

MOSFET FIGURES OF MERIT

Let's consider a MOSFET n-type with source and bulk shorted to ground while the gate is biased at $V_G > V_T$. The drain is biased with $V_D > 0$. In the channel we have

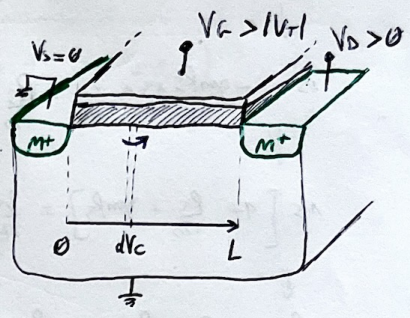
$$dV_C = I_{DS} \cdot dR = I_{DS} \cdot \frac{dx}{q \mu_n C_{ox} W \Delta} = I_{DS} \cdot \frac{dx}{Q'_n(x) \mu_n W}$$

where $Q'_n(x) = C_{ox} (V_G - V_C - V_T)$

$$\int_0^{V_{DS}} W \mu_n C_{ox} (V_G - V_C - V_T) dV_C = \int_0^L I_{DS} dx$$

$$I_{DS} = \mu_n C_{ox} \frac{W}{L} \left[(V_G - V_T) V_{DS} - \frac{V_{DS}^2}{2} \right]$$

$$I_{DS}^{SAT} = \frac{1}{2} \mu_n C_{ox} \frac{W}{L} (V_{GS} - V_T)^2$$



For $V_{DS} \text{ value} > V_{DS}^{SAT} = (V_{GS} - V_T)$ the parabolic relation is no more valid. At steady state the current being the same is equal to the one reaching the drain side, so the drift velocity (and the electric field) must increase as we move towards the drain, i.e. $n \downarrow$ and $n \cdot v \approx \text{CONSTANT}$. In the point called PINCH-OFF we have $Q'_n = 0$ (negligible) and it is expected to see backwards as V_{DS} increases above V_{DS}^{SAT} , which is the area behind the pinch-off point the parabolic relation is still valid, so

$$I_{DS}^{SAT'} = K' \frac{W}{L'} (V_G - V_T)^2 \quad \left[K' = \frac{1}{2} \mu_n C_{ox} \right]$$

where $L' < L$, so $I_{DS}^{SAT'} > I_{DS}^{SAT}$. Assuming that $(L - L')$ remains small, as suggested by the experimental I-V curves, we get that

$$I_{DS} = I_{DS}^{SAT} + \left(\frac{\partial I_{DS}}{\partial V_{DS}} \right) \Big|_{V_{DS} = V_{DS}^{SAT}} \cdot (V_{DS} - V_{DS}^{SAT})$$

where

$$\frac{\partial I_{DS}}{\partial V_{DS}} = \frac{\partial I_{DS}^{SAT}}{\partial L} \frac{\partial L}{\partial V_{DS}^{SAT}} = - \frac{I_{DS}^{SAT}}{L} \cdot \frac{\partial L'}{\partial V_{DS}^{SAT}} \Big|_{V_{DS} = V_{DS}^{SAT}}$$

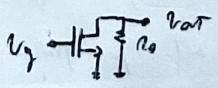
$$I_{DS} = I_{DS}^{SAT} \left[1 + \lambda (V_{DS} - V_{DS}^{SAT}) \right] \quad \left(\lambda = - \frac{1}{L} \frac{\partial L'}{\partial V_{DS}^{SAT}} \Big|_{V_{DS} = V_{DS}^{SAT}} \right)$$

We define $\frac{1}{\lambda} = V_A$ EARLY VOLTAGE.

Does not depend on current!

→ GAIN

The intrinsic gain of a MOSFET is $g_m r_{o} = \mu = \frac{2I}{V_{ov}} \cdot \frac{V_A}{I} = \frac{2V_A}{V_{ov}}$



2

RESISTANCES

• R_D

$v_o = i_s \cdot R_S$

$i_s = -g_m v_o + \frac{v_s - v_o}{r_o}$

↓

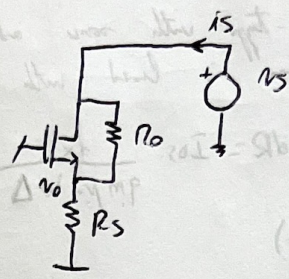
$i_s = -g_m R_S i_s + \frac{v_s}{r_o} - \frac{R_S}{r_o} i_s$

