I_s}{kT}$

$$\text{So } I_s = g_{D0} \cdot \frac{kT}{q} \rightarrow \underline{S_I = 4q I_s = 4kT g_{D0}}$$

This is very familiar to a resistor white noise. In fact, we can apply the same reasoning of the Nyquist experiment:



Weak inversion MOSFET shot noise

$$I_{\text{DRAIN}} = I_0 e^{\frac{q V_{\text{GS}} - V_T}{n k T}} \rightarrow g_m = \frac{\partial I_D}{\partial V_{\text{GS}}} = \frac{q I_0 e^{\frac{q V_{\text{GS}} - V_T}{n k T}}}{n k T} = \frac{q I_D}{n k T}$$
$$n g_m \frac{k T}{q} = I_D \rightarrow \underline{S_I} = 2 q I_D = 2 k T n g_m$$

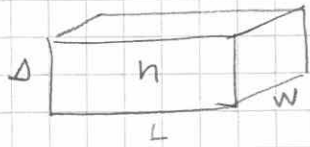
We can adjust the formula to $4 k T \gamma g_m$:

S_I \rightarrow $\gamma = \frac{2}{3}$ saturation (2 for short channel)

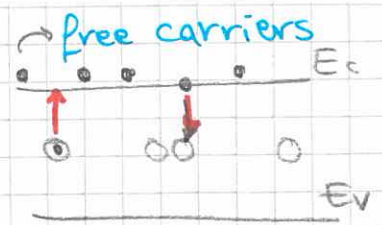
\rightarrow $\gamma = 1$ ohmic

\rightarrow $\gamma = \frac{n}{2}$ weak inversion (shot noise)

21) Trapping noise in a resistor



$$I = G \cdot V = q \mu n \frac{W \Delta}{L} \cdot V$$



$N = \# e^-$ contributing to current flow = $n \cdot \Delta \cdot W \cdot L$ so
 ↳ free carriers ↳ $\# e^- / \text{cm}^3$

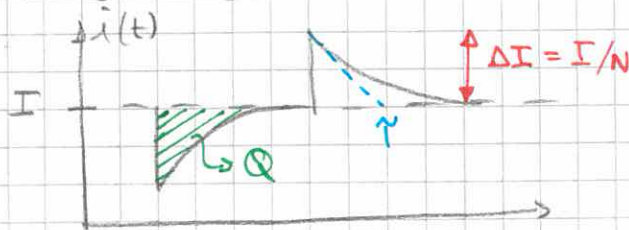
$I = q \cdot \mu \cdot \frac{N}{L^2} V$ we can see that $I \propto N$ for a given

voltage, so if some electrons are trapped inside the material, I will decrease $\rightarrow \Delta I / I = \Delta N / N$

↳ single electron gets captured/emitted $\rightarrow \Delta N = |N - (N \pm 1)|$

If $\Delta N = 1$ the current variation for a capture/release event will be $\Delta I = \frac{I}{N}$. At steady state, the emission/capture processes are in a dynamic equilibrium (if captures \nearrow then emissions \nearrow to return to steady state and vice versa)

Transients recover with a time constant τ



Waveforms are

$$i(t) = \Delta I \cdot e^{-\frac{t}{\tau}}$$

$i(t) = Q$ (area) $h(t) = (\Delta I \cdot \tau) \frac{1}{\tau} e^{-\frac{t}{\tau}}$ where $\int_0^{\infty} \frac{1}{\tau} e^{-\frac{t}{\tau}} dt = 1$

Area of the pulse is $Q = \frac{I}{N} \cdot \tau$

↳ Just for capture or emission events

Knowing that $\langle i(t) \rangle = \langle Q h_1(t) + Q h_2(t) + \dots \rangle$ we can recover the same shot noise reasoning and end up

with $S_I = 2 q^2 \lambda |H(\omega)|^2$

At equilibrium the rate of emission and capture will be the same $\rightarrow \lambda_e = \lambda_c = \lambda$ and $H(\omega)$ has a LPF like transfer function, therefore

$$S_I = 2 Q^2 \lambda \frac{1}{(1 + \omega^2 \tau^2)} = 2 \left(\frac{I}{N} \right)^2 \tau^2 \cdot \lambda \cdot \frac{1}{1 + \omega^2 \tau^2}$$

We need to estimate λ , since λ is a rate of events, we can say $\lambda = \frac{1}{\tau} N_T \cdot \beta$ where

N_T = number of traps in the volume

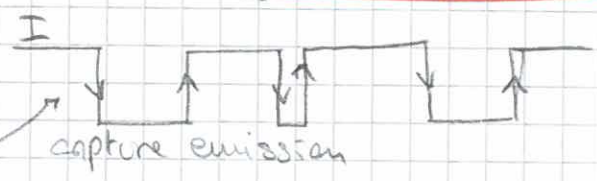
β = proportionality factor that adjusts N_T/τ

$$S_I = 2 \cdot 2 \left(\frac{I}{N}\right)^2 \frac{N_T \beta}{\tau} \tau^2 \frac{1}{1 + \omega^2 \tau^2}$$

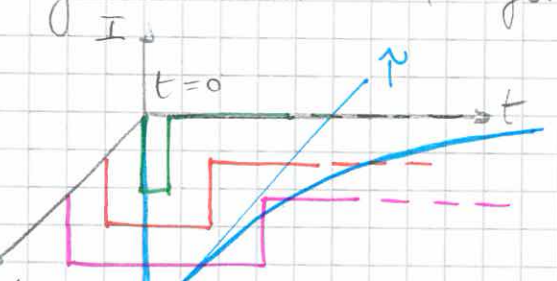
↳ double because of capture + emission contribution

$S_I = 4 \left(\frac{I}{N}\right)^2 \beta N_T \frac{\tau}{1 + \omega^2 \tau^2}$ → Random Telegraph Noise

Measurements show a binary change:



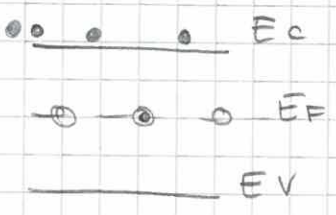
Recovery time should be a step (not exponential) because a single electron can't generate an exp current variation, but if we look to all concurrent events:



We discover that the sum of all events will merge into one exponential transient →

↳ all concurrent shapes 1st order model works well!

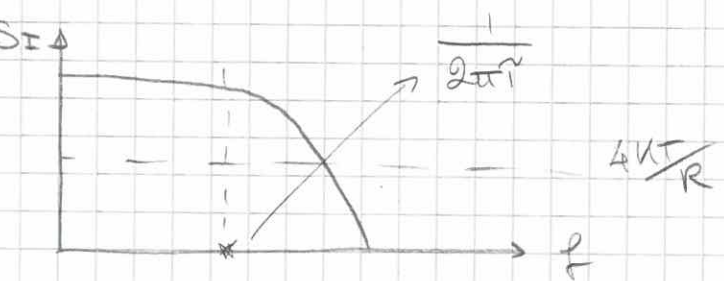
If we limit ourselves to just a single layer of traps placed at Fermi level → $\beta = 1/4$ for EF



for a more rigorous derivation, it is

$$\tau = \frac{\tau_e \tau_c}{\tau_e + \tau_c} \quad \beta = \frac{\tau_e \tau_c}{(\tau_e + \tau_c)^2} \approx \frac{1}{4} \text{ at } E_{FERRI}$$

Spectrum will be a Lorentzian shape:

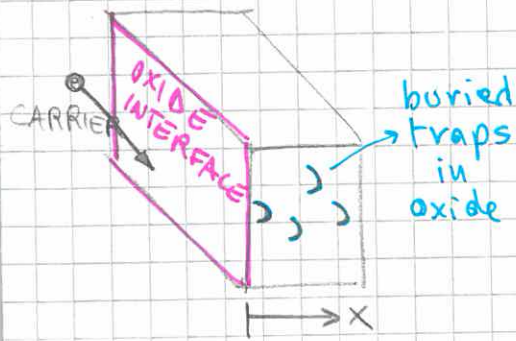


for $\beta = 1/4$

$S_I = N_T \left(\frac{I}{N}\right)^2 \frac{\tau}{1 + \omega^2 \tau^2}$

22) McWorther model for $\frac{1}{f}$ + Tsvidis formula

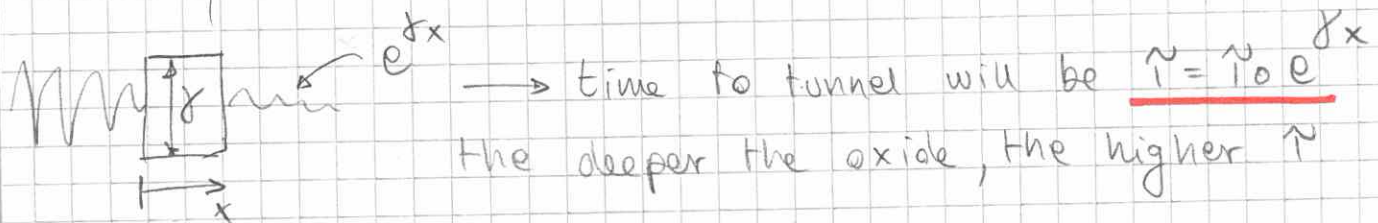
For now we considered only a set of traps with a single time constant. But there can be defects with a different τ . In fact, McWorther pointed out that carriers flowing in the MOSFET channel can also be trapped in deeper layers of the oxide.



We expect S_I to be a superposition of many Lorentian shapes with various τ . We said $S_I = N_T \left(\frac{I}{N}\right)^2 \frac{\tau}{1 + \omega^2 \tau^2}$ but now τ Traps at Fermi level

We need to take into account the trapping centers with $g(\tau) d\tau =$ fraction of N_T trapping centers at Fermi level

level with a time constant between τ and $\tau + d\tau$. From quantum mechanics, we know that:



$g(\tau)$ represents the distribution of trap numbers for a given τ , therefore:

$$dN_T(\tau) = N_T g(\tau) d\tau \rightarrow \text{We need to link the distribution } g(\tau) \text{ to the spatial dimension } x$$

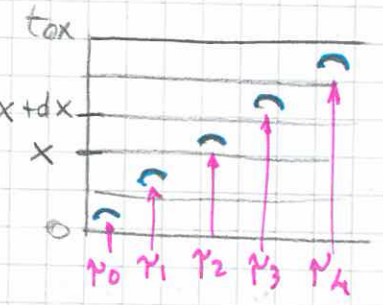
$\underbrace{dN_T(\tau)}_{\text{elementary trap centers}} = \underbrace{N_T}_{\text{total traps}} \underbrace{g(\tau) d\tau}_{\text{portion of traps in a } d\tau}$

The spectrum changes now form to:

$$S_I = N_T \left(\frac{I}{N}\right)^2 \int_{\tau_{\min}}^{\tau_{\max}} \frac{\tau}{1 + \omega^2 \tau^2} g(\tau) d\tau$$

We consider that traps are spread on the whole oxide thickness.

So the whole N_T will be spread between $x=0$ and $x=t_{ox}$.
 The number of traps contained in a dx segment will be:
 $N_T \frac{dx}{t_{ox}} = N_T g(\tilde{\nu}) d\tilde{\nu}$ → clarify this *1



If the traps are evenly distributed through the whole t_{ox} :

$$\tilde{\nu}_1 = \tilde{\nu}_0 e^{\delta x_1} \quad \tilde{\nu}_2 = \tilde{\nu}_0 e^{\delta x_2} \quad \tilde{\nu}_3 = \tilde{\nu}_0 e^{\delta x_3}$$

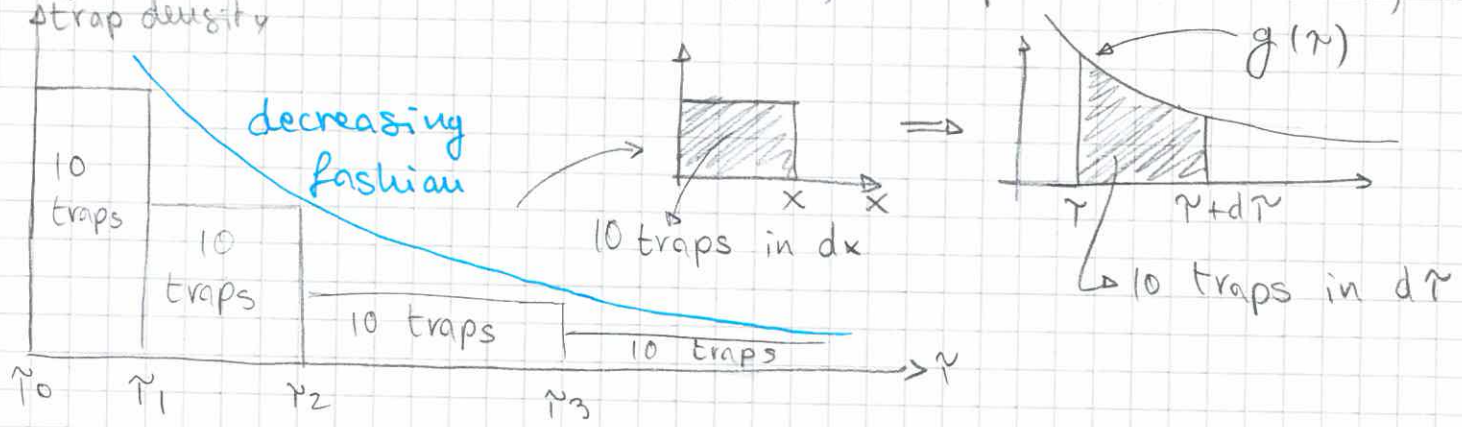
↳ $\tilde{\nu}$ for $x=0$

If *1 then $x_2 = 2x_1$ and $\tilde{\nu}_2 = \tilde{\nu}_0 e^{2\delta x_1}$ so

$$\frac{\tilde{\nu}_2}{\tilde{\nu}_1} = \frac{e^{2\delta x_1}}{e^{\delta x_1}} = e^{\delta x_1} \rightarrow \text{Same result for } \frac{\tilde{\nu}_3}{\tilde{\nu}_2} \text{ and so on}$$

Therefore the same portion of traps will be mapped to an exponentially increasing portion of $\tilde{\nu}$.

e.g. 10 traps are between $\tilde{\nu}_0, \tilde{\nu}_1$; 10 traps are between $\tilde{\nu}_1, \tilde{\nu}_2$

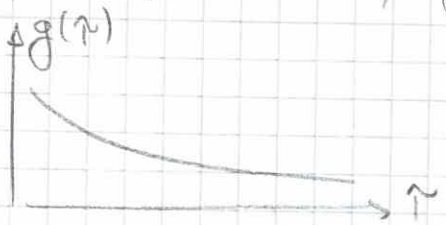


We confirmed $N_T \frac{dx}{t_{ox}} = N_T g(\tilde{\nu}) d\tilde{\nu} \rightarrow g(\tilde{\nu}) = \frac{dx}{d\tilde{\nu}} \cdot \frac{1}{t_{ox}}$

Since $\tilde{\nu} = \tilde{\nu}_0 e^{\delta x} \quad d\tilde{\nu} = \delta \tilde{\nu}_0 e^{\delta x} dx = \delta \cdot \tilde{\nu} dx \rightarrow \frac{dx}{d\tilde{\nu}} = \frac{1}{\delta \tilde{\nu}}$

Therefore $g(\tilde{\nu}) = \frac{1}{\delta \tilde{\nu} t_{ox}}$ we can clearly see that

the trap density for a given $\tilde{\nu}$ decreases (hyperbolic form)



Using this result in the integral:

$$S_I(f) = \frac{N_T}{f \tau_{ox}} \left(\frac{I}{N}\right)^2 \int_{\tau_{min}}^{\tau_{max}} \frac{d\tau}{1 + \omega^2 \tau^2}$$

τ_{max} = traps deeper into oxide
 τ_{min} = traps on the interface

$$= \frac{N_T}{f \tau_{ox} \omega} \left(\frac{I}{N}\right)^2 \left[\arctg(\omega \tau_{max}) - \arctg(\omega \tau_{min}) \right]$$

τ_{max} = longest time scale measurable by the experiment

τ_{min} = shortest time constant that can be measured by instruments

The experimental observation window corresponds to

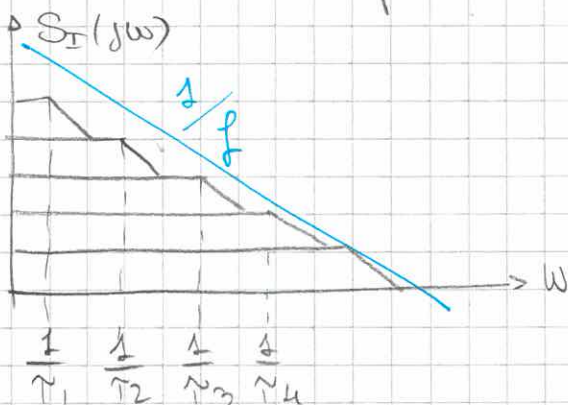
$$\frac{1}{\tau_{max}} < \omega < \frac{1}{\tau_{min}} \rightarrow \omega \tau_{max} \gg 1 \quad \omega \tau_{min} \ll 1 \rightarrow \text{so}$$

measurable frequencies

$$\left[\arctg(\omega \tau_{max}) - \arctg(\omega \tau_{min}) \right] \approx \frac{\pi}{2} - 0 \quad \text{Therefore}$$

$$\underline{S_I(f)} = \frac{N_T}{f \tau_{ox}} \left(\frac{I}{N}\right)^2 \cdot \frac{1}{\omega} \cdot \frac{\pi}{2} = \underline{\frac{N_T}{4 f \tau_{ox}} \left(\frac{I}{N}\right)^2 \frac{1}{f}}$$

Experimental measures saw $1/f$ relationship. Now we have a link between $1/f$ and the superposition of Lorentzian shapes:



Let's rewrite the formula to be more meaningful to designers:

$$I = K V_{ov}^2 \quad \text{MOSFET current}$$

$$N = \frac{C_{ox} W L}{q} (V_{ov} - V_T) \quad \text{carriers in the conductive channel}$$

$$N_T = n_T \cdot W L t_{ox} \quad n_T = \text{average trap density per unit volume within the oxide}$$

So:

$$S_I = \frac{n_T W L t_{ox}}{4 t_{ox}} \cdot q^2 \left(\frac{K V_{ov}^2 \cdot I}{C_{ox}^2 (W L) V_{ov}^2} \right) = \frac{n_T}{4} q^2 \cdot \frac{1}{2} \frac{\mu C_{ox} W}{C_{ox}^2 W L^2} =$$

$$= \frac{n_T q^2 \mu}{8 C_{ox}} \cdot \frac{I}{L^2} \cdot \frac{1}{f} = K_I \cdot \frac{I}{L^2} \cdot \frac{1}{f} = S_I / f$$

Refer this to a voltage input noise:

$$S_V g_m^2 = S_I \rightarrow S_V = \frac{S_I}{g_m^2} \quad \text{where } g_m = 2K V_{ov}$$

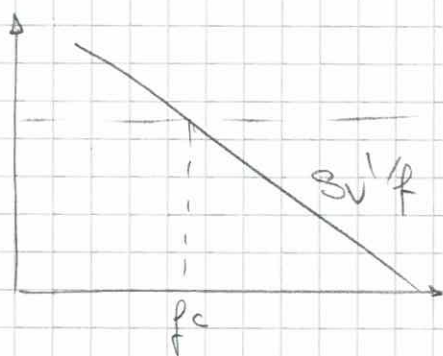
$$S_V = \frac{K_I^{1/2} / f}{(2K V_{ov})^2} \cdot \frac{K V_{ov}^2}{L^2} \cdot \frac{1}{f} = \frac{K_I^{1/2} / f}{4 \cdot \frac{1}{2} \mu C_{ox} \frac{W}{L} \cdot L^2} \cdot \frac{1}{f} =$$

$$= \frac{K_I^{1/2} / f}{2 \mu} \cdot \frac{1}{C_{ox} W L} \cdot \frac{1}{f} = \frac{K_V^{1/2} / f}{C_{ox} W L} \cdot \frac{1}{f}$$

$$S_I^{1/2} = K_I^{1/2} \cdot \frac{I}{L^2} \cdot \frac{1}{f}$$

$$S_V^{1/2} = \frac{K_V^{1/2}}{C_{ox} W L} \cdot \frac{1}{f}$$

The latter is the Tsividis formula



$$\frac{4kT}{g_m}$$

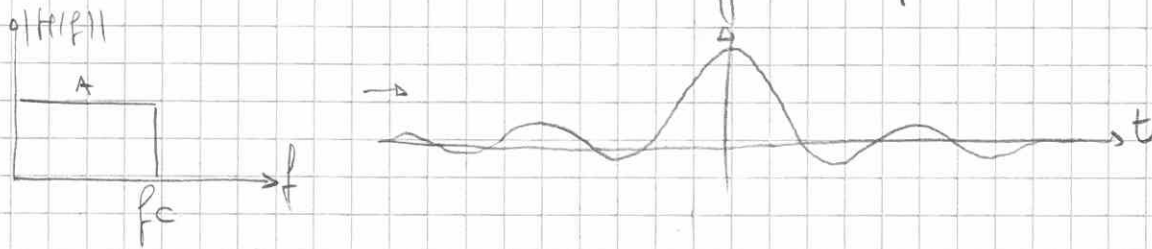
To lower IBIS means worsening thermal noise. We could increase

WL. This makes sense because the larger area makes the statistical

variations and captures/emission count less with respect to the noise. Notice the similarity with Pelgrom's formula

23) Introduction to filters: ideal. Limits. Group delay/distortion

Analog filters select harmonics or change the phase relations between harmonics. Ideal filters would select (brickwall) and attenuate with a rectangle shape:



A rect in frequency domain corresponds to a sinc in time domain \rightarrow this means that in order to be able to have a brickwall filter, we would need to see the future.

Therefore these can't be implemented.

Ideally the requirements for a filter are:

- Flat response in the band-pass $|H(j\omega)|_{BP} = A$
- No phase distortion between harmonics at the output.

These translate to:

$$x_{in}(t) = A \sin(\omega_1 t) + B \sin(\omega_2 t) + C \sin(\omega_3 t)$$

Filter cuts harmonics after ω_2 , $|H(j\omega)|_{BP} = G$ so:

$$x_{out}(t) = AG \sin(\omega_1(t - \tau)) + BG \sin(\omega_2(t - \tau))$$

$$= AG \sin(\omega_1 t - \omega_1 \tau) + BG \sin(\omega_2 t - \omega_2 \tau)$$

We can see that to have constant delay applied to all harmonics we need to apply a proportional phase shift:

$$\varphi_1 = -\omega_1 \tau \quad \varphi_2 = -\omega_2 \tau \rightarrow \underline{\varphi = -\omega \tau}$$

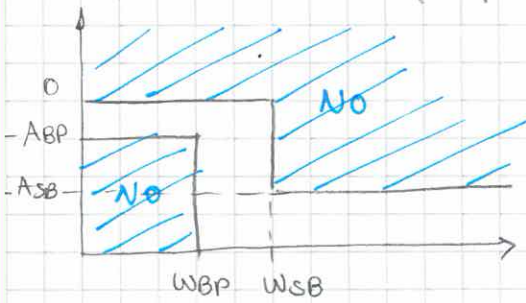
From that, we can define the group delay as

$$\tau = -\frac{\varphi}{\omega} \quad \underline{\tau_{GD} = -\frac{d\varphi(\omega)}{d\omega}} \quad \rightarrow \text{If the phase of the filter is linear, the group delay}$$

is constant ($\varphi = -\omega \tau$). If not, we have a max percentage allowable on group delay

2.4) LPF, Mask, ϵ_{BP} , ϵ_{SB} , K , K_E , n , Butterworth, Cheby

Since we don't have ideal filters, we have to select the limits to properly size the real analog filter \rightarrow masks!



ω_{BP} = band-pass frequency

ω_{SB} = stop-band frequency

A_{BP} = maximum attenuation allowable in band

A_{SB} = minimum attenuation required in stop-band

Filter families

All-poles: Butterworth, Chebyshev type 1, Bessel

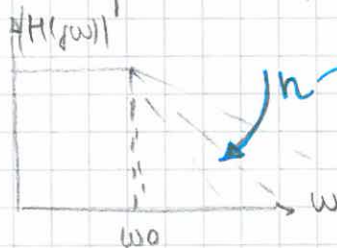
poles + zeros: Chebyshev type 2, Cauer, generalized elliptic

LPF mask - Butterworth

\rightarrow out of band ripple independent in band out of band ripple \rightarrow in/out ripple

For a particular Butterworth function, we know from literature that

$$|H(f, \omega)|^2 = \frac{1}{1 + \left(\frac{\omega}{\omega_0}\right)^{2n}}$$



For $\omega \leq \omega_{BP}$ we have $A_{BP} = 1 \text{ dB}$ so $\frac{1}{|D(\omega)|} \geq \frac{1}{|A_{BP}|}$

Therefore $|D(\omega)| \Big|_{\omega \leq \omega_{BP}} \leq A_{BP} \implies 1 + \left(\frac{\omega_{BP}}{\omega_0}\right)^{2n} \leq (A_{BP})^2$

$$\left(\frac{\omega_{BP}}{\omega_0}\right) \leq \sqrt{A_{BP}^2 - 1} = \epsilon_{BP} \quad \text{defined for a 1st order filter}$$

For a higher order: $\left(\frac{\omega_{BP}}{\omega_0}\right)^n \leq \epsilon_{BP} \implies \frac{\omega_{BP}}{\omega_0} \leq \epsilon_{BP}^{1/n}$

If A_{BP} is expressed in dB $\implies \epsilon_{BP} = \sqrt{10^{\frac{A_{BP}}{10}} - 1}$

We can do the same reasoning for the stopband and obtain (for a 1st order Butterworth)

$$\left(\frac{W_{SB}}{W_0}\right) > \sqrt{A_{SB}^2 - 1} = \epsilon_{SB} \quad \text{for a higher order} \quad \frac{W_{BP}}{W_0} = \epsilon_{SB}^{1/n}$$

We can now define two coefficients:

$$K_\epsilon = \frac{\epsilon_{BP}}{\epsilon_{SB}} \quad \text{called discrimination coefficient}$$

$$K = \frac{W_{BP}}{W_{SB}} \quad \text{called selectivity coefficient}$$

Since $W_{SB} > W_{BP}$ and $\epsilon_{SB} > \epsilon_{BP}$ $\rightarrow K, K_\epsilon$ are always < 1

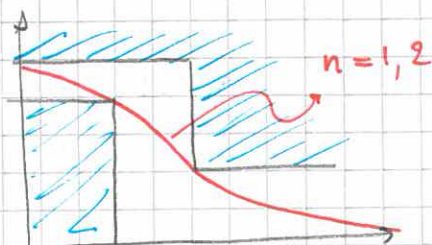
For a n order filter we can write

$$\frac{W_{BP}}{W_{SB}} \leq \left(\frac{\epsilon_{BP}}{\epsilon_{SB}}\right)^{1/n} \rightarrow K \leq K_\epsilon^{1/n} \quad \text{we can drop the}$$

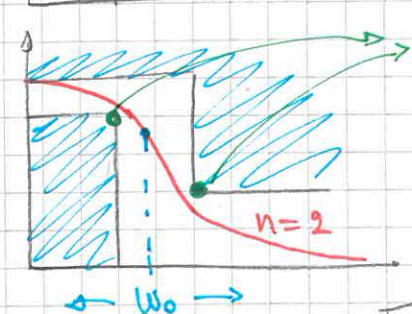
exponent and find the filter order for Butterworth:

$$\ln K \leq \frac{1}{n} \ln K_\epsilon \rightarrow n \geq \frac{\ln K_\epsilon}{\ln K} \quad \text{because } K, K_\epsilon < 1$$

Once we set the filter order, we can move ω_0 so that we match requirements, for example:



$n=1, 2$ \rightarrow The n that touches both A_{BP}, A_{SB} is $n=1, 2$. The real implementable filter needs to be $n \geq 2$



$n=2$ doesn't touch both points. We therefore need to move ω_0 so that we meet some given requirements ($A_{SB}, A_{SB} + \text{margin}$ and so on). To do this, we do:

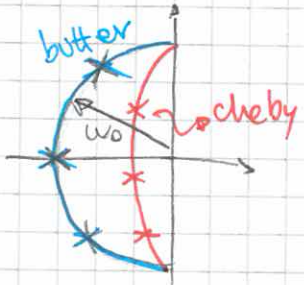
Choose the wanted ω_0 \rightarrow

$$\begin{cases} 1 + \left(\frac{W_{BP}}{W_0}\right)^{1/n} = A_{BP} \pm \text{margin} \\ 1 + \left(\frac{W_{SB}}{W_0}\right)^{1/n} = A_{SB} \pm \text{margin} \end{cases}$$

For a Chebyshev it is $n \geq \frac{\text{Ch}^{-1}(k\epsilon^{-1})}{\text{Ch}^{-1}(k^{-1})}$

Type of filters

Butterworth: requirement of maximum flatness in passband

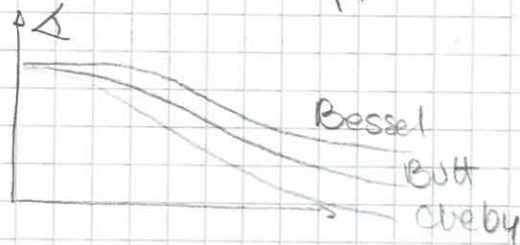
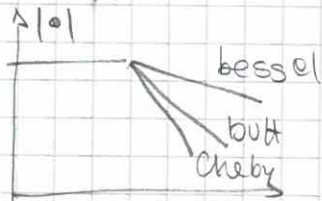


The result is that all poles stay on the circle and are either complex-conj or complex-conj + one real pole (for odd order filters)

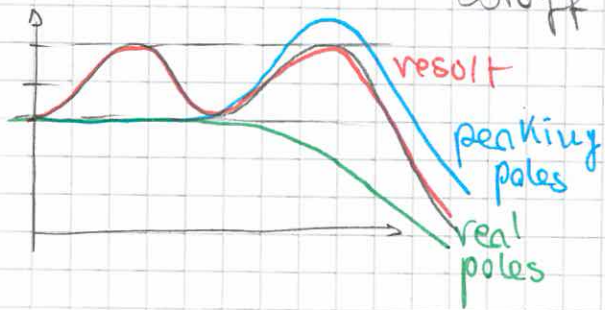
pole pairs will have $s^2 + s \frac{\omega_0}{Q} + \omega_0^2$ $Q = \frac{1}{2\epsilon}$

Bessel: used for a very smooth phase dependence in

Tradeoff with a not sharp cutoff



Chebyshev type 1: used to have even more sharpness in cutoff but allowing in-band ripple



It's the result of a sum of real poles + peaking poles → generation of in-band ripple
Still all-poles filter

Chebyshev type 2: zeros in the function allow for a flat in-band response but generating ripples in the out-of-band

Cauer steeper wt than cheby type 2 but now we have in-band + out-of-band ripple. Also A_{se}, A_{sp} can't be set independently

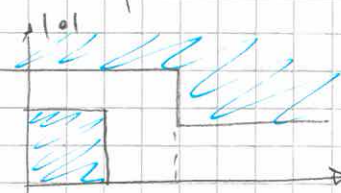
Generalized elliptic: Steeper than Cauer with independent A_{sp}, A_{se}

Issue: worst phase response

25) Mapping: HP to LP, BP to LP transformations

For a LPF, we can map the frequency axis to a normalized frequency axis, where $\omega_{BP} = 1 \text{ rad/s}$

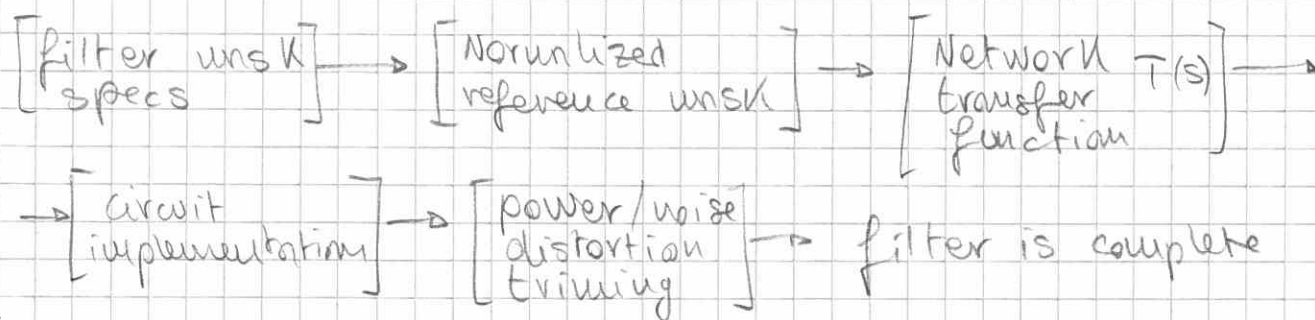
To emphasize the two masks, use $\omega_{BP} = 1 \text{ rad}$; therefore



$$\text{and } \omega_{SB} = \frac{\omega_{SB}}{\omega_{BP}}$$

The normalization procedure is helpful for designing all types of filters

(LP, HP, BP, ...). The process is:



Having the normalized parameters, we can apply a kind of backward transformation to obtain the wanted response (from the normalized filters)

$LP(j\omega) \rightarrow LP(j\Omega)$	$\hat{s} = s / \omega_{BP} \rightarrow \text{intuitive}$ $\hat{s} = \omega_{BP} / s$ $\hat{s} = \frac{Q(s^2 + \omega_0^2)}{s \omega_0} \rightarrow \text{see later}$
$HP(j\omega) \rightarrow LP(j\Omega)$	
$BP(j\omega) \rightarrow LP(j\Omega)$	

Notice that the transformations make sense when the arriving filter is compared with the normalized lowpass

These transformations can now be applied to the normalized LP TF. Example: consider a 1st order Butterworth:

$$T(\hat{s}) \Big|_{LP} = \frac{1}{1 + \hat{s}} = \frac{1}{1 + \frac{s}{\omega_0}}$$

Norm ω_0

To get to a HP we apply $\hat{s} = \omega_{BP}/s$ so

$$T(s) = \frac{1}{1 + \frac{1}{\Omega_0} \cdot \frac{\omega_{BP}}{s}} = \frac{s \Omega_0 / \omega_{BP}}{1 + \frac{s \Omega_0}{\omega_{BP}}}$$

To get to a BP instead:

$$T(s) = \frac{1}{1 + \frac{Q}{\Omega_0} \frac{(s^2 + \omega_0^2)}{s \omega_0}}$$

if we call $Q_p = \frac{Q}{\Omega_0}$ we will end up with the classic BPF response:

$$= \frac{s \omega_0 / Q_p}{\frac{s^2}{\omega_0^2} + s \frac{\omega_0}{Q_p} + 1}$$

Example: 3rd order Butterworth:

$$H(\hat{s}) = \frac{1}{(p+1)(p^2+p+1)} \quad \omega_0 = -3dB \text{ wt at } 1 \text{ rad/s}$$

Expansion of the freq axis $p = \frac{\hat{s}}{\Omega_0}$

$$H(\hat{s}) = \frac{1}{\left(\frac{\hat{s}}{\Omega_0} + 1\right) \left(\left(\frac{\hat{s}}{\Omega_0}\right)^2 + \frac{\hat{s}}{\Omega_0} + 1\right)} = \frac{-\Omega_0^3}{(\hat{s} + \Omega_0) [\hat{s}^2 + \Omega_0 \hat{s} + \Omega_0^2]}$$

Finally, we need another shift $\hat{s} = s / \omega_{BP}$

$$H(s) = \frac{-\Omega_0^3}{\left(\frac{s}{\omega_{BP}} + \Omega_0\right) \left(\frac{s^2}{\omega_{BP}^2} + \Omega_0 \frac{s}{\omega_{BP}} + \Omega_0^2\right)}$$

$$= \frac{(\Omega_0 \omega_{BP})^3}{(s + \Omega_0 \omega_{BP}) \left(s^2 + (\Omega_0 \omega_{BP})s + (\Omega_0 \omega_{BP})^2\right)}$$

BPF transformation derivation LP \rightarrow BP

A normalized BP can be derived from a LP with $\hat{s} = \hat{p} + \frac{1}{\hat{p}}$ where $\hat{s} = \Lambda + j\Omega$ $\hat{p} = \alpha + j\omega$ are two

complex variables, so:

$$\underline{\hat{p}\hat{s} + \hat{p}^2 + 1 = 0} \quad \hat{p} = \frac{\hat{s} \pm \sqrt{\hat{s}^2 - 4}}{2} \rightarrow \text{develop this}$$

$$\alpha + j\omega = \frac{\Lambda + j\Omega}{2} \pm \frac{\sqrt{\Lambda^2 - \Omega^2 + 2j\Lambda\Omega - 4}}{2}$$

Since we're interested in transformations using ω and Ω , focus just on the imaginary axis $\rightarrow \Lambda = 0$:



$\alpha = 0$ and

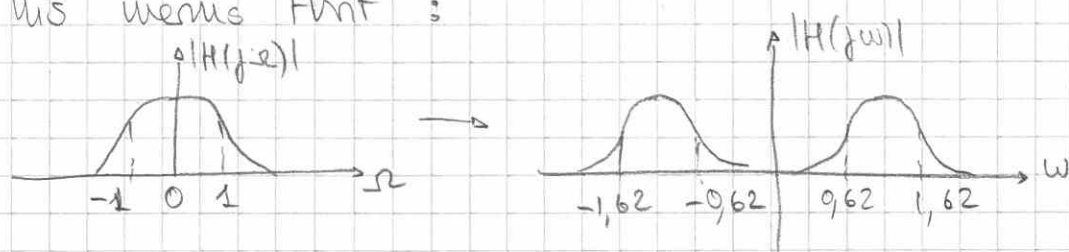
$$j\omega = j\frac{\Omega}{2} \pm j\sqrt{\frac{-\Omega^2 + 4}{2}}$$

$$\underline{\omega = \frac{\Omega}{2} \pm \sqrt{1 - \left(\frac{\Omega}{2}\right)^2}}$$

We therefore conclude that a single Ω point maps on two points in the ω axis. It is:

• $\underline{\Omega = 0} \rightarrow \underline{\omega_{1,2} = \pm 1}$ • $\underline{\Omega = 1} \rightarrow \underline{\omega_{1,2} = \frac{1}{2} \pm \sqrt{1 - \frac{1}{4}}}$ $\begin{matrix} \nearrow +0,62 \\ \searrow -0,62 \end{matrix}$

This means that:

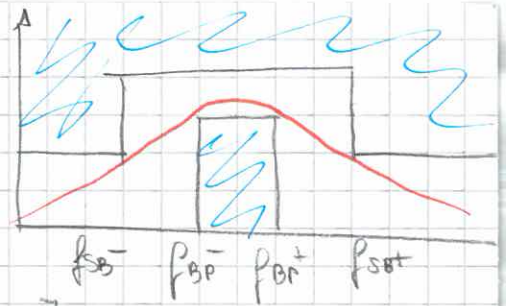


So we can clearly see that for $\Omega = 0 \div 1$ we map a LPF to a BPF. We can set the usual values for the LPF ω_{sk} and then retransform everything

$$f_0 = \sqrt{f_{BP^+} f_{BP^-}}$$

$$Q = \frac{f_0}{BW}$$

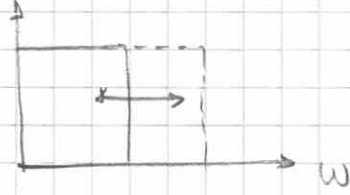
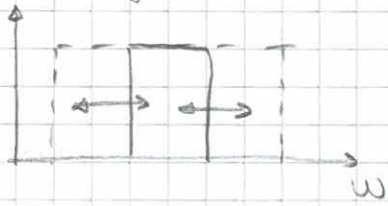
$$BW = f_{BP^+} - f_{BP^-}$$



See that we have a Q factor.