↓

$i_s \left[1 + \frac{R_S}{r_o} + g_m R_S \right] = \frac{v_s}{r_o}$

↓

$\frac{v_s}{i_s} = r_o + R_S + g_m R_S r_o = r_o + R_S [1 + g_m r_o]$



• R_S

$v_o = i_s \cdot R_D$

$i_s = g_m v_T + \frac{v_T - v_o}{r_o}$

↓

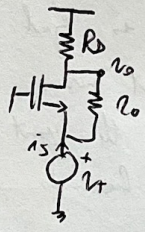
$i_s = -\frac{R_D}{r_o} i_s + \frac{v_T}{r_o} + g_m v_T$

↓

$v_T \left[g_m + \frac{1}{r_o} \right] = i_s \left[1 + \frac{R_D}{r_o} \right]$

↓

$\frac{v_T}{i_s} = \frac{1 + \frac{R_D}{r_o}}{\frac{1}{r_o} + g_m} = \frac{r_o + R_D}{1 + g_m r_o}$



WEAK INVERSION

In this regime electrons are not dominating the electrons in the channel, so the holes are almost flat. The concentration of electrons at the same node is computed by using the M-B statistics

$n(x) = N_s \cdot e^{-\frac{q\phi_B}{kT}}$

big $\phi_B = V_{GS} - \psi_s$. The unit is due to diffusion!

$I_m = q D_n \frac{dn}{dx} = q D_n \frac{n(x)}{L} = q D_n \frac{n_i^2}{L N_A} e^{\frac{q\psi_s}{kT}}$

It can be shown that

$I_{DS} = 4n \left(\frac{1}{2} \mu_{eff} C_{ox} V_{eff}^2 \right) e^{\frac{q(V_{GS} - V_T)}{kT}}$ ($m = 1 + \frac{C_{dep}}{C_{ox}} \approx 1.5$)

$\frac{\partial I_{DS}}{\partial V_{GS}} = g_m = \frac{I_{DS}}{m V_{TH}}$ and $\mu = g_m r_o = \frac{V_A}{m V_{TH}}$

→ MODERATE INVERSION

EKV model

$$I_C = \frac{I_D}{k_m \left[\frac{1}{2} \mu_m C_{ox} \frac{W}{L} (V_{TH})^2 \right]}$$

$$g_m = \frac{2}{1 + \sqrt{1 + 4 \cdot I_C}} \cdot \frac{I_D}{m V_{TH}}$$

TAXONOMY	IC	BIAS RANGE
WEAK INV.	$I_C \leq 0,1$	$V_{DS} \leq V_{TH} - 0,1V$
MODERATE INV.	$0,1 \leq I_C \leq 10$	$ V_{TH} - 0,1V \leq V_{DS} \leq V_{TH} + 0,2V$
STRONG INV.	$I_C \geq 10$	$V_{DS} \geq V_{TH} + 0,2V$

$$(m = 1 + \frac{C_{diff}}{C_{ox}} = 1,5)$$

→ CUT-OFF FREQUENCY

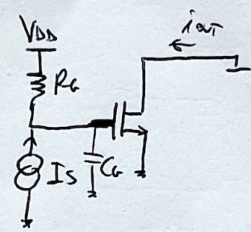
Freqny for a unity unit gain

$$\frac{i_{out}}{i_s} = g_m R_c$$

$$|f_p = \frac{1}{2\pi} \frac{1}{R_c \cdot C_c}$$

↓

$$f_T = g_m R_c \cdot \frac{1}{2\pi C_c \cdot R_c} = \frac{g_m}{2\pi C_c}$$



Assing that $C_c = C_{gs} \approx C_{ox} \cdot WL$ then

$$f_T = \frac{2 \cdot \frac{1}{2} \mu_m C_{ox} \frac{W}{L} V_{ov}}{2\pi C_{ox} WL} = \frac{1}{2\pi} \frac{\mu_m F}{L} = \frac{1}{2\pi} \frac{v}{L} = \frac{1}{2\pi} \frac{1}{\tau_{DRIFT}}$$