We need to connect that to the

wrapping, so we can use $\hat{s} = Q \left[\hat{p} + \frac{1}{\hat{p}} \right]$



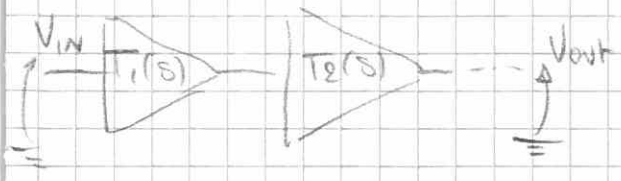
As usual we apply the transformation needed to set ω_0 as the center BPF frequency:

$$s = \omega_0 \hat{p} \rightarrow \hat{p} = \frac{s}{\omega_0} \rightarrow \hat{s} = Q \left[\hat{p} + \frac{1}{\hat{p}} \right] = Q \left(\frac{s^2 + \omega_0^2}{\omega_0 s} \right)$$

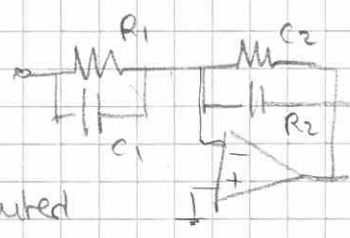
This is indeed the transformation listed in the previous table.

2.6) Active cells: SK cell. Sizing + sensitivity

To build the circuits for the wanted transfer function we need active cells so that we can cascade them easily:



$$V_{out}(s) = T_1 \cdot T_2 \cdot T_3 \dots V_{in}(s)$$

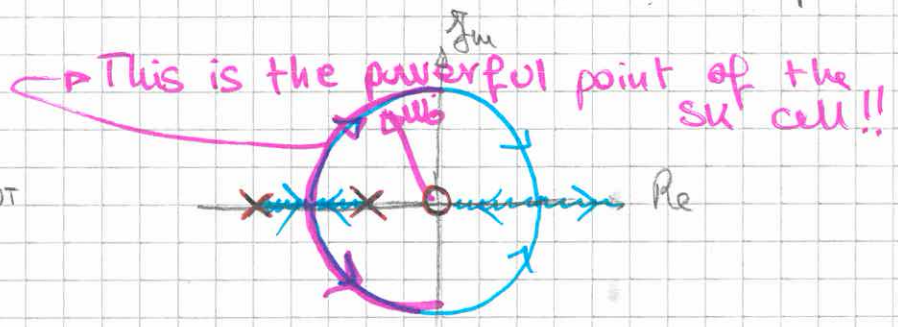
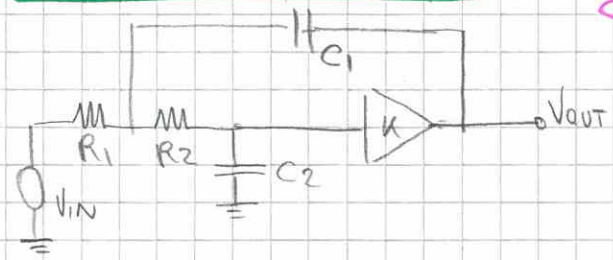


$$T(s) = \frac{1 + s R_2 C_2}{1 + s R_1 C_1}$$

Pole + zero can be implemented

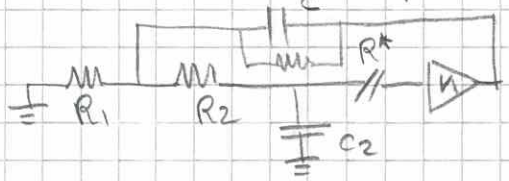
Using a single opamp, but how about a complex pole pair?

Sallen - Key cell



Even though G_{loop} is positive, the root locus shows a stable and unstable region. If $K \gg \rightarrow$ unstable.

We'll solve the poles by deriving $G_{loop}(s) - 1 = 0$:



$$G_{loop}(s) = \frac{a_1 s + 1}{b_2 s^2 + b_1 s + 1} \cdot \frac{R_1}{R_1 + R^*} \cdot K$$

To bypass the zero in the origin, use R^* and then $R^* \rightarrow \infty$

$$b_2 = C C_2 (R_1 // R^*) R_2 \xrightarrow{\infty} C C_2 R_1 R_2$$

$$b_1 = C (R_1 // R^*) + C_2 (R_2 + R_1 // R^*) \xrightarrow{\infty} C R_1 + C_2 (R_1 + R_2)$$

$$a_1 = C R^* \xrightarrow{\infty} C R^* s$$

$$G_{loop}(s) = K \frac{s R_1 C}{s^2 C C_2 R_1 R_2 + s [C R_1 + C_2 (R_1 + R_2)] + 1}$$

Compute now $G_{loop}(s) - 1 = 0 \rightarrow$

$$s^2 R_1 R_2 C C_2 + s [(1-K) C R_1 + C_2 (R_1 + R_2)] + 1 = 0$$

$$\omega_0 = \frac{1}{\sqrt{R_1 R_2 C C_2}}$$

$$Q = \frac{\sqrt{R_1 R_2 C C_2}}{(1-K) C R_1 + C_2 (R_1 + R_2)}$$

To exploit the dependence of Q on R, C , rewrite Q :

$$Q = \sqrt{\frac{R_2}{R_1} \frac{C_2}{C_1}} \quad \text{while} \quad \omega_0 = \frac{1}{\sqrt{R_1 R_2 C_1 C_2}}$$

$$(1-k) + \left(1 + \frac{R_2}{R_1}\right) \frac{C_2}{C_1}$$

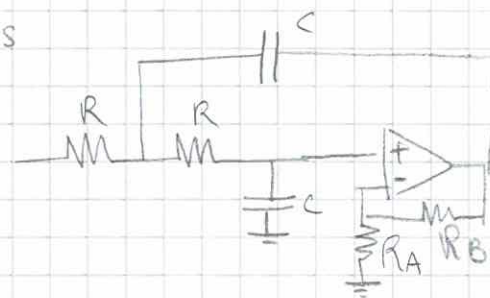
ω_0 depends on absolute R, C values (and mismatches!)

Q depends on relative mismatch between $R_2, R_1 / C_2, C_1$

If we size R, C areas to be big, variability lowers with area ($\frac{1}{\sqrt{WL}}$) so filter ω_0, Q will change less

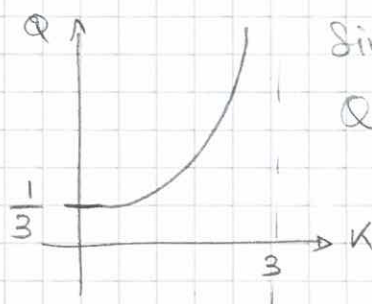
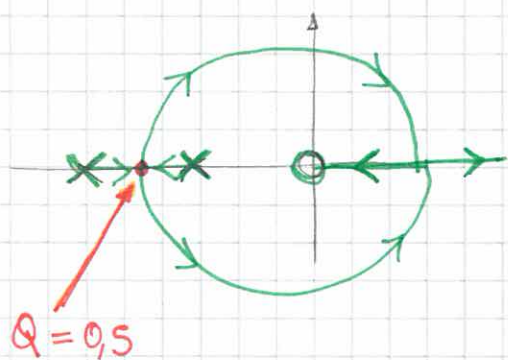
SK with equal values of components

$$K = \frac{R_B}{R_A} + 1 \quad R_1 = R_2 = R \quad C = C_2 = C$$



$$\omega_0 = \frac{1}{RC} \quad Q = \frac{1}{3-K}$$

If $K \nearrow$, poles become complex conjugate as they take off from the root locus. For $Q > 0,5$ we have comp. conjugates



Since $K = 1 \rightarrow +\infty$

Q goes from $\frac{1}{3}$ to $+\infty$

As we can see, to get high Q we need $K \sim 3$, but

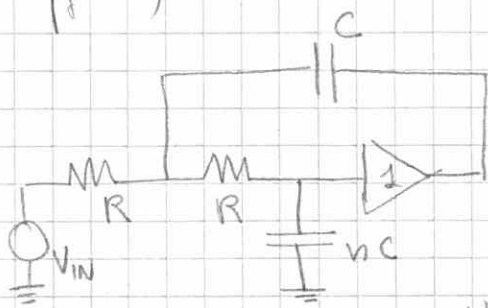
$$\frac{dQ}{dK} = \frac{1}{(3-K)^2} \rightarrow \frac{dQ}{Q} = \frac{1}{(3-K)^2} \frac{K}{Q} \frac{dK}{K} \quad \text{express as a function of } Q$$

$$Q = \frac{1}{3-K} \rightarrow K = 3 - \frac{1}{Q} \quad \text{therefore}$$

$$\frac{1}{\left(3 - 3 - \frac{1}{Q}\right)^2} \cdot \frac{1}{Q} \cdot \left(3 - \frac{1}{Q}\right) = \left(3 - \frac{1}{Q}\right) Q \rightarrow \frac{dQ}{Q} = \left(3 - \frac{1}{Q}\right) Q \frac{dK}{K}$$

The last formula shows that for a mismatch of K , the relative variability of Q increases as Q increases, this makes sense if we look the high slope in the Q/K plot (previous page).

To bypass this, we see what happens if we fix K (buffer):



$$\omega_0 = \frac{1}{RC\sqrt{n}} \quad Q = \frac{\sqrt{n}}{2}$$

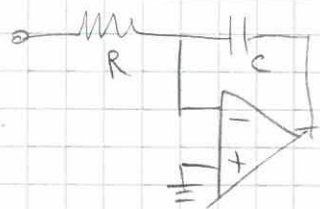
We can derive $\frac{dQ}{Q} = \frac{1}{2} \frac{dn}{n}$

We can see that now relative variability of Q doesn't change.

However, to get high Q $n \gg 1 \rightarrow nC$ occupies a large area

See additional notes to see why SK cell is so powerful!

97) Universal cell: block diagram + Tow Thoms



$$\frac{V_{out}(s)}{V_{in}(s)} = \frac{1}{sRC}$$

From previous courses we found out that this is unstable (by itself).

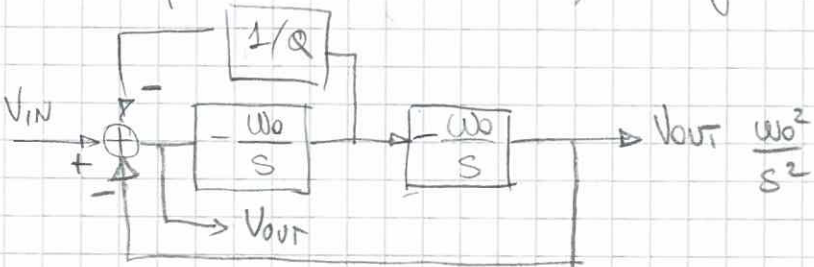
Let's do some magic: HPF design

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

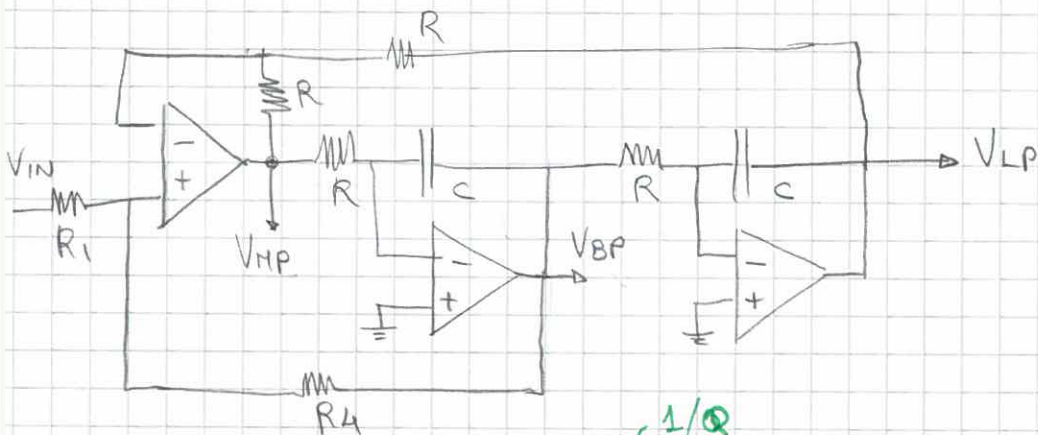
$$\frac{V_{out}}{V_{in}} = \frac{K}{1 + \frac{\omega_0}{sQ} + \frac{\omega_0^2}{s^2}}$$

$$V_{out} \left(1 + \frac{\omega_0}{sQ} + \frac{\omega_0^2}{s^2} \right) = K V_{in} \rightarrow V_{out} = K V_{in} - V_{out} \cdot \frac{1}{Q} \cdot \frac{\omega_0}{s} - V_{out} \frac{\omega_0 \omega_0}{s s}$$

We exploited the T(s) by using single integrators



This can be implemented in a circuit:



$$V_{HP} = V_{in} \frac{2 \cdot R_4}{R_1 + R_4} + V_{BP} \frac{2 R_1}{R_1 + R_4} - V_{LP}$$

$$K = \frac{2 R_4}{R_1 + R_4} \quad Q = \frac{1 + \frac{R_4}{R_1}}{2} \rightarrow K = \frac{R_4}{R_1} \cdot \frac{1}{Q}$$

Using superposition of VLP, VBP, VHP

If Q is set, then K will be linked to Q with R4/R1.

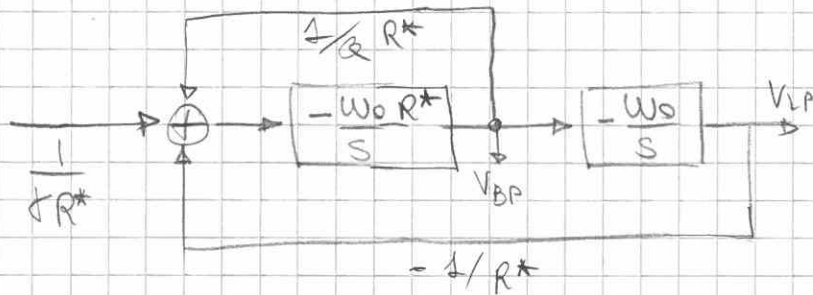
The procedure is helpful to build whatever function:

$$T(s) = \frac{s^2 + \frac{\omega_z}{Qz} s + \omega_z^2}{s^2 + \frac{s\omega_0}{Q} + \omega_0^2} = \text{split} = \frac{s^2}{(\quad)} + \frac{s\omega_z/Q}{(\quad)} + \frac{\omega_z^2}{(\quad)}$$

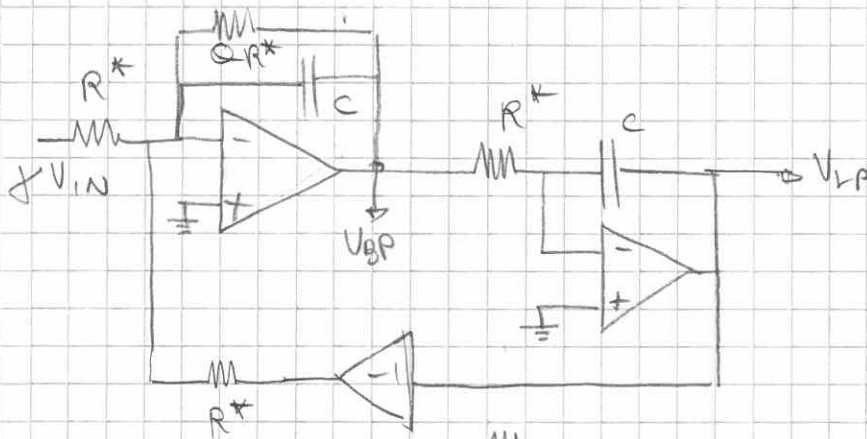
We can always use the general cell and then proceed with a weighted sum to get the wanted output

Tow Thoms

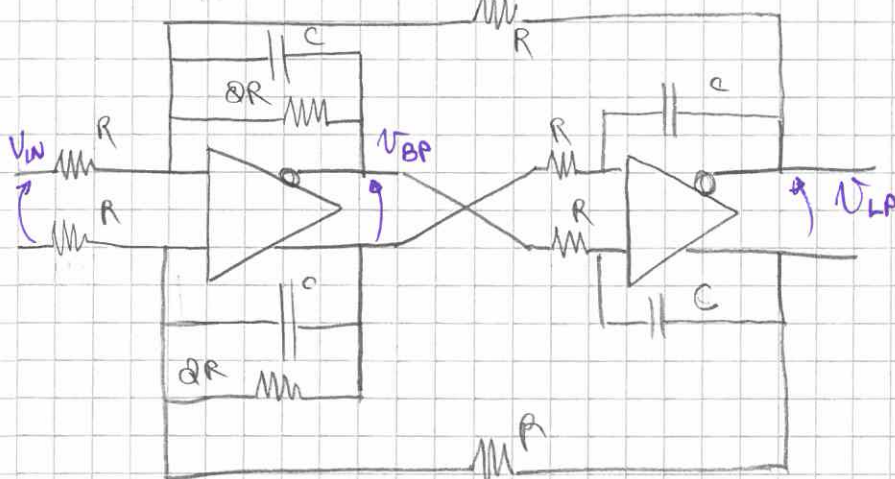
It is possible to simplify the system even more by exploiting the sum of currents at integrator inputs. We switch from voltage to current by using an arbitrary resistor R^*



However, the V_{HP} node will be missing therefore the circuit can implement BPF and LPF only



Notice that we still need a -1 gain. This is not a problem for fully differential opamps



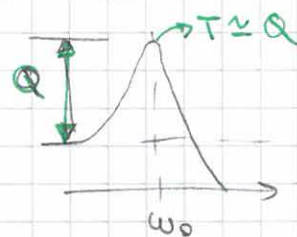
Filter variability

The more filters we cascade to get the proper (high number of poles/zeros) transfer function, the higher we will be subjected to variability of components.

For example consider $Q = \frac{(R_4 + 1)}{R_1} \frac{1}{2} \rightarrow \frac{dQ}{Q} = \frac{1}{2} \frac{d(R_4/R_1)}{R_4/R_1}$

Q is sensitive to the ratio of resistor, which is better than the absolute value of the two

In wo we clearly see that $\frac{\Delta T}{T} = \frac{\Delta Q}{Q}$



The relative T variation is directly connected to the relative Q variation. If we consider the absolute T variation

$$\Delta T = T \frac{\Delta Q}{Q} = Q \frac{\Delta Q}{Q} = Q \frac{d(R_4/R_1)}{(R_4/R_1)}$$

$\rightarrow T|_{\omega=\omega_0} \approx Q$

We see that the absolute variation of T increases for an increasing Q given a fixed $\frac{d(R_4/R_1)}{R_4/R_1}$

$$\text{If } \frac{\Delta R}{R} = \frac{K_R'}{\sqrt{WL}} \frac{1}{\sqrt{2}} \text{ then } \frac{d(R_4/R_1)}{R_4/R_1} = \frac{dR_4}{R_4} + \frac{dR_1}{R_1} = \frac{K_R'}{\sqrt{2W_H L_H}} + \frac{K_R'}{\sqrt{2W_L L_L}}$$

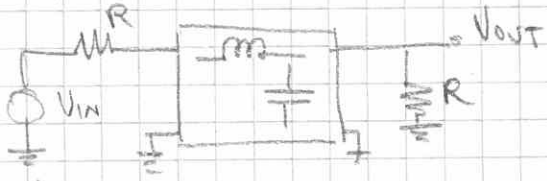
where $K_R' = KR\sqrt{2}$ given by manufacturer

We see that for a reduced variation we need large area of silicon. Small changes on a large area will be more negligible with respect to a small area resistor

28) Ladder Networks : Orchard theorem, implementation.

Flow graph. Decomposition

Consider a lossless network (L, C are ideal) in a doubly terminated network:



At the peaks of the transfer function, the reactive elements resonate in a way that the impedance seen Z_{IN} will be R . The power delivered to the load will then be maximum and its derivative will be nil:

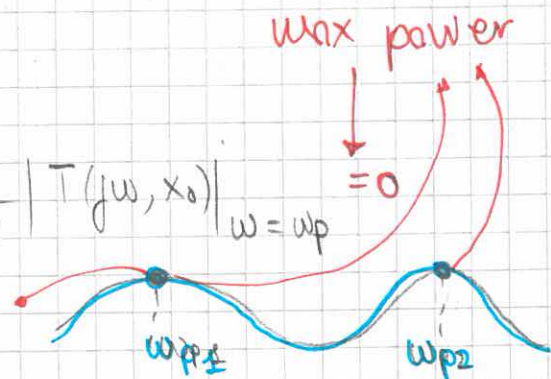
$$P_L = \frac{V_{in}^2}{2R} |T(j\omega, X_0)|^2 \rightarrow \text{reactive elements}$$

This means that no matter what, the reactive elements on the peaks won't give any contribution \rightarrow their variability will be negligible.

Exploit the derivative of the power:

$$\begin{aligned} \left. \frac{\partial P_L}{\partial \omega} \right|_{\omega = \omega_{peak}} &= \frac{V_{in}^2}{2R} \left. \frac{\partial}{\partial \omega} |T(j\omega, X_0)|^2 \right|_{\omega = \omega_p} = \\ &= \frac{V_{in}^2}{2R} \cdot 2 |T(j\omega_p, X_0)| \cdot \left. \frac{\partial}{\partial \omega} |T(j\omega, X_0)| \right|_{\omega = \omega_p} \end{aligned}$$

$$\text{Therefore } \left. \frac{\partial}{\partial \omega} |T(j\omega, X_0)| \right|_{\omega = \omega_p} = 0$$



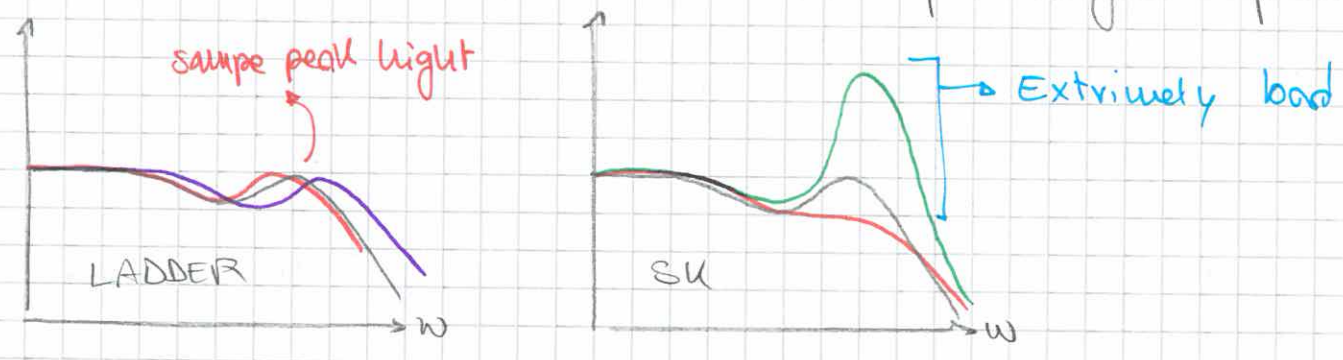
If power is maximum, the transfer function is on a peak

the transfer function depends also on reactive elements X , but when the $P_L = \max$ only R remains, therefore also the derivative on reactive elements is nil:

$$\left. \frac{\partial P_L}{\partial X} \right|_{X=X_0} = \dots \left. \frac{\partial}{\partial X} |T(j\omega, X)| \right|_{\substack{\omega = \omega_{peak} \\ X = X_0}} = 0$$

The last condition is a very powerful result and it's called Orford Theorem.

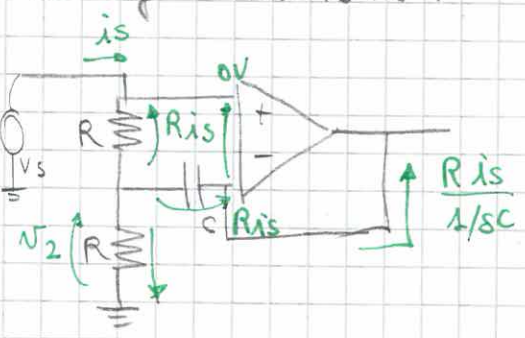
Example: III order cheby implemented with ladder v.s. SK cell. Consider a 10% variation of a single capacitor:



See that peaks move slightly (ω_0 changes), but the peak light won't be affected because $P_L|_{\max} = \frac{V_{in}^2}{2R}$

29) Implementation with gyrators

The issue here is to implement the filter without the use of inductors.



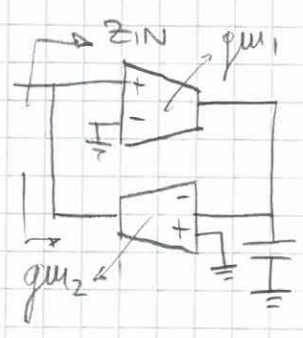
$$v_2 = (i_s + sCRi_s)R = i_sR + sCR^2i_s$$

$$v_s = v_2 + v_1 = 2i_sR + sCR^2i_s$$

$$Z_{IN} = 2R + sCR^2 = R_{eq} + sL_{eq}$$

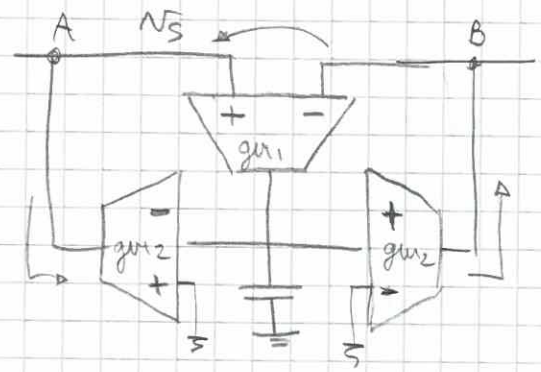
$$R_{eq} = 2R \quad L_{eq} = CR^2$$

Other kinds of gyrators:



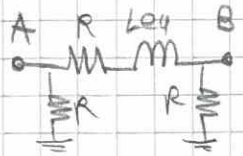
$$Z_{IN} = \frac{sC}{g_{m1}g_{m2}}$$

$$L_{eq} = \frac{C}{g_{m1}g_{m2}}$$



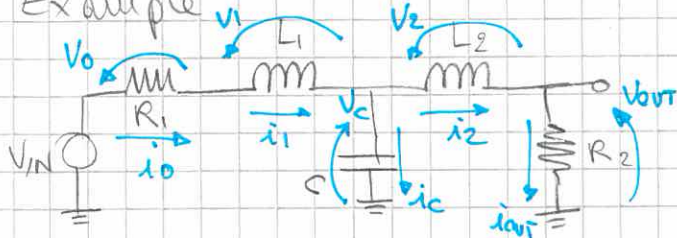
$$Z_{IN} = s \frac{C}{g_{m1}g_{m2}}$$

Issues with gyrators: finite OTA R_{out} , mismatches on geom, noise (we use active elements), distortion, BW is limited because of the GBWP of OTAs, voltage swing is limited, common voltage is limited for floating inductors.



28 part II) Flowgraph procedure

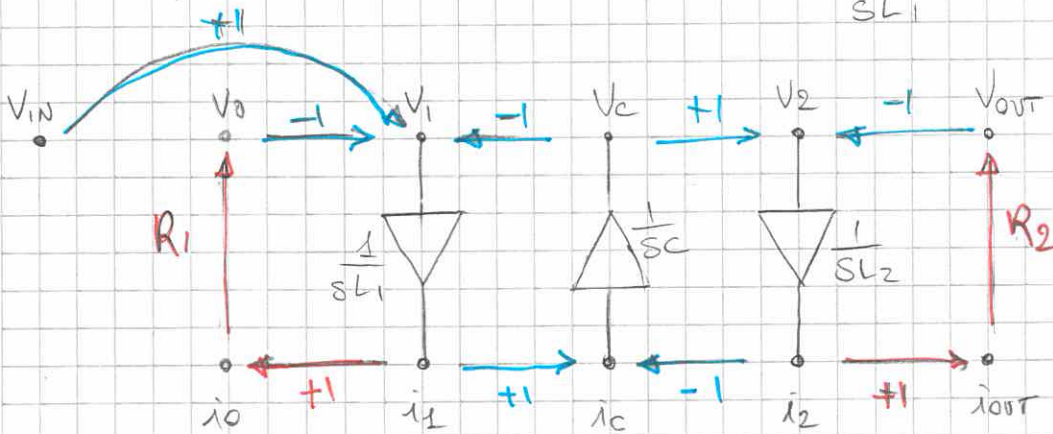
Example



1) highlight all currents and voltages

2) Exploit the integration

link between reactive elements $I_1 = \frac{V_1}{sL_1}$ $I_2 = \frac{V_2}{sL_2}$ $V_c = \frac{I_c}{sC}$

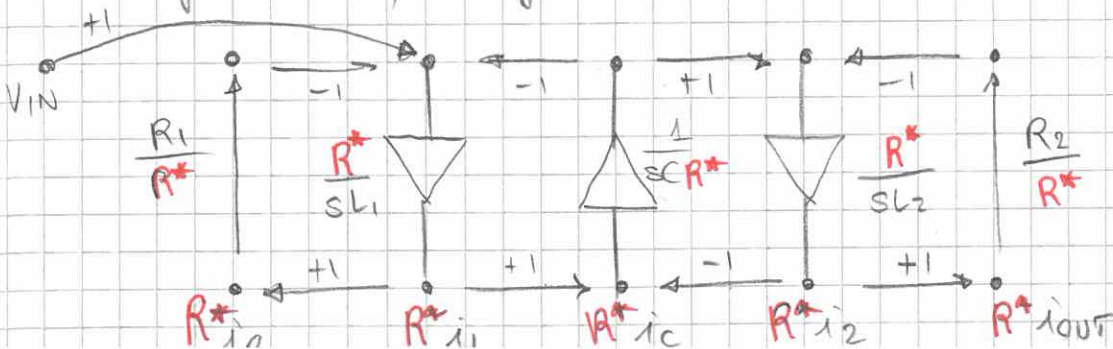


3)

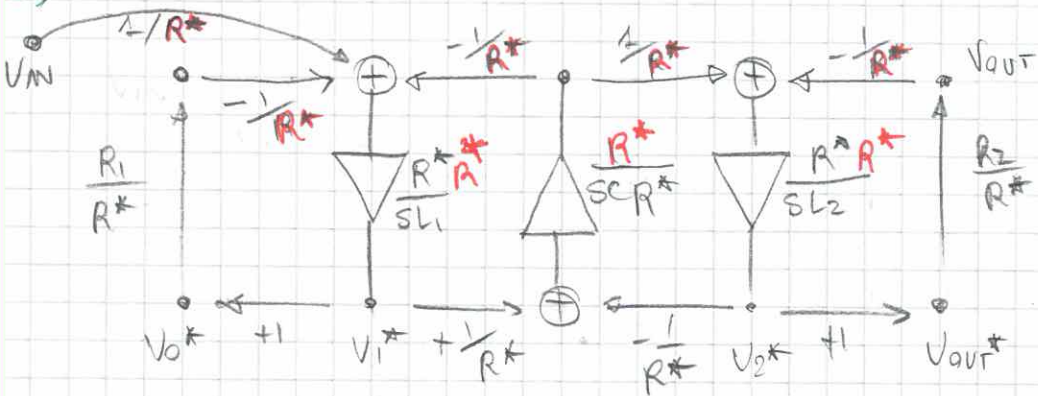
Then, starting from the integrator input, link all the variables by looking at all the voltage/current relations

Then, finish connecting the rest of the nodes, note he $i_0 \cdot R_1 = V_0$ and $i_{out} \cdot R_2 = V_{out}$

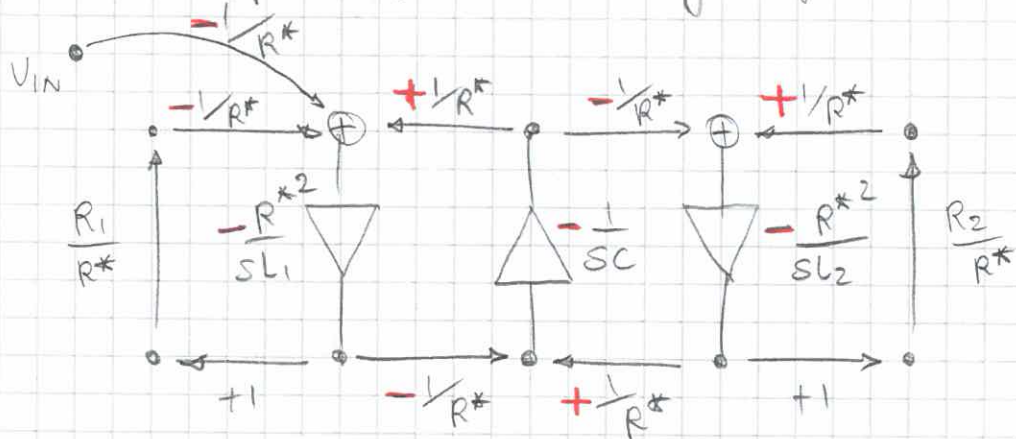
4) It's more convenient to sum currents instead of voltages (less opamps used) change $i_0, i_c, i_1, i_2, i_{out}$ into voltages and adjust everything



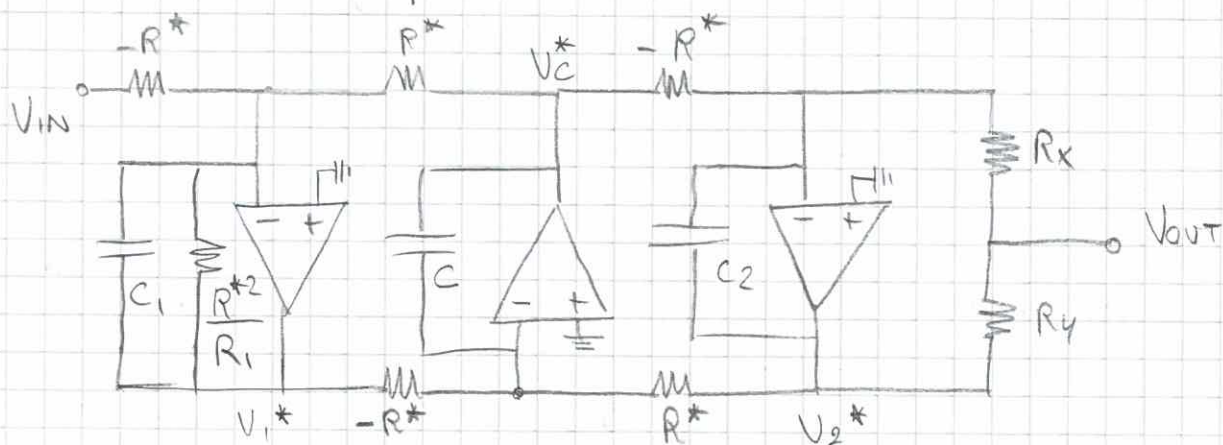
5) Continue with the current sum



6) Final step: adjust the sign for inverting integrators



We can now implement the real circuit



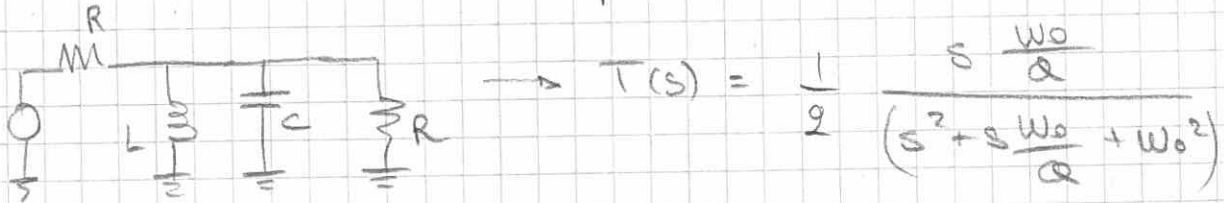
$$R_x + R_y = \frac{R^{*2}}{R_2} \quad \frac{R_y}{R_x + R_y} = \frac{R_2}{R^*} \quad \rightarrow \quad \underline{R_x = \frac{R^{*2}}{R_2} - R^*} \quad \underline{R_y = R^*}$$

$$\underline{C_1 = L_1 / R^{*2}} \quad \underline{C_2 = L_2 / R^{*2}}$$

Denormalization procedure

Ladder networks are usually listed for 1 rad/s.