In weak inversion instead, we have that

$$Q' = \frac{q}{2} \cdot n(\phi) \cdot L \quad [\text{charge per unit area}]$$

$$J_n = q D_n \frac{n(\phi)}{L} \quad [\text{current density}]$$

$$\tau_{DIFF} = \frac{Q'}{J_n} = \frac{L^2}{2 D_n}$$

and $f_T = \frac{1}{2\pi \tau_{DIFF}}$

$$\tau_{DRIFT} < \tau_{DIFF} \Rightarrow \frac{L^2}{\mu_m V_{ov}} < \frac{L^2}{2 D_n} \Rightarrow V_{ov} > \frac{2 \mu_m kT}{\mu_m q} = (2 V_{TH})$$

QUANTITATIVE DESCRIPTION OF NOISE

(4)

Main sets constraints on the minimum current value. We can split any signal into

$$y(t) = \underbrace{s(t)}_{\text{INFORMATION}} + \underbrace{n(t)}_{\text{NOISE}} + \underbrace{d(t)}_{\text{DISTURB}}$$

\downarrow \downarrow \downarrow
 USEFUL UNAVOIDABLE AVOIDABLE

We'll consider Gaussian noise (Gaussian amplitude distribution), with zero mean value and parameters σ time-invariant. We consider σ as a reference value to study noise fluctuations. Moreover, we consider ERGODIC noise, where time average is indeed equal to the ensemble average. In general, noise, like signals, can be described as the superposition of orthogonal harmonics with suitable amplitudes. Let's consider a signal made of the superposition of two harmonics, its mean square value is:

$$\langle x(t)^2 \rangle = \langle A^2 \sin^2(\omega t + \phi) + B^2 \sin^2(\omega t + \psi) + 2AB \sin(\omega t + \phi) \sin(\omega t + \psi) \rangle$$

$$\downarrow$$

$$= \frac{A^2}{2} + \frac{B^2}{2}$$

and it is also equal to the variance, $\text{long} \langle x(t) \rangle = 0$. Therefore the variance is equal to the sum of the variances of each harmonic contributing to the noise $x(t)$.

$$\sigma^2 = \sum_i \sigma_i^2 = \int_0^{+\infty} S_n(f) df$$

long $S_n(f)$ the POWER SPECTRAL DENSITY.

Let's consider a resistor, whose noise is due to thermal fluctuations of carriers. We may assume that the potential fluctuations have a very broad spectrum, long then made of sequential spikes in the time domain.

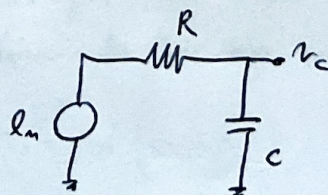
It can be concluded that each component in an electronic circuit can be modelled as an equivalent PSD generator (voltage or current), whose effect is transferred to the output and is mixed to the other contributions to noise, long then uncorrelated.

RESISTORS

As already said, to find out we may assume $S_n(f) = W$ with (WHATEVER)

Let's consider this circuit

$$\frac{V_c(s)}{I_m(s)} = \frac{1/sC}{R + 1/sC} = \frac{1}{1 + sRC}$$



⑤ So

$$\langle v_c^2 \rangle = \sigma^2 = \int_0^{+\infty} S_v(f) |T(f, \omega)|^2 df = W \cdot \int_0^{+\infty} |T(f, \omega)|^2 df$$

$$= \frac{W}{2\pi C} \int_0^{+\infty} \frac{2\pi C df}{1 + (2\pi f RC)^2} = \frac{W}{2\pi C} \left[\arctan(2\pi f RC) \right]_0^{+\infty} = \frac{W}{2\pi C} \cdot \frac{\pi}{2} = \frac{W}{4C}$$

We call $\frac{1}{4C} = \text{EN BW, EQUIVALENT NOISE BANDWIDTH}$.
 The msg inside the capacitor is

$$\frac{1}{2} C \langle v_c^2 \rangle = \left(\frac{kT}{2} \right) \quad \left[\text{EQUI PARTITION PRINCIPLE} \right]$$

$$\downarrow$$

$$\langle v_c^2 \rangle = \frac{kT}{C} = \frac{W}{4RC} \Rightarrow \boxed{W = 4kTR}$$

→ MOSFET

• Ohmic region

$$I_{DS} = \frac{1}{2} \mu_m C_{ox} \left(\frac{W}{L} \right) \left[2(V_{GS} - V_T) V_{DS} - V_{DS}^2 \right] \approx \mu_m C_{ox} \left(\frac{W}{L} \right) (V_{GS} - V_T) V_{DS}$$

$$R_{ch} = \frac{1}{g_{ch}} = \frac{1}{\mu_m C_{ox} \frac{W}{L} V_{ov}} = \frac{1}{g_m} \quad \text{like a unit channel}$$