To shift the filter cutoff we need to scale components so that we can have the proper ω_0 and a realistic capacitor size. For example:



$$\omega_0 = \frac{1}{\sqrt{LC}} \quad Q = \omega_0 \frac{R}{2} \cdot C$$

To shift from 1 rad/s to ω_0 we need

$$\omega_0' = N \frac{1}{\sqrt{C^0 L^0}} \rightarrow \omega_0' = \frac{1}{\sqrt{\frac{C^0}{N} \frac{L^0}{N}}} \quad \underline{C' = \frac{C^0}{N}} \quad \underline{L' = \frac{L^0}{N}}$$

$$Q' = N \omega_0 \cdot \frac{R}{2} \cdot \frac{C}{N} = Q^0 \rightarrow Q \text{ doesn't change}$$

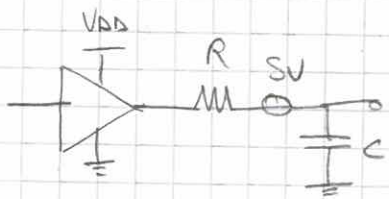
To change resistor size we need to: Keep the same Q without changing ω_0 , so

$$\underline{R' = \pi R^0} \rightarrow Q' = \omega_0 \cdot \frac{R \cdot \pi}{2 R'} \cdot \frac{C^0}{\pi} \quad \text{We need to divide } C \text{ by } \pi \text{ so that } Q' = Q^0$$

unchanged \nearrow

$$\text{If } \underline{C'' = \frac{C^0}{N \cdot \pi}} \text{ then } \underline{L'' = \frac{\pi}{N} L^0} \text{ so } \omega_0 \text{ doesn't change with the new } R$$

30) Dynamic Range: ENOB, guidelines to reduce noise floor



$$\left(\frac{S}{N}\right)^2 = \frac{\frac{A^2}{2}}{\frac{4kT}{RC} \cdot \frac{1}{4RC}} = \frac{A^2}{\frac{2}{C}}$$

This result is obtained using a noiseless opamp, if $A = \frac{V_{DD}}{2} \rightarrow$ half range $(SNR)^2 = \frac{V_{DD}^2/8}{\frac{kT}{C}}$

With noisy opamps $\rightarrow (SNR)^2 = \frac{V_{DD}^2/8}{\frac{kT}{C}(1+F)}$ Where F is an additional noise factor

that degrades the SNR because of active elements. More in general, if signal amplitude is lower:

$$SNR = \sqrt{\frac{\alpha^2 \frac{V_{DD}^2}{8}}{\frac{kT}{C}(1+F)}} = \text{DR dynamic range}$$

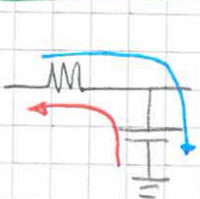
For a given frequency coeff: If $C \uparrow \uparrow$, $R \downarrow \downarrow \rightarrow S \downarrow \downarrow$. Higher capacitor values minimize the noise.

If the system is connected to an ADC:

$$2^n \geq DR \quad n \geq \log_2 DR = \frac{\log_{10} DR}{\log_{10} 2} \quad n \geq \frac{DR/dB}{6dB}$$

$$DR_{dB} = 20 n \log 2 = 6,02 dB \cdot n$$

Noise - power dissipation link



\rightarrow We're charging, discharging the capacitor with a αV_{DD} signal

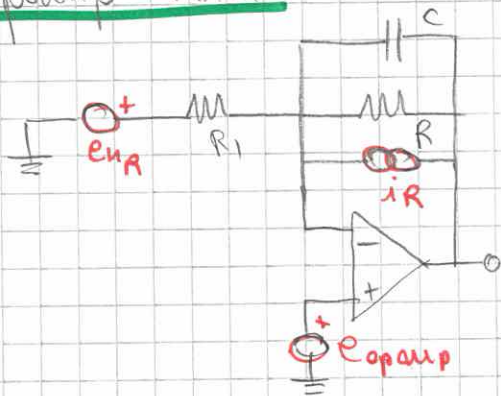
$$Q = C \Delta V = \alpha V_{DD} \cdot C \rightarrow \text{asked charge}$$

We need to calculate the energy dissipated per cycle.

We're taking charge from power supply V_{DD}

$$E_{diss} = \underbrace{\alpha V_{DD} \cdot C}_Q \cdot \underset{\substack{\uparrow \\ \text{supply}}}{V_{DD}} \rightarrow \text{for a complete cycle} \quad P_{diss} = \frac{\alpha V_{DD}^2 C}{T_s} = \alpha V_{DD}^2 C f_s$$

Opamp noise



We want to calculate all the contributions to the output noise

$$G = -\frac{R}{R_1} \quad \omega_0 = \frac{1}{RC}$$

• R1: $S_{out} = 4kTR_1 G^2 \left| \frac{1}{1 + s/\omega_0} \right|^2 = \underline{4kTR_1 G^2 \frac{\omega_0}{4}}$

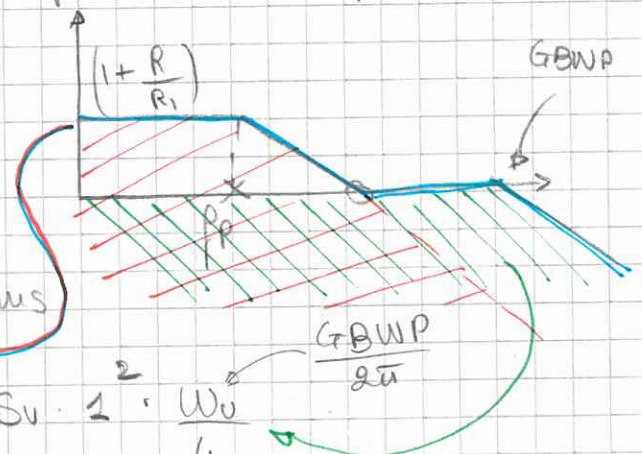
• R: $S_{out} = \underline{4kTR \cdot \frac{\omega_0}{4}}$

• opamp: noise gen at v^+ sees a different transfer function:

$$T(s) = 1 + \frac{R}{R_1} \frac{1 + s(R/R_1)C}{1 + sRC}$$

S_{out} can be seen as a superposition of two contributions

$$\underline{S_{out}}|_{opamp} = \underbrace{S_v \cdot (1 + G)^2 \cdot \frac{\omega_0}{4}}_{\text{RED AREA}} + \underbrace{S_v \cdot 1 \cdot \frac{\omega_0}{4}}_{\text{GREEN AREA}}$$



This is however just an approximation (total noise will be lower → look at tables to find the correct estimate)

• Lastly, input noise will lead to $\underline{S_{out} = S_{in} \cdot G^2 \frac{\omega_0}{4}}$

We finally can compare the various sources to input noise:

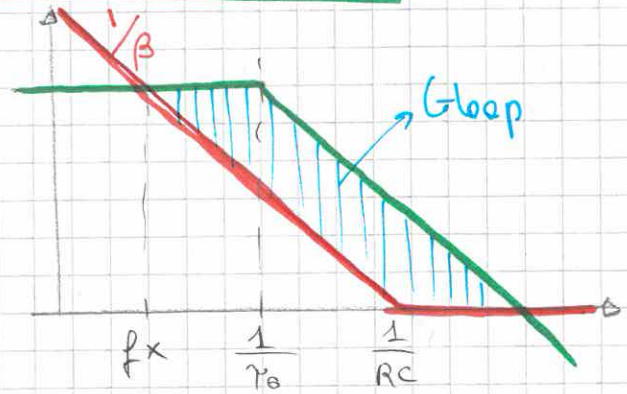
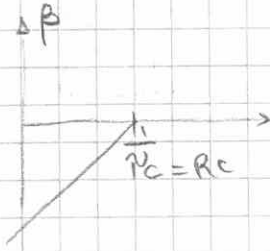
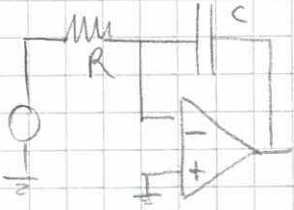
$$\langle V_n^2 \rangle = S_{IN} G^2 \frac{\omega_0}{4} \left[1 + \underbrace{\frac{4kTR_1}{S_{IN}}}_{\textcircled{1}} + \underbrace{\frac{4kTR}{S_{IN} G^2}}_{\textcircled{2}} + \underbrace{\frac{SVA}{S_{IN}} \frac{(1+G)^2}{G^2}}_{\textcircled{3}} + \underbrace{\frac{SVA}{S_{IN} G^2} \frac{\omega_0}{\omega_0}}_{\textcircled{4}} \right]$$

- 1) Use low R_1 values as it directly compares with input noise
- 2) Increase G ? Not bad because a higher gain increases input noise as well as input signal
- 3) G does not have any influence $\left(\frac{1+G^2}{G^2} \sim 1 \right) \rightarrow$ Just select a low noise opamp so that $SVA \ll S_{IN}$
- 4) Gain reduces this contribution. $\omega_0 = \frac{GBWP}{2\pi}$ so ω_0 would lower noise. However $2\pi GBWP = \omega_0$ so it's definitely a bad idea.

Instead, we could add a LPF at the output that cuts at ω_0 with the sum contribution of $\frac{kT}{C}$

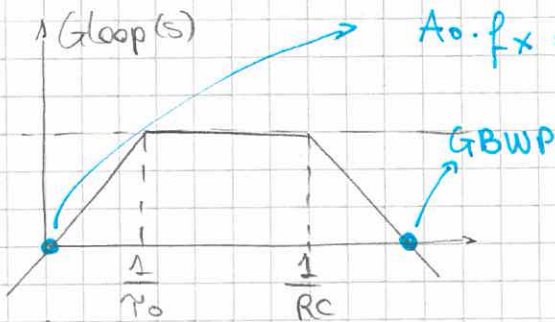
Note: $SVA = \frac{8kTf}{g_m} \left(\frac{3}{2} \right) \rightarrow$ lower SVA means higher g_m therefore higher power dissipation (bias current $\nearrow \nearrow$)

3) Impact of opamp non-idealities on ω_0 and Q for biquad cells

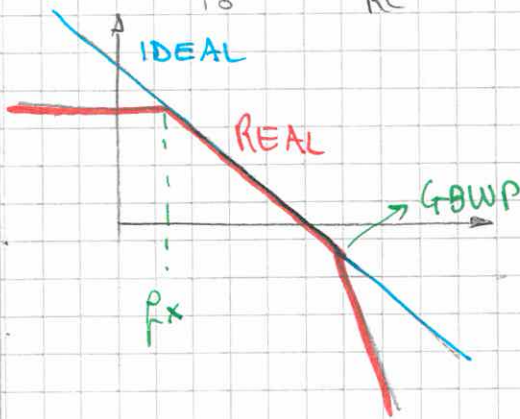


$$\frac{1}{\beta} = 1 + \frac{\omega_0}{s} \quad |A(\omega)| = \frac{A_0}{1 + s\tau_0}$$

$$A_0 \cdot f_x = 1 \cdot \frac{1}{2\pi RC} \rightarrow f_x = \frac{1}{2\pi A_0 RC}$$



G_{loop} is > 1 in the range f_x and $\frac{1}{2\pi RC}$

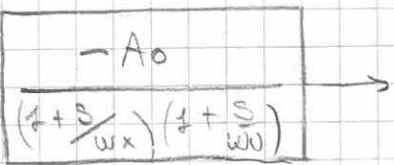


For an ideal integrator: $\rightarrow \left[-\frac{\omega_0}{s} \right]$

With the real one we get:

$$\omega_x = \frac{1}{A_0 RC}$$

$$\omega_0 = 2\pi GBWP$$



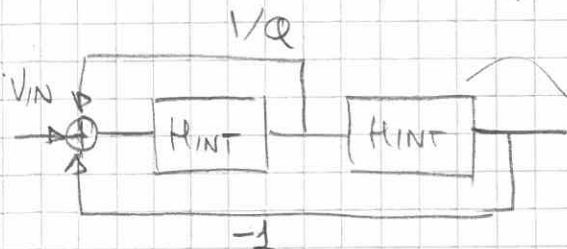
$$A_0 RC = \omega_0$$

Let's use this in the biquad cell.

$$T(s) \Big|_{\text{BIQUAD}} = \frac{\omega_0^2}{s^2 + s \frac{\omega_0}{Q} + \omega_0^2} \rightarrow \text{highlight the integrators}$$

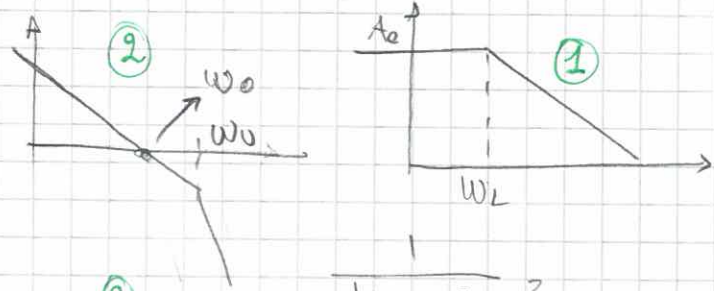
$$= \frac{\left(-\frac{\omega_0}{s}\right)^2}{\left(-\frac{\omega_0}{s}\right)^2 - \frac{1}{Q} \left(-\frac{\omega_0}{s}\right) + 1} = \frac{H_{INT}^2(s)}{H_{INT}^2(s) - \frac{1}{Q} H_{INT}(s) + 1}$$

With the usual topology:



where now $H_{INT}(s)$ takes into account real integrators

The full transfer function is difficult to analyze, so split the analysis with the two poles:



$$H_{INT}(s) \textcircled{2} = \frac{-A_0}{1 + \frac{s}{\omega_L}} = \frac{1}{\frac{1}{A_0} + \frac{s}{\omega_L}}$$

$$H_{INT}(s) \textcircled{1} = \frac{-\omega_0}{s(1 + \frac{s}{\omega_0})}$$

$$T^{-1}(s) \textcircled{2} = \frac{1}{\left(\frac{1}{A_0} + \frac{s}{\omega_0}\right)^2} = \frac{1}{\left(\frac{1}{A_0} + \frac{s}{\omega_0}\right)^2 + 1} = \frac{1}{\left(\frac{1}{A_0} + \frac{s}{\omega_0}\right)^2 + \frac{1}{Q} \left(\frac{1}{A_0} + \frac{s}{\omega_0}\right) + 1}$$

$$= \frac{\omega_0^2}{s^2 + s\omega_0 \left(\frac{1}{Q} + \frac{2}{A_0}\right) + \omega_0^2 \left(1 + \frac{1}{QA_0} + \frac{1}{A_0^2}\right)}$$

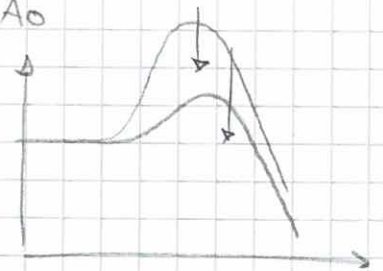
$$\omega_0' = \omega_0 \sqrt{1 + \frac{1}{QA_0} + \frac{1}{A_0^2}} \rightarrow \text{for } A_0 \rightarrow \infty \rightarrow \omega_0' = \omega_0$$

$$\rightarrow \text{for finite } A_0 \rightarrow \text{minor } \omega_0 \text{ shift}$$

$$\frac{\omega_0'}{Q'} = \omega_0 \left(\frac{1}{Q} + \frac{2}{A_0}\right) \text{ approx to } \omega_0' \approx \omega_0, \text{ then } \frac{1}{Q'} \approx \frac{1}{Q} + \frac{2}{A_0}$$

$$\frac{1}{Q'} - \frac{1}{Q} = \frac{2}{A_0} \quad \frac{Q - Q'}{QQ'} = \frac{2}{A_0} \quad \frac{Q' - Q}{Q'} = -\frac{2}{A_0} Q$$

$$\frac{Q' - Q}{Q} \approx \frac{\Delta Q}{Q} \text{ so } \frac{\Delta Q}{Q} \approx -\frac{2}{A_0} Q$$

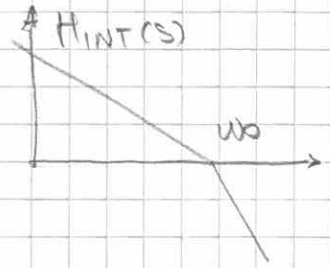


We see Q factor will be quenched because of the opamp finite DC gain

We can allow a % of Q error that is $\epsilon_{ERR} \% = \left| \frac{\Delta Q}{Q} \right|$
 Thus deriving the minimum A_0 requirement

This type of discussion can be made for every active cell filter.

Consider now $H_{INT}' = -\frac{\omega_0}{s} \cdot \frac{1}{1 + \frac{s}{\omega_0}}$



$$T(s) = \frac{\omega_0^2}{s^2 \left(1 + \frac{s}{\omega_0}\right)^2} = \frac{\omega_0^2}{s^2 \left(1 + \frac{s}{\omega_0}\right)^2 + \frac{\omega_0}{Q} s \left(1 + \frac{s}{\omega_0}\right) + \omega_0^2}$$

$$= \frac{\omega_0^2}{s^2 \left(1 + \frac{s}{\omega_0}\right)^2 + \frac{s}{\omega_0 Q} \left(1 + \frac{s}{\omega_0}\right) + 1}$$

We get the poly

$$\left(1 + \frac{s}{\omega_0}\right)^2 \left[\frac{s^2}{\omega_0} + \frac{s}{\omega_0 Q \left(1 + \frac{s}{\omega_0}\right)} + \frac{1}{\left(1 + \frac{s}{\omega_0}\right)^2} \right] = 0$$

It's a 4th-th order polynomial. However if GBWP is large enough $\omega_0 \ll \omega_u$, we can assume that $\frac{s}{\omega_0} \ll 1$ (we're interested in peak reduction that is around ω_0 , so $\frac{s}{\omega_0} \approx \frac{\omega_0}{\omega_0} \ll 1$), therefore we can approx:

$$\frac{1}{\left(1 + \frac{s}{\omega_0}\right)^2} \approx \left(1 - \frac{s}{\omega_0}\right)^2 \quad \text{and} \quad \frac{1}{\left(1 + \frac{s}{\omega_0}\right)} \approx \left(1 - \frac{s}{\omega_0}\right)$$

This way the polynomial is reduced to:

$$\frac{s^2}{\omega_0^2} + \frac{s}{Q \omega_0} \left(1 - \frac{s}{\omega_0}\right) + \left(1 - \frac{s}{\omega_0}\right)^2 = 0$$

$$\frac{s^2}{\omega_0} \left[1 - \frac{\omega_0}{Q \omega_0} + \left(\frac{\omega_0}{\omega_0}\right)^2 \right] + \frac{s}{\omega_0} \left(\frac{1}{Q} - 2 \frac{\omega_0}{\omega_0} \right) + 1 = 0$$

$$s^2 + s \frac{\omega_0}{1 - \frac{\omega_0}{Q \omega_0} + \left(\frac{\omega_0}{\omega_0}\right)^2} \frac{\left(1 - 2 \frac{\omega_0}{\omega_0} Q\right)}{Q} + \frac{\omega_0^2}{1 - \frac{\omega_0}{\omega_0} + \left(\frac{\omega_0}{\omega_0}\right)^2} = 0$$

We can finally derive ω_0' , Q' :

$$\omega_0' = \frac{\omega_0}{\sqrt{1 - \frac{\omega_0}{Q\omega_0} + \left(\frac{\omega_0}{\omega_0}\right)^2}} \rightarrow \text{slightly larger than } \omega_0$$

$\omega_0 \ll \omega_0$

We then have $\omega_0 \left[\frac{1}{Q} - \frac{2\omega_0}{\omega_0} \right] \rightarrow \frac{1}{Q'} = \frac{1}{Q} - \frac{2\omega_0}{\omega_0}$

$$\frac{1}{Q'} - \frac{1}{Q} = -\frac{2\omega_0}{\omega_0} \quad \frac{Q' - Q}{Q Q'} \approx \frac{\Delta Q}{Q} = \frac{2\omega_0}{\omega_0}$$

Recap

Finite DC gain \rightarrow minor ω_0 shift (slightly larger)
 $\frac{\Delta Q}{Q} = -\frac{2}{A_0} Q$

Finite BW \rightarrow minor ω_0 shift (slightly larger)
 $\frac{\Delta Q}{Q} = \frac{2\omega_0}{\omega_0}$

We see that for the finite BW limit, Q is enhanced.