~~...~~

$$S_v(f) = \frac{4kT}{R_{ch}} = 4kT g_m$$

• Saturation

$$S_v(f) = 4kT g_m \quad (\text{a constant for the non-inform channel})$$

$$\left(\frac{S}{N} \right)^2 = \frac{\frac{g_m^2 v_s^2}{2}}{4kT g_m \text{ BW}} = \frac{g_m}{8kT} \cdot v_s^2$$

⇒ By increasing g_m (unit) we increase $\left(\frac{S}{N} \right)^2$, ⇒ NOISE sets constraints on current.

INPUT REFERRED NOISE SOURCES

(6)

A theorem states that for a TWO-PORT LINEAR NETWORK (characterized by two pairs of input and output terminals), we can substitute the real network with an ideal noiseless one with two equivalent noise sources at the input, whose values are INDEPENDENT OF SOURCE AND LOAD IMPEDANCES.

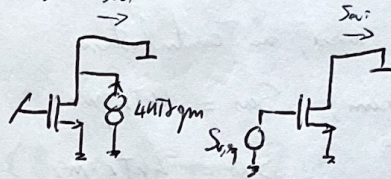
Thus, to compute $S_{i,eq}$ and $S_{v,eq}$, we may short the output port and take the noise of the short circuit as output value.

$$\begin{cases} S_{i,eq} \Rightarrow \text{open at the input} \\ S_{v,eq} \Rightarrow \text{short at the input} \end{cases}$$

→ MOSFETs

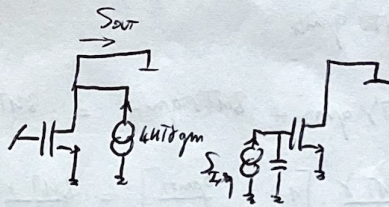
It's a 2-port network, the noise being a ~~source~~ common terminal for both input and output port.

$$\begin{cases} S_{out}(f) = 4kTR gm \\ S_{out}(f) = S_{v,eq} gm^2 \quad (\text{input shorted}) \end{cases}$$



$$\Downarrow \\ S_{v,eq} = \frac{4kTR}{gm}$$

$$\begin{cases} S_{out}(f) = 4kTR gm \\ S_{out}(f) = S_{i,eq} \cdot \frac{gm^2}{\omega^2 C_g^2} = S_{i,eq} \cdot \left(\frac{\omega_T}{\omega}\right)^2 \end{cases} \quad (\text{input open})$$

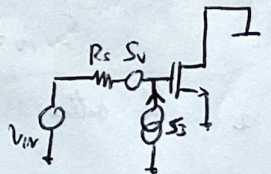


$$\Downarrow \\ S_{i,eq} = 4kTR gm \left(\frac{\omega}{\omega_T}\right)^2$$

The relative importance of the two quantities is set by the same resistance R_s

$$S_{v,IN} = S_{v,eq} + S_{i,eq} \cdot R_s^2 = \frac{4kTR}{gm} \left[1 + (gm R_s)^2 \left(\frac{\omega}{\omega_T}\right)^2 \right]$$

so for $f \leq \frac{1}{gm R_s}$ the noise contribution is negligible.



→ DIFFERENTIAL STAGE

Strictly speaking, the differential stage is not a two-port linear network, since its output is sensitive to signals on both input terminals. We may write

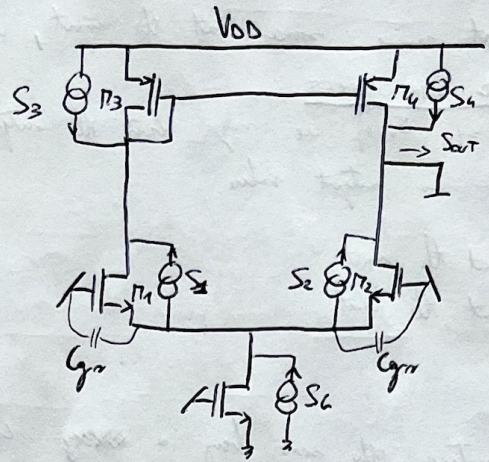
$$\begin{cases} V_1 = v_{cm} + \frac{v_d}{2} \\ V_2 = v_{cm} - \frac{v_d}{2} \end{cases} \quad i_{out} = G_{m1} V_1 - G_{m2} V_2 = v_d \left(\frac{G_{m1} + G_{m2}}{2} \right) + v_{cm} (G_{m2} - G_{m1})$$

(7) If we assume $g_{m2} = g_{m1} = g_{m0}$ then $i_{out} = g_{m0} v_d$ and the output is only proportional to the input differential voltage, \therefore the OTA becomes a 2-port network.