We therefore have two effects, droop because of finite DC and enhancement because of the finite BW.

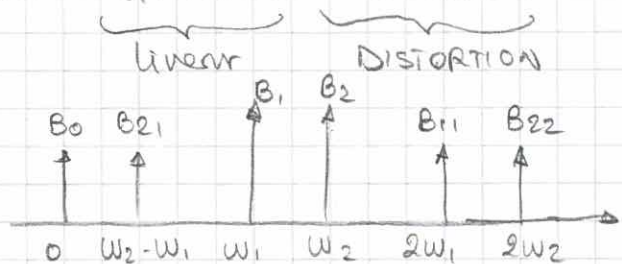
We need to take into account all the contributions:

$$\frac{\Delta Q}{Q} \approx Q \left[\frac{2\omega_0}{\omega_0} + \frac{2\omega_0}{\omega_{p2}} + \dots - \frac{2}{A_0} \right]$$

Distortion

$$x_{IN}(t) = A_1 \cos(\omega_1 t) + A_2 \cos(\omega_2 t)$$

$$y(t) = \alpha_1 x_{IN}(t) + \alpha_2 x_{IN}^2(t)$$

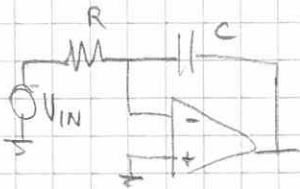


Even though we're filtering our input harmonics, filter distortion can generate

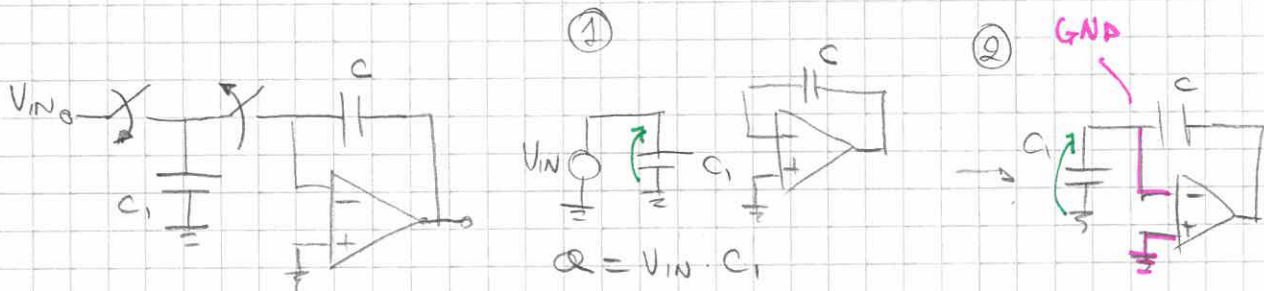
other harmonics, this process is called spectral regrowth

32) SW cap filters: Motivations, concepts, implementation + stray insensitive topologies

Audio range is $BW \sim 10\text{kHz}$ \rightarrow with 1pF capacitors we would need $\sim 10\text{M}\Omega$ \rightarrow even with $R_{sq} = 2\mu\Omega/\square$ the area needed would be too large for implementation. Solution \rightarrow SW capacitors



$$V_{out} = -\frac{V_{in} t}{RC} \quad \text{with DC input}$$



- 1) V_{in} charges $C_1 \rightarrow Q_{C_1} = V_{in} \cdot C_1 \rightarrow$ sampling of V_{in}
- 2) C_1 is between gnd and virtual $gnd \rightarrow$ charge is transferred on $C \rightarrow \Delta Q_c = C_1 V_{in} = C \cdot V_{out} \rightarrow \Delta V_c = \frac{C_1}{C} V_{in}$

Assuming C, C_1 ideal, we'll see a staircase at the output: because we're continuously applying new charge to C



\rightarrow Slope $\frac{\Delta V_c}{T}$ will be $\frac{C_1 E}{CT} = \frac{\Delta V_c}{T}$ if

Linear analog integrator ramp is $\frac{\Delta V_{out}}{\Delta t} = -\frac{V_{in}}{RC}$ while it is $\frac{\Delta V_{out}}{\Delta t} = -\frac{V_{in} C_1}{TC}$ \rightarrow We see a similarity by posing

$$R_{eq} = \frac{T}{C_1} = \frac{1}{f_s \cdot C_1} \rightarrow T = 10\mu s, C_1 = 1\text{pF} \rightarrow R_{eq} = 10\text{M}\Omega$$

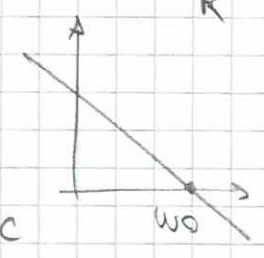
It's convenient to use switched caps

The result can be found in an alternative way by saying:

$$Q_s = C_1 V_{in} \quad i_{average} = \frac{C_1 V_{in}}{T} \quad \text{in a period } T$$

$i_{in} = \frac{C_1 V_{in}}{T}$ the same happens for a resistor $i_{in} = \frac{V_{in}}{R}$

Thus deriving $R_{eq} = \frac{T}{C_1}$



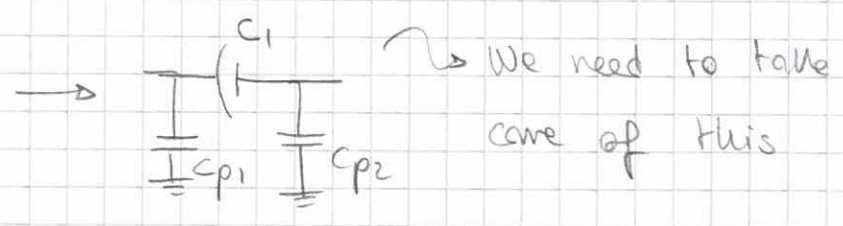
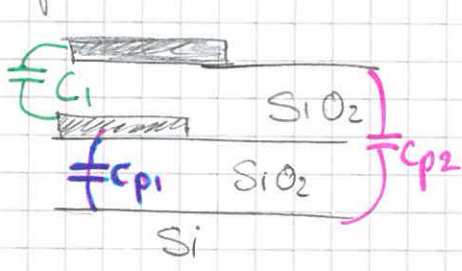
The resulting unity gain frequency is $\omega_0 = \frac{1}{R_{eq} \cdot C}$

Of course (aliasing) $f_{clock} \gg BW$ of signal

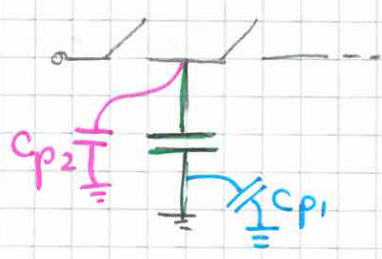
At the same time $f_{clock} \ll GBWP$ of the amplifier in order to provide a good virtual ground and limiting nonlinearity

Stray insensitive topologies

Integrated capacitors suffer from (ironically) parasitic capacitances:



We need to take care of this



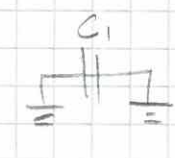
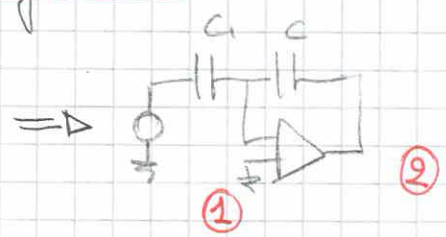
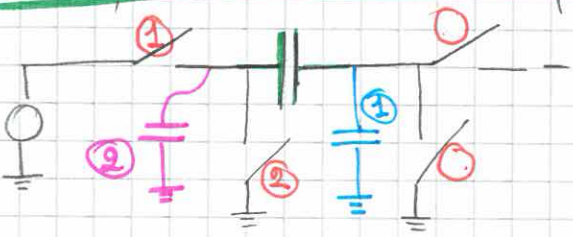
Since $C_{p1} > C_{p2}$ (look at the picture), it's better to short it to gnd.

Now only C_{p2} generates issues

$R_{eq} = \frac{T}{C_1 + C_{p2}}$

We can't control C_{p2} value
Can we eliminate it?

Stray insensitive configuration

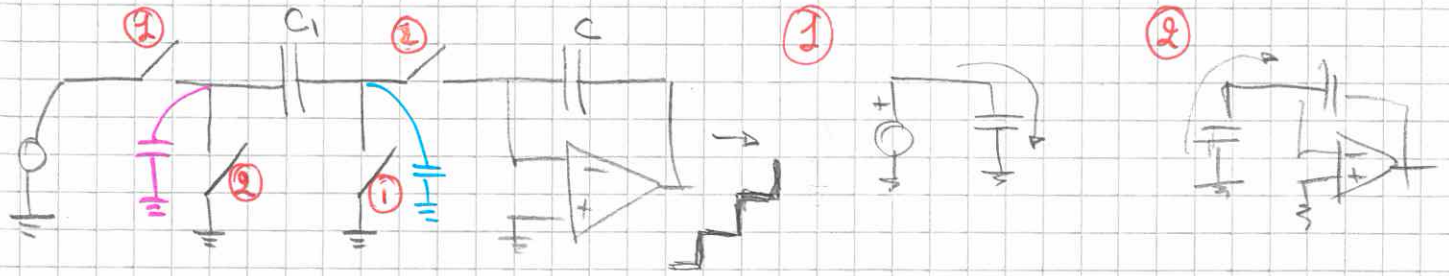


1 cap is tied to V_{gnd} during ① and gnd during ②

2 cap is charged by V_{in} on ①, shorted by ②

Therefore V_{out} will not suffer from C_p (ideally)

Moreover, if we now switch phases:

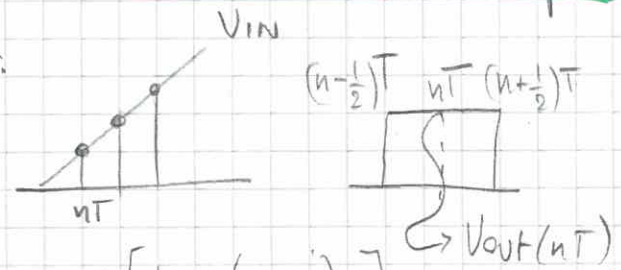


It's easy to see that C_p can't contribute either.

With those switched phases, we can also implement a non inverting integrator with ease, so Tow Thoms $-R^*$ can be implemented without additional opamps

33) SC: sampling, t_f , out spectrum. Anti-aliasing filter and f_{cu}

Out spectrum: series of rectangles:



$$V_{out}(t) = \sum_0^{+\infty} V_{out}(nT) \left\{ 1 \left[t - \left(n - \frac{1}{2} \right) T \right] - 1 \left[t - \left(n + \frac{1}{2} \right) T \right] \right\}$$

↙ Laplace

$$V_{out}(s) = \sum_0^{+\infty} V_{out}(nT) \frac{1}{s} \left\{ e^{-s \left(n - \frac{1}{2} \right) T} - e^{-s \left(n + \frac{1}{2} \right) T} \right\}$$

↙ Fourier

$$= \sum_0^{+\infty} V_{out}(nT) \frac{1}{s} e^{-snT} \left[e^{s \frac{T}{2}} - e^{-s \frac{T}{2}} \right]$$

$$V_{out}(j\omega) = \sum_0^{+\infty} V_{out}(nT) \frac{1}{2j\omega \frac{T}{2}} \cdot e^{-j\omega nT} \cdot \frac{2}{2} T \left[e^{-j\omega \frac{T}{2}} - e^{+j\omega \frac{T}{2}} \right]$$

↘ $\text{sinc} \left(\frac{\omega T}{2} \right)$

$$V_{out}(j\omega) = \sum_0^{+\infty} V_{out}(nT) e^{-j\omega nT} \cdot T \text{sinc} \left(\frac{\omega T}{2} \right)$$

↳ This is the nT delay of $V_{out}(nT)$

$$V_{out}(j\omega) = \sum_0^{+\infty} V_{out}(nT) e^{-j\omega nT} \Rightarrow V_{out}(z) = \sum_{n=0}^{+\infty} V_{out}(n) z^{-n}$$

↑
definition of Z transform

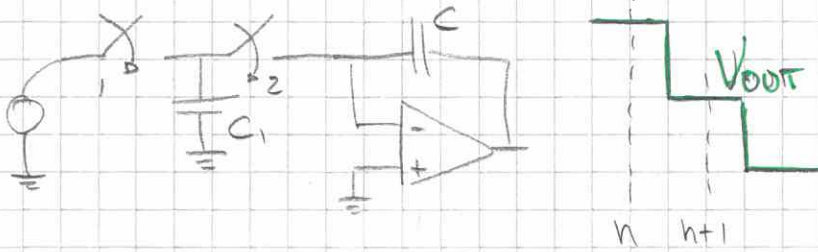
Therefore

$V_{out}(j\omega) = V_{out}(z) \Big|_{z=e^{j\omega T}} \cdot T \text{sinc} \left(\frac{\omega T}{2} \right)$

output sequence
Z-transform

Filter transfer function $H(z)$

We now need to find the relation between V_{in} and V_{out} :



@ time = n



@ time = n+1



$$Q_c = C_1 \cdot V_{in}$$

$$V_{out}(n+1) = -V_c(n+1)$$

Therefore we can write

$$V_{out}(n+1) = V_{out}(n) - V_{in}(n) \frac{C_1}{C}$$

$$V_{out}(z) z = V_{out}(z) - V_{in}(z) \frac{C_1}{C} \rightarrow \frac{V_{out}(z)}{V_{in}(z)} = -\frac{C_1}{C} \frac{1}{z-1}$$

$$\underline{H(z) = -\frac{C_1}{C} \frac{1}{z-1}}$$

↳ sampled transfer function of the circuit

Total output spectrum

$$V_{out}(j\omega) = V_{in}(z) \cdot H(z) \Big|_{z=e^{j\omega T}} \cdot T \operatorname{sinc}\left(\frac{\omega T}{2}\right)$$

Where:

$$V_{in}(z) \Big|_{z=e^{j\omega T}} = \sum_{n=0}^{+\infty} V_{in}(n) e^{-j\omega n T} \quad \text{which is the Fourier}$$

transform of $V_{in}(t) \sum_{n=-\infty}^{+\infty} \delta(t-nT) \rightsquigarrow$ sampled V_{in}

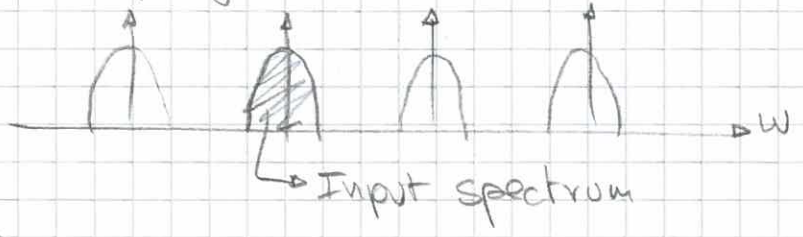
$$V_{in}(z) \Big|_{z=e^{j\omega T}} = \mathcal{F} \left[V_{in}(t) \sum \delta(t-nT) \right] = V_{in}(j\omega) * \mathcal{F} \left(\sum \delta(t-nT) \right) =$$

$$= V_{in}(j\omega) * \sum_{k=-\infty}^{+\infty} \frac{2\pi}{T} \delta\left(\omega - \frac{2\pi}{T} k\right)$$

Therefore:

$$V_{out}(j\omega) = \underbrace{V_{in}(j\omega) * \frac{2\pi}{T} \sum \delta\left(\omega - \frac{2\pi}{T} k\right)}_{*1} \cdot \underbrace{H(z) \Big|_{z=e^{j\omega T}} \cdot T \operatorname{sinc}\left(\frac{\omega T}{2}\right)}_{*2}$$

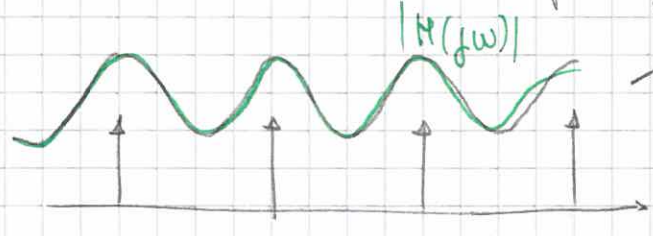
*1) sampling in time domain \rightarrow replicas in frequency domain



$$*2) \quad H(z) \Big|_{z=e^{j\omega T}} = -\frac{c_1}{c} \frac{1}{e^{j\omega T} - 1} = -\frac{c_1}{c} \frac{1}{\cos(\omega T) + j \sin(\omega T) - 1}$$

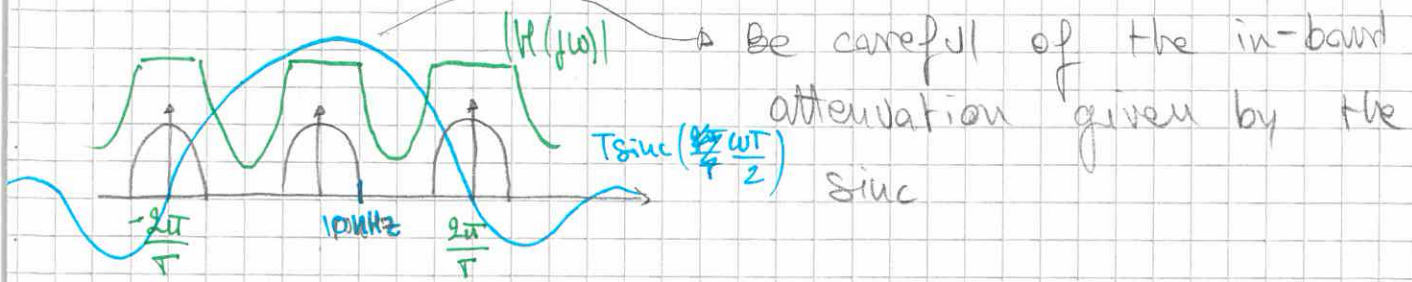
$$\underbrace{|H(z)|}_{z=e^{j\omega T}} = \frac{c_1}{c} \frac{1}{\sqrt{[-1 + \cos(\omega T)]^2 + \sin^2(\omega T)}}$$

\rightarrow This has a transfer function that is something like:



\rightsquigarrow The simple integrator will have $H(j\omega) \rightarrow \infty$ on each δ as it would do an average one

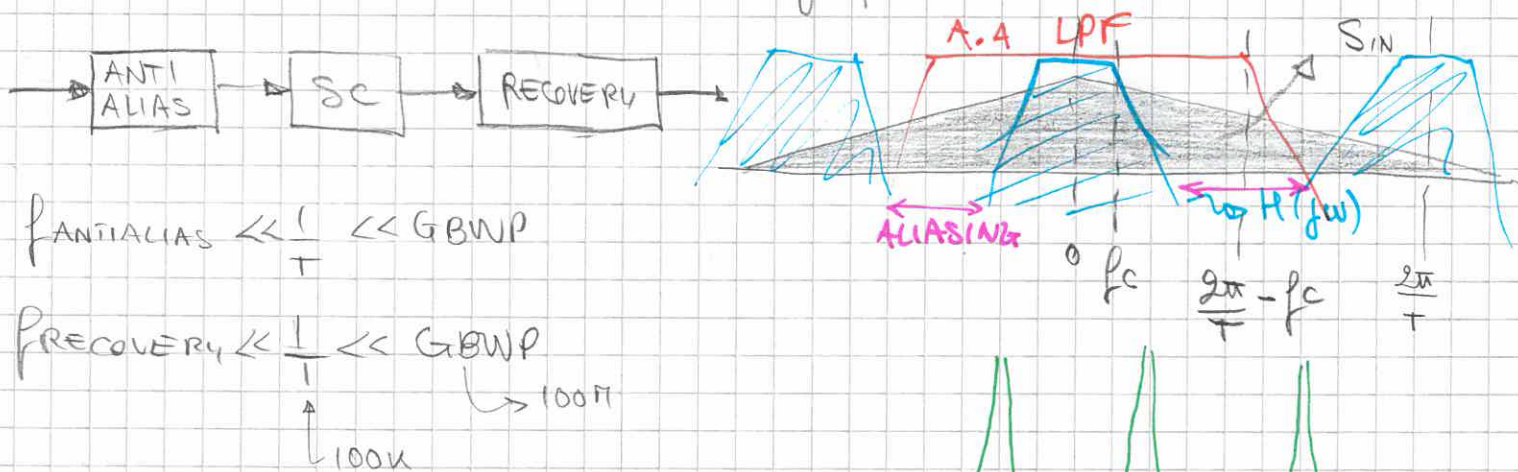
Every filter will have a different $H(z)$, but the replicast sinc function one typical of SC themselves, so



$H(j\omega)$ shows $2\pi f$ periodicity because after the sampling, the circuit is no more able to recognize f_0 harmonic from the others at $f_0 + \frac{2n\pi}{T}$.

Sinc is related to the stepped output. However, if the SC out is read by an ADC, the sinc is lost since we need not to acquire the signal in continuous time domain.

Also, since output BW is limited to the filter, to get rid of replicas we can use an analog filter:



$$f_{\text{ANTI ALIAS}} \ll \frac{1}{T} \ll \text{GBWP}$$

$$f_{\text{RECOVERY}} \ll \frac{1}{T} \ll \text{GBWP}$$

↑ 100k
↓ 100π

Note: for a simple SC integrator for 0Hz it goes to ∞ like an analog integrator does

AA filter: $f_{AA} = f_{cu} - f_0$ (f_0 = cutoff of the integrator)

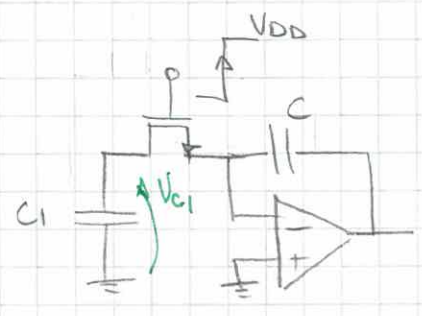
This means that for $f < f_0 \rightarrow$ no aliasing

for $f_0 < f < f_{cu} - f_0 \rightarrow$ ALIASING but it's filtered by H (and by sinc for a little part). This means that

f_{AA} can be relaxed with respect to f_0

34) Non idealities: trade-off between settling time and charge sharing in sizing the switches

Switches are not ideal. Transient time needs to be low enough ($R_{on} C_i < \frac{T}{2}$) so that the step isn't damaged.



Based on V_{G1} , MOS could be SAT or OHMIC. Suppose that MOS is ohmic, then:

$$R_{on} = \frac{1}{g_{D0}} = \frac{1}{2\mu V_{ov}}$$



$$V_{DS} = V_{G1} - V_{Gnd} = V_{G1}$$

$$I_{DS}|_{OHMIC} = 2\mu \left[(V_{DD} - V_T) V_{G1} - \frac{V_{G1}^2}{2} \right]$$

The cap discharge current is the following:

$$I_{DS} = -C_i \frac{dV_{G1}}{dt} \rightarrow dt = \frac{-C_i}{I_{DS}} dV_{G1} \rightarrow \text{solve this:}$$

$$\int_0^{T_{sw}} dt = -\frac{C_i}{\mu} \int_{V_F}^{V_F} \frac{dV_{G1}}{2V_{ov} V_{G1} - V_{G1}^2} \rightarrow T_{sw} = \frac{C_i}{2\mu V_{ov}} \ln \left\{ \frac{(2V_{ov} - V_F)V_F}{(2V_{ov} - V_T)V_F} \right\}$$

$V_F \rightarrow V_{G1} > V_F$

See notes at the end of sheets

To first order, we can say $T_{sw} = \frac{C_i}{2\mu V_{ov}}$

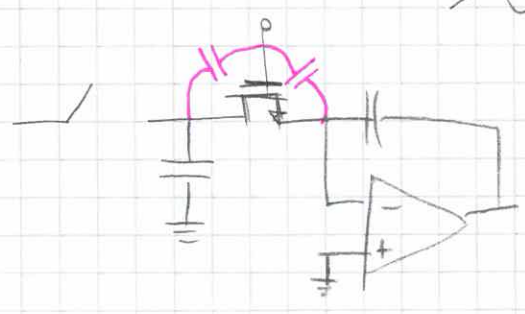
Usually C_i is set by noise requirements, therefore

$$T_{sw} = \frac{C_i}{2\mu n C_{ox} (V_{DD} - V_T)} \frac{L}{W} \rightarrow L = L_{min} \text{ to reduce } T_{sw}$$

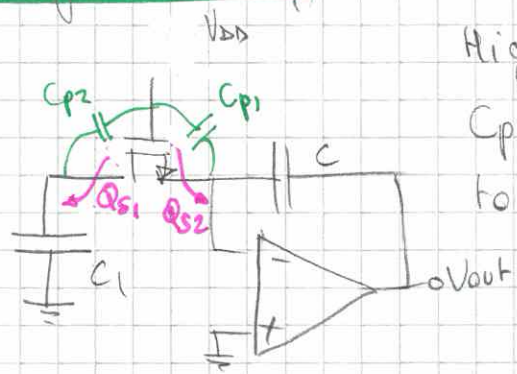
The last choice would be $W \uparrow \uparrow$, but large area means

larger $C_{ox} = WL C'_{ox}$

\rightarrow We now need to take into account the C_{par}

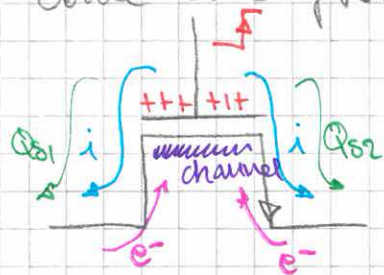


Charge sharing



High W means that τ_{OS} has C_{p1} , C_{p2} of a moderate value that needs to be controlled.

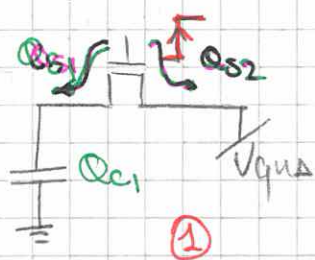
To form the channel, some electrons come from C_1 and some come from v_{gs} .



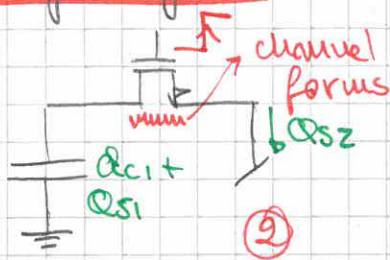
When we switch the transistor on, some charge is needed to form the channel.

This means that e^- flow inside the τ_{OS} from C_1 and v_{gs} , therefore the current flows through the other direction. This means that there's charge injected to C_1 and v_{gs} .

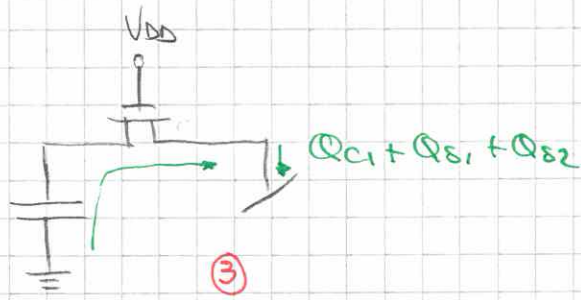
This is called "charge injection".



①



②



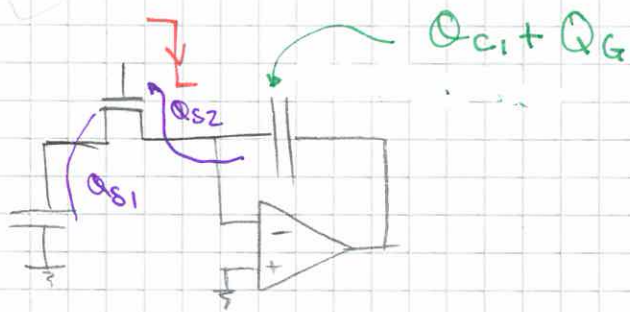
③

① ON transient, charge is moving to form the channel

② Q_{s2} is injected, Q_{s1} accumulates on Q_{s1}

③ τ_{OS} is on. Channel is formed and now $Q_{c1} + Q_{s1}$ can flow to v_{ground} too.

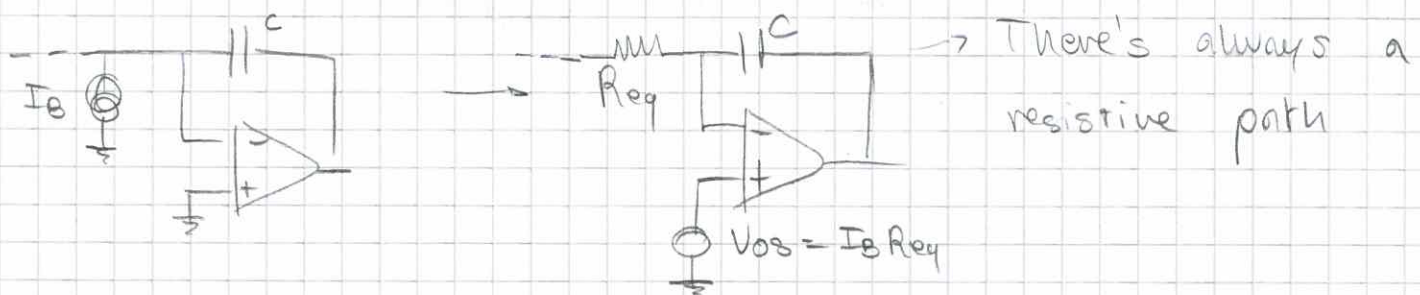
Total charge that will go on C will be $Q_{c1} + Q_{s2} + Q_{s1}$



On the OFF transient, e^- flow towards C_1 and v_{out} . Some portion (because of different impedances and non-idealities of the opamp) will be lost on C_1 , therefore after the two cycles there will be some net positive charge accumulated on $C \rightarrow Q_C = \alpha Q_{C1} + \alpha Q_G$ where $Q_G = C_{ox} \cdot (V_{DD} - V_t)$ and α is a proportional factor

Some upgrade can be done by using NMOS + PMOS so that the differential switching on V_{in} and V_{op} will reduce the additional charge effect

We can see this accumulated charge for each clock cycle as an injected current $I_B = \alpha \frac{Q_{ch}}{T_{clk}}$, modeled like:



I_B would generate a ramp on C , but since SC filters always have resistive paths on v_{out} , we can see I_B as an input referred offset.

Therefore, SC filters will have higher DC offsets

MOSFET sizing + clock

1) Ensure T_{sw} is lower than T_{clk} :

$N = \text{constant that divides}$

$$R_{on} C_i = \frac{C_i}{2kV_{ov}} = \frac{T_{clk}}{N} = \frac{1}{N f_{clk}}$$

the clock

$$V_{ov} = V_{DD} - V_T$$

This leads to $C'_{ox} W (V_{DD} - V_T) > \frac{N f_{clk} C_i L}{\mu}$

2) Limit charge injection effects

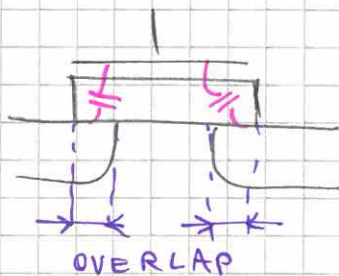
$$\alpha \frac{Q_{ch}}{C} < \Delta V_{max} \rightarrow \alpha \frac{C'_{ox} (WL) (V_{DD} - V_T)}{C} < \Delta V_{max}$$

Combine 1 + 2:

$$f_{clk} < \frac{C}{C_i} \frac{\Delta V_{max} \mu}{L_{min}^2} \cdot \frac{1}{N \alpha} \rightarrow W < \frac{C}{C'_{ox} (V_{DD} - V_T)} \frac{\Delta V_{max}}{\alpha L_{min}}$$

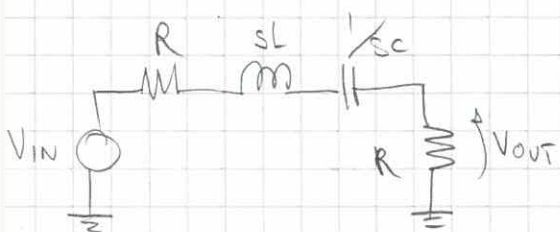
By selecting max transient time and the lowest effect allowable upper limits for f_{clk} and W (if we use L_{min})

Note: overlap between gate and drain, source is there not to have issues with potential barriers



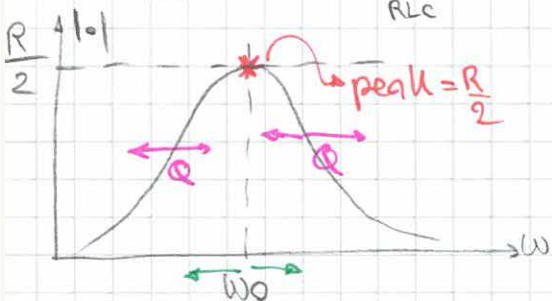
This generates additional parasitic capacitances

Additional: Orchard Theorem example



$$\frac{V_{OUT}}{V_{IN}} = \frac{R}{2R + \frac{1}{sC} + sL} = \frac{sRC}{s^2LC + s2RC + 1}$$

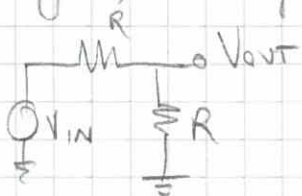
$$\omega_0 = \frac{1}{\sqrt{LC}} \quad Q \Big|_{\text{SERIES RLC}} = \frac{1}{\omega_0 2R \cdot C}$$



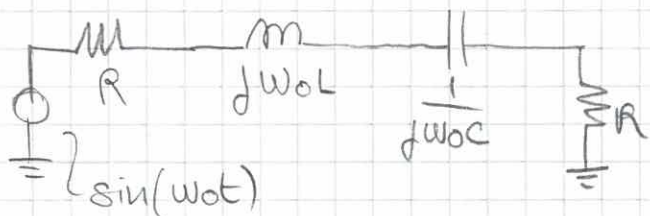
for a 10% value shift, ω_0 changes by $\sim 5\%$ ($\frac{1}{\sqrt{C \cdot L}} = \omega_0$), Q also changes

but the peak value isn't influenced

at all. In fact, at resonance, the reactance of the capacitor and inductor will have equal value but opposite signs, therefore they will be cancelled out, leading to:



To be more clear, at resonance ω_0 :



$$L = 15 \mu\text{H}$$

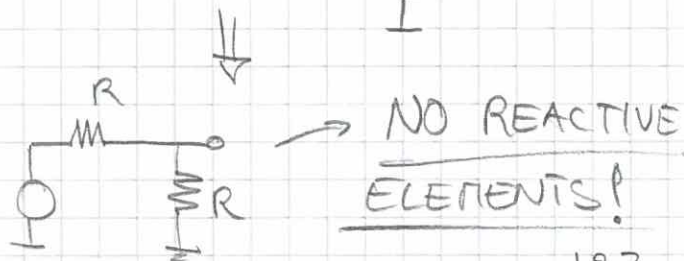
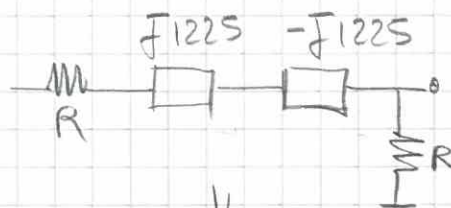
$$C = 10 \text{ pF}$$

$$\omega_0 = \frac{1}{\sqrt{LC}} = 2\pi (13 \text{ MHz})$$

At 13 MHz:

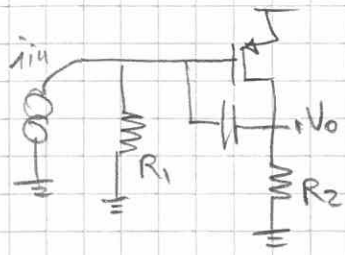
$$j \cdot 2\pi \cdot 13 \cdot 10^6 \cdot 15 \mu\text{H} \approx j 1225$$

$$\frac{1}{j 2\pi \cdot 13 \cdot 10^6 \cdot 10 \cdot 10^{-12}} = -j 1225$$

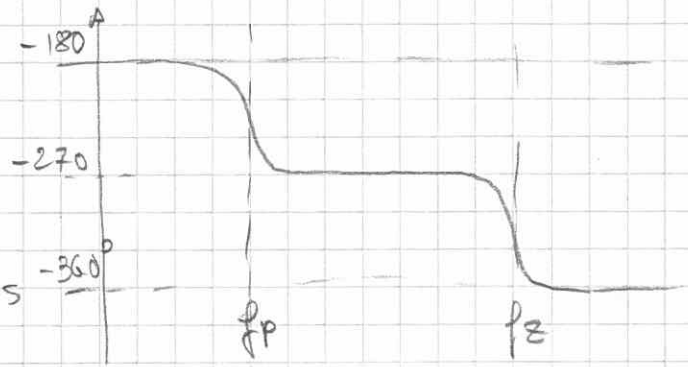
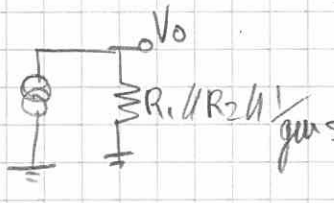


Additional: Why the two stages zero is negative?