DEF:

$$S_{out} = S_{v,eq} \cdot g_{m1}^2$$

$$S_{out} = 4 S_{I,eq} \left(\frac{W_T}{W} \right)^2$$



in place with the capacitor path of the input, then it covers a diff pair equal to $\frac{i_{in}}{2g_m}$, \therefore
 $i_{out} = \frac{2g_m}{2g_m} i_{in}$

M_6 : common mode noise, no contribution

M_1 : splitting theorem, $S_{out} = 4g_{m1}^2 g_{m1}$

M_2 : " " , $S_{out} = 4g_{m2}^2 g_{m2}$

M_3 : $S_{out} = 4g_{m3}^2 g_{m3}$

M_4 : $S_{out} = 4g_{m4}^2 g_{m4}$

$$\Rightarrow S_{out} = 8g_{m1}^2 g_{mD} + 8g_{m2}^2 g_{mT} = 8g_{m1}^2 g_{mD} \left[1 + \frac{g_{mT}}{g_{mD}} \right] = S_{v,eq} g_{mD}^2$$

$$\Rightarrow S_{v,eq} = \frac{8g_{m1}^2}{g_{mD}} \left[1 + \frac{g_{mT}}{g_{mD}} \right] = \frac{8g_{m1}^2}{g_{mD}} \left[1 + \frac{V_{DD}}{V_{ov,17}} \right]$$

$$\Rightarrow S_{I,eq} = S_{v,eq} \cdot \frac{g_{mD}^2}{4} \left(\frac{W}{W_T} \right)^2$$

their relative importance again depends on the source resistance

$$S_{v,iv} = S_{v,eq} + S_{I,eq} \cdot \frac{g_{mD}^2}{4} \cdot R_s^2 \cdot \left(\frac{W}{W_T} \right)^2$$

\therefore the noise is dominant up to $W \sim \frac{2W_T}{g_{mD} R_s}$

→ THERMAL NOISE (DEF.)

Let's consider two resistors R_0 connected through a lossless coaxial cable, supposed to be in thermal. The cable is used to match R_0 .

The left-hand side resistor causes fluctuations that reach the resistor on the right with no reflections, being then properly matched. The same happens on the opposite ~~direction~~ direction, and then the flows are equal because of the 2nd principle of thermodynamics. (NO NET TRANSFER OF ENERGY AT TH. EQ.)

Now, let's shoot the two ends, thus trapping the energy inside. This energy is distributed along the characteristic modes inside the cable, solution of the wave equation (with boundary conditions on the voltage):

$$\lambda_n = \frac{2L}{n} \quad f_n = \frac{c}{2L} n \quad \psi_n(x) = C \quad [\text{WAVE FUNCTION}]$$

∞ , in a frequency interval df we have

$$\frac{df}{\frac{c}{2L}} \quad (\bar{B} \text{ and } \bar{E} \text{ degrees of freedom})$$

modes, each of them having two degrees of freedom, so for the equipartition principle, kT , so

$$dE = \frac{2L}{c} kT \cdot df$$

The two way flows are equal to P

$$E = 2 \cdot P \cdot T = 2 \cdot \left(\frac{V_{rms}}{2}\right)^2 \cdot \frac{1}{2R_0} \cdot \frac{L}{c}$$