Consider just C_{π} :



$f \rightarrow \infty$



At 0 Hz: gain is negative $\rightarrow \phi = -180^\circ$

At ∞ Hz: gain is positive $\rightarrow \phi = -360^\circ$ or 0°

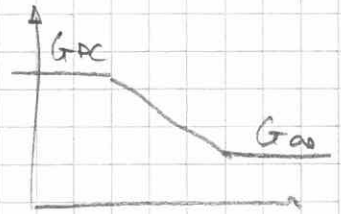
From circuit inspection we find the usual pole at

$$f_L = \frac{1}{2\pi g_m R_1 R_2 C_{\pi}}$$

and the gain

$$G_{DC} = -R_1 g_m R_2$$

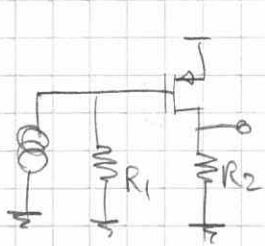
$$G_{\infty} = +\frac{1}{g_m}$$



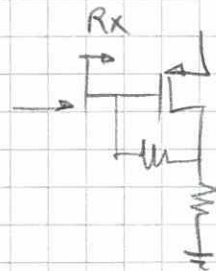
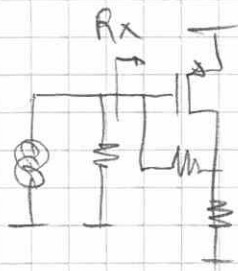
Therefore $G_{DC} f_L = G_{\infty} f_z \quad f_z = \frac{g_m}{2\pi C_{\pi}}$

To have $V_{out} < 0$ at 0 Hz and $V_{out} > 0$ at ∞ Hz, the only way to achieve this is to have a RHP zero

With nulling resistor:



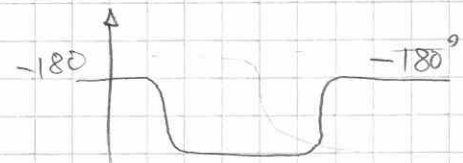
$f \rightarrow \infty$



$$R_x = \frac{R_2 + R_N}{1 + g_m R_2} \approx \frac{1}{g_m}$$

$$G_{DC} = R_1 g_m R_2$$

$$G_{\infty}: V_{out} = i_{in} \cdot (R_1 \parallel R_x) - i_{in} \cdot \frac{R_1}{R_1 + R_x} \cdot R_N = i_{in} \left(\frac{1}{g_m} - R_N \right)$$



If $R_N > \frac{1}{g_m}$, V_{out} at ∞ is $< 0 \rightarrow$ LHP zero recovers the 90° lost

If $R_N < \frac{1}{g_m}$, V_{out} at ∞ is $> 0 \rightarrow$ RHP zero again

Additional: Solving T_{sw}

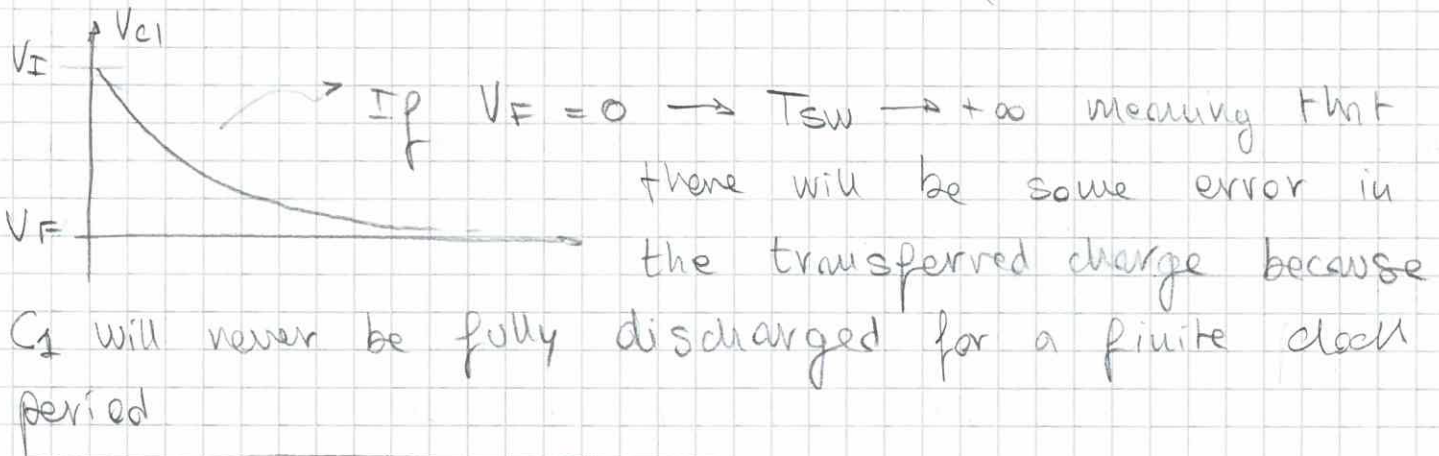
$$T_{sw} = \frac{C_1}{k} \int_{V_i}^{V_f} \frac{1}{2V_{ov}x - x^2} dx \Rightarrow \frac{1}{x(2V_{ov}-x)} = \frac{A}{x} + \frac{B}{2V_{ov}-x}$$

$$\frac{A(2V_{ov}-x) + Bx}{1} = \frac{1}{x(2V_{ov}-x)} \quad A2V_{ov} + x(B-A) = 1$$

$$\begin{cases} 2V_{ov}A = 1 \\ B-A = 0 \end{cases} \rightarrow A = B = \frac{1}{2V_{ov}}$$

$$\int_{V_i}^{V_f} \left(\frac{1}{2V_{ov}} \cdot \frac{1}{x} + \frac{-1}{2V_{ov}} \cdot \frac{-1}{2V_{ov}-x} \right) dx =$$

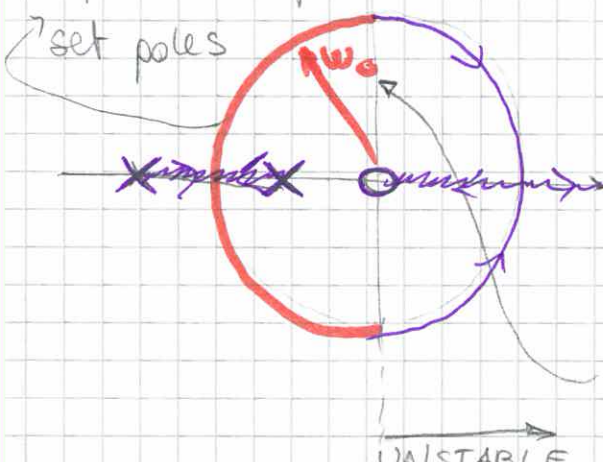
$$\frac{1}{2V_{ov}} \left[\ln x - \ln(2V_{ov}-x) \right]_{V_i}^{V_f} = \frac{1}{2V_{ov}} \ln \left\{ \frac{(2V_{ov}-V_f)V_i}{(2V_{ov}-V_i)V_f} \right\}$$



Additional: Why we use the SK cell

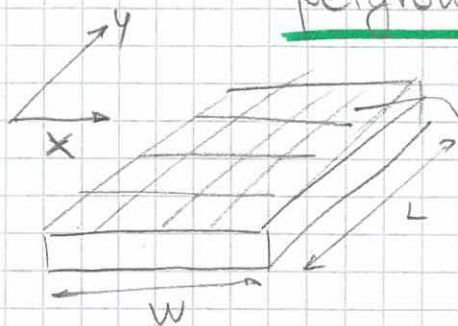
We have multiple ways to implement complex conjugate poles, but the SK cell can:

- useful locus part to
- low component number
- Q can be independently set by just using k (ω_0 is the same)
- just one amp for two complex conjugate poles



$$\omega_0 = \frac{1}{RC} \quad Q = \frac{1}{3-k}$$

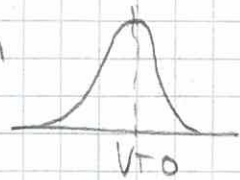
Additional: how do we choose the "elementary cells" in perlgrom formulas



This is a $\Delta x \times \Delta y = A_0$ square
 Δx length Δy area

How do we choose A_0 ? (This is valid for both K_V and K_R calculations).

Each square must be spatially uncorrelated with the adjacent squares.



If the square was too large, it wouldn't be a gaussian distribution anymore, therefore all the statistical reasoning we make would be affected by deterministic processes. In few words:

$$\left. \begin{aligned} \sigma_{V_T}^2 &= \sigma_{V_{T1}}^2 + \sigma_{V_{T2}}^2 + \sigma_{V_{T3}}^2 + \dots \\ \sigma_{G_T}^2 &= \sigma_{G_1}^2 + \sigma_{G_2}^2 + \sigma_{G_3}^2 + \dots \end{aligned} \right\} \rightarrow \text{These wouldn't be valid anymore}$$

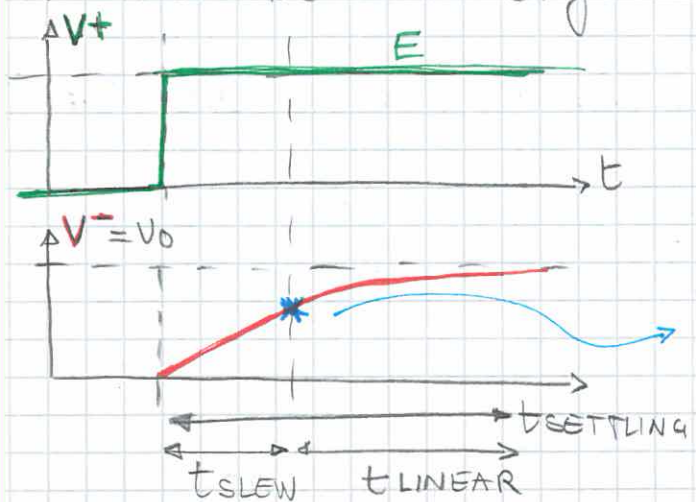
To select the right A_0, Δ we would need to compute the double autocorrelation (across x and y axis) in order to have uncorrelated squares (or at least, autocorrelation length is small enough).

Additional: notes on SR

We have an OTA buffer



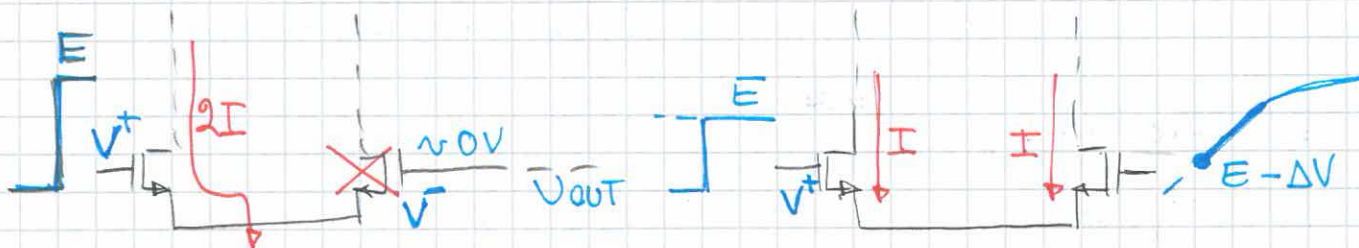
At the start we have the SR limitation (see SR question). Right in this time, we can say that the big input (on V^+ node) step will shut off the MOSFETs and etc, etc... But after the t_{SLEW} , when V_{out} has risen enough, V^+ won't be that big with respect to V^- :



$V^+ = E$ while $V^- = E - \Delta V$

This means that V^+ and V^- are similar enough so

that the differential stage will recover the linear operation:

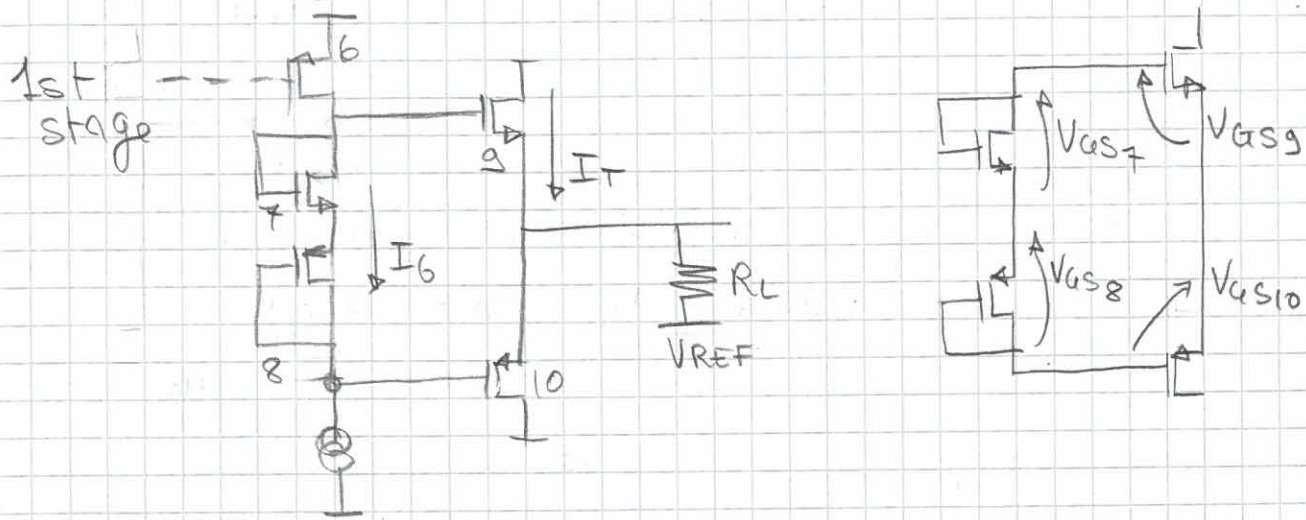


At the start of SR

When linearity is recovered

Condition will be met when $\frac{\Delta V}{2} = SR$

Additional: class AB voltage shifter



We could think of using just nMOS as transdiodes, but if we mimic the output configuration we can compensate process variability on V_T

$$V_{gs7} + V_{gs8} = V_{gs9} + V_{gs10}$$

$$\cancel{V_{Tn}} + \sqrt{\frac{I_7}{K_7}} + \cancel{V_{Tp}} + \sqrt{\frac{I_8}{K_8}} = \cancel{V_{Tn}} + \sqrt{\frac{I_9}{K_9}} + \cancel{V_{Tp}} + \sqrt{\frac{I_{10}}{K_{10}}}$$

$$\left. \begin{aligned} \sqrt{\frac{I_7}{K_7}} + \sqrt{\frac{I_8}{K_8}} &= \sqrt{\frac{I_9}{K_9}} + \sqrt{\frac{I_{10}}{K_{10}}} \\ I_7 = I_8 = I_6 \\ I_9 = I_{10} = I_T \end{aligned} \right\} \rightarrow I_T = n I_6$$

$$\sqrt{\frac{I_T}{I_6}} = \frac{\sqrt{\frac{1}{K_{10}} + \frac{1}{K_9}}}{\sqrt{\frac{1}{K_7} + \frac{1}{K_8}}} \quad \begin{aligned} K_7 &= n K_9 \\ K_8 &= n K_{10} \end{aligned}$$

$$\frac{I_T}{I_6} = n \left(\frac{1 + \sqrt{\frac{K_9}{K_{10}}}}{1 + \sqrt{\frac{K_7}{K_8}}} \right)^2$$

Result:

If $n \nearrow$, $I_T \nearrow \rightarrow$ distortion \downarrow because output transistors are well biased.

This means that higher current

leads to more power burned $\rightarrow \eta$ of the stage \downarrow

We clearly see a tradeoff between distortion and efficiency

Additional: root locus for the RHP zero

Consider to use the OTA in a feedback configuration, then:

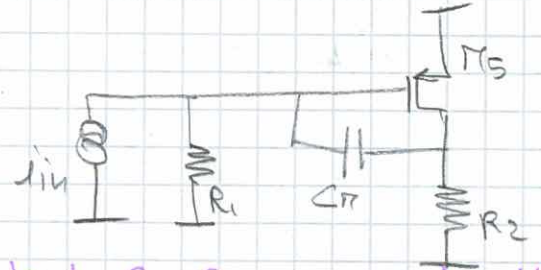
$$G_{REAL} = G_{IDEAL} \frac{-G_{loop}(s)}{1 - G_{loop}(s)}$$

denominator has roots

Root locus: locus of the singularities that solve $1 - G_{loop}(s)$ for a varying $G_{loop}(0)$. Consider the circuit below

$$G_{loop}(s) = G_0 \frac{1 - s\tilde{\tau}_z}{1 + s\tilde{\tau}_p}$$

$$\tilde{\tau}_z = \frac{C_{in}}{g_{m5}} \quad \tilde{\tau}_p \approx C_{in} R_1 g_{m5} R_2$$



Neglect C_1, C_L , we can do the same reasoning with those

$$1 - G_{loop}(s) = 0 \rightarrow G_0 \frac{(1 - s\tilde{\tau}_z)}{1 + s\tilde{\tau}_p} = 1 \rightarrow -G_0 \frac{\tilde{\tau}_z}{\tilde{\tau}_p} \cdot \frac{s - \frac{1}{\tilde{\tau}_z}}{s + \frac{1}{\tilde{\tau}_p}} = 1$$

The solution of this is split in two equations

$$\frac{|G_0| \frac{\tilde{\tau}_z}{\tilde{\tau}_p} \left| \frac{s - \frac{1}{\tilde{\tau}_z}}{s + \frac{1}{\tilde{\tau}_p}} \right|}{1} = 1 \rightarrow \text{absolute value equation}$$

$$\arg(-G_0 \frac{\tilde{\tau}_z}{\tilde{\tau}_p}) + \arg(s - \frac{1}{\tilde{\tau}_z}) - \arg(s + \frac{1}{\tilde{\tau}_p}) = 0^\circ$$

phase equation

Note that $G_0 < 0, \tilde{\tau}_z > 0, \tilde{\tau}_p > 0 \rightarrow G_x = -G_0 \frac{\tilde{\tau}_z}{\tilde{\tau}_p} > 0$ so

$$G_x \text{ is real and } > 0 \rightarrow \arg(G_x) = \tan^{-1}\left(\frac{0}{G_x}\right) = 0^\circ$$

$$\arg\left(j\omega - \frac{1}{\tilde{\tau}_z}\right) = \tan^{-1}\left(\frac{\omega}{-\frac{1}{\tilde{\tau}_z}}\right) = -\tan^{-1}(\tilde{\tau}_z \omega)$$

We can rewrite the requirement

$$\arg\left(j\omega + \frac{1}{\tilde{\tau}_p}\right) = \tan^{-1}\left(\frac{\omega}{\frac{1}{\tilde{\tau}_p}}\right)$$

on phase

zero gives a -90° contribution!!

$$0^\circ = 0^\circ - \tan^{-1}(\omega \tilde{\tau}_z) - \tan^{-1}(\omega \tilde{\tau}_p)$$

$$\frac{|G_0| \frac{\tilde{\tau}_z}{\tilde{\tau}_p} \left| \frac{s - \frac{1}{\tilde{\tau}_z}}{s + \frac{1}{\tilde{\tau}_p}} \right|}{1} = 1$$