$[P \stackrel{D}{=} \text{average power due to one resistor}]$

Considering the harmonics in the df interval, we get

$$dE = S_v \cdot df \cdot \frac{1}{2R_0} \cdot \frac{L}{c} = \frac{2L}{c} kT \cdot df$$

$$\left[\frac{V_{rms}^2}{2} = \langle V_{rms}^2 \rangle = S_v \cdot df \right]$$

$$\downarrow$$

$$S_v = 4kT R_0$$

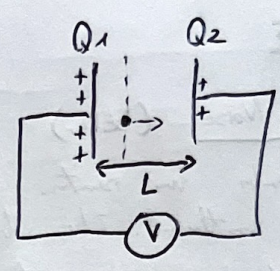
Harmonic signal

→ JUNCTION SHOT NOISE

The average number of electrons crossing a junction in a junction is represented by the average current, while the actual number per unit time is subjected to fluctuations. It can be compared to an electron being ejected from one plate of a capacitor and reaching the other. This charge causes an exponential change on the two plates ~~proportional~~ proportional to its distance from it:

9

$$\begin{cases} Q_1 = q \frac{(L-x)}{L} \\ Q_2 = q \frac{x}{L} \end{cases} \quad Q_1 + Q_2 = q$$



This induced charge covers a unit

$$\frac{dQ_2}{dt} = \left| \frac{dQ_1}{dt} \right| = \frac{q}{L} \cdot v(t) \Rightarrow \text{instantaneous carrier speed}$$

This velocity is

- ~~increasing~~ increasing proportionally to the constant electric field between the plates in the case of the charge between the capacitor's plates;
- constant if we assume the carrier to move in a semiconductor at $v = v_{SAT}$;

in both cases, we get $\int_0^T i(t) dt = q$, being it a discrete pulse of a constant duration T .

We can divide the unit in a section or the superposition of many discrete current pulses starting at random time instants, namely

$$q h(t) \quad [h(t) \triangleq \text{normalized shape of the pulse}]$$

We call

$$\lambda = \frac{I}{q} \quad \left[\begin{array}{l} \text{average number of carriers crossing the} \\ \text{Section per-unit time} \end{array} \right]$$

Let's consider a unit measured at $t = \bar{t}$, we can say that it is given by the superposition of all the pulses started at $t < \bar{t}$,

$$i(t) = q h(t_1) + q h(t_2) + \dots$$

To switch from discrete to continuous we must weight every contribution for the probability for a pulse to occur between t_1 and $t_1 + dt$, $\propto \lambda dt$

$$\langle i(t) \rangle = \int_0^{+\infty} h(t) q \lambda dt = q \lambda = I \quad \checkmark$$

Similarly we can compute the average square value: SQUARED VALUES CROSS PRODUCTS

$$i^2(t) = [q h(t_1) + q h(t_2) + \dots]^2 = q^2 h^2(t_1) + q^2 h^2(t_2) + \dots + q h(t_1) q h(t_2) + \dots$$

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

using that the variance of an equal variable is

$$\sigma_x^2 = \langle x^2 \rangle - \langle x \rangle^2$$

use my units

$$\sigma_I^2 = \langle i^2 \rangle - \langle i \rangle^2 = \int_0^{+\infty} h^2(\omega) d\omega (q^2 \lambda + (q\lambda)^2) - (q\lambda)^2 = q^2 \lambda \int_0^{+\infty} h^2(\omega) d\omega$$

$$\downarrow$$

$$= q I \int_0^{+\infty} h^2(\omega) d\omega$$

Now, using the Parseval theorem, we can write

$$\int_0^{+\infty} |h(\omega)|^2 d\omega = \int_{-\infty}^{+\infty} |H(\omega)|^2 d\omega = 2 \int_0^{+\infty} |H(\omega)|^2 d\omega$$

\downarrow
 $h(\omega)$ real $\Leftrightarrow |H(\omega)|$ even function

So, in the end, we get

$$\sigma_I^2 = 2qI \int_0^{+\infty} |H(\omega)|^2 d\omega = \int_0^{+\infty} S_I(\omega) d\omega = 2q^2 \lambda \int_0^{+\infty} |H(\omega)|^2 d\omega$$

and from the equation we derive

$$S_I(\omega) = 2qI |H(\omega)|^2$$

Finally, since our pulses have a duration T of the order of ps, then the amplitude of their Fourier transform will be ≈ 100 GHz, so to first order we can consider $|H(\omega)|^2 \approx 1$ in the BW of interest. ✓

Accounting for both physical mechanisms involved in diode's current (DRIFT + DIFFUSION) we get that

$$S_I = 2q(I + 2I_0) \quad \begin{cases} I_{DRIFT} = I + I_0 \\ I_{DIFF} = -I_0 \end{cases}$$

So, we get

REVERSE BIAS $\Rightarrow S_I = 2qI_0$

FORWARD BIAS $\Rightarrow S_I \approx 2qI$

ZERO BIAS $\Rightarrow S_I = 4qI_0 = 4qKT \frac{I_0}{hT} = 4qT qm, 0$

(11)

Finally, we can say that also the current in near-intrinsic MOSFETS is affected by hot noise, being it caused by the re-generation of current pulses generated by carriers drifting along the barrier.

$$S_I = 2q I_D = 2q I_D \cdot \frac{m V_{th}}{m V_{th}} = 2q m \frac{kT}{q} \cdot q m = 4kT \frac{m}{2} q m = 4kT \frac{m}{2} q m$$

where $\gamma = \frac{m}{2}$

→ RTN

Let's consider a resistor made of m-Si. The average current flowing through it is

$$I = \frac{V}{R} = V \frac{w \cdot \Delta}{L} q \mu n = q \mu n \cdot \frac{V}{L^2} (I \propto N)$$

being N the total number of free carriers in a volume $V = w \cdot \Delta \cdot L$

If a carrier is trapped or emitted by a trap, this causes a variation on the number of the total current. Each event causes a variation of N equal to ΔN , and so

$$\frac{\Delta I}{I} = \frac{\Delta N}{N} \Rightarrow \Delta I = \frac{I}{N} \quad (\text{for } \Delta N = 1)$$

Each trapping or emission event causes current variations that happen to recover the steady state value exponentially, since the time needed to recover it is a statistical variable. So we can write each pulse like:

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

At steady state λ (# of emission events per unit time) is equal to # of capture events per unit time. We expect

$\lambda \propto N_T$ (number of traps in the volume)

$$\lambda = \beta \frac{N_T}{\tau}$$

As already shown for the pn-junction, a noise characterized by the re-generation of well-defined current pulses of area Q can be written as

$$S_I(f) = 2\lambda Q^2 |H(f)|^2 = 2\lambda Q^2 \frac{1}{(1 + \omega^2 \tau^2)}$$

$$\downarrow = 2 \beta \frac{N_T}{\tau} \Delta I^2 \cdot \tau^2 \cdot \frac{1}{1 + \omega^2 \tau^2} = 2 \beta N_T \Delta I^2 \frac{\tau}{1 + \omega^2 \tau^2}$$

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

This contribution is small to the one due to emission events, being thus statistically independent from the capture events.

we get

$$S_I(\omega) = 4\beta N_T \left(\frac{I}{N}\right)^2 \underbrace{\frac{\tau}{(1 + \omega^2 \tau^2)}}_{\text{Lorentzian shape}}$$

where

$$\beta = \frac{\tau \partial c}{(\partial \bar{c} + \partial c)^2} \quad \tau = \frac{\tau \partial c}{\partial \bar{c} + \partial c}$$

asymmetry coefficient,
 maximum for $\partial \bar{c} = \partial c$
 and eqd to $\frac{1}{4}$
 given by states at the Fermi level

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

→ 1/f NOISE

Defects with different time constants may exist, so let's introduce the function $g(\tau)$ which is a ~~density~~ distribution of defects over the τ axis, so that

$$dN_T(\tau) = N_T g(\tau) d\tau \quad (g(\tau) \text{ is normalized})$$

therefore we can write

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

- { τ_{min} = smallest τ that can be measured, set by the BW of the apparatus
- { τ_{max} = set by the experiment duration

McWhorter pointed out that in a MOSFET free carriers can be captured by defects at the interface and recombine it, due to tunneling. Defects inside the oxide have a $\tau = \tau_0 e^{\gamma x}$ { γ = height of the barrier
{ $\tau_0 \pm \tau$ for defects at the interface

let's consider defects to be distributed uniformly all over the oxide thickness

$$n_T = \frac{N_T}{t_{ox}} \quad [\text{density per unit length}]$$

so we can write

$$n_T dx = N_T \frac{dx}{t_{ox}} = N_T g(\tau) d\tau$$

and

$$d\tau = \gamma \tau_0 e^{\gamma x} dx = \gamma \tau dx \Rightarrow dx = \frac{d\tau}{\gamma \tau}$$

so

$$g(\tau) = \frac{1}{\gamma \tau t_{ox}} \quad (\text{we've linked } g(\tau) \text{ to other quantities we need})$$

13

$$\Rightarrow S_{\Sigma}(f) = \frac{N_T}{\gamma t_{ox}} \left(\frac{I}{N}\right)^2 \int_{\omega_{min}}^{\omega_{max}} \frac{d\omega}{1 + \omega^2 C^2} = \frac{N_T}{\gamma t_{ox}} \frac{1}{W} \left(\frac{I}{N}\right)^2 \left[\arctan(\omega C) \right]_{\omega_{min}}^{\omega_{max}}$$

$$= \frac{N_T}{\gamma t_{ox}} \frac{1}{W} \left(\frac{I}{N}\right)^2 \left[\arctan(\omega_{max} C) - \arctan(\omega_{min} C) \right]$$

assuming $\omega_{max} C \gg 1$ and $\omega_{min} C \ll 1$ we get

$$S_{\Sigma}(f) = \frac{N_T}{\gamma t_{ox} W} \left(\frac{I}{N}\right)^2 \frac{\pi}{2} = \frac{N_T}{4 \gamma t_{ox} W} \left(\frac{I}{N}\right)^2 \cdot \frac{1}{f}$$

→ TSVIBS FORMULA

$N = C_{ox} (WL) (V_L - V_T) \cdot \frac{1}{q}$ carriers in the channel
 $N_T = M_T \cdot WL t_{ox}$ charges in the oxide

$$S_{\Sigma}(f) = \frac{M_T \cdot WL \cdot t_{ox}}{\gamma t_{ox} \cdot 4} q^2 \frac{\left[\frac{1}{2} \mu_m C_{ox} \left(\frac{W}{L}\right) (V_L - V_T)^2 \right]^2}{\left[C_{ox} (WL) (V_L - V_T) \right]^2} \cdot \frac{1}{f}$$

$$= \frac{M_T \cdot W \cdot L \cdot q^2}{4 \cdot \gamma \cdot (W \cdot L)^2} \cdot \frac{1}{f} \cdot \frac{1}{4} \mu_m^2 \cdot \left(\frac{W}{L}\right)^2 \cdot (V_L - V_T)^2$$

$$= \frac{M_T q^2}{8 \gamma L^2} \cdot \frac{\mu_m}{C_{ox}} \cdot \frac{I}{f} = \frac{K_{\Sigma}^{1/4}}{L^2} \cdot \frac{I}{f}$$

$$S_V(f) = g_m^{-2} S_{\Sigma}(f) = \frac{M_T \mu_m q^2}{8 \gamma C_{ox} L^2} \cdot \frac{I}{f} \cdot \frac{V_{ov}^2}{4 \cdot I^2} = \frac{M_T}{8 \gamma C_{ox}} \cdot \frac{1}{2} \mu_m C_{ox} \left(\frac{W}{L}\right) V_{ov}^2 \cdot \frac{1}{I} \cdot \frac{1}{C_{ox} WL}$$

$$= \frac{1}{8 \gamma C_{ox}} \cdot \frac{M_T q^2}{C_{ox} WL} \cdot \frac{1}{C_{ox} WL} = \frac{K_V^{1/4}}{C_{ox} WL} \cdot \frac{1}{f} \checkmark$$

Differential amplifier \Rightarrow amplify the potential difference, regardless of the average potential

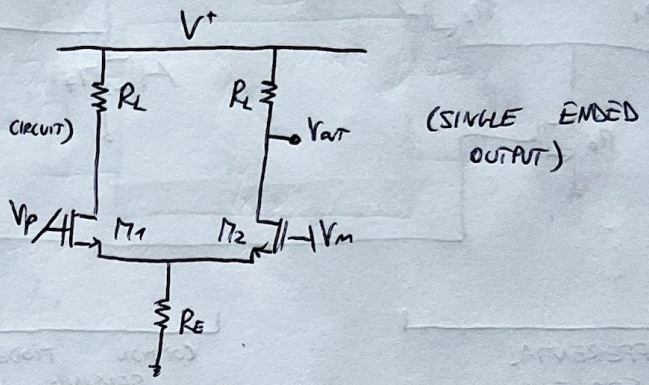
$$OVR \triangleq \left| \frac{G_d}{G_{in}} \right|$$

① RESISTANCES

$$G_d = \frac{g_m R_L}{2} = \frac{I R_L}{2 V_{ov}}$$

$$G_{in} = \frac{-v_{in}}{2R_E + \frac{1}{g_m}} \cdot R_L = -v_{in} \frac{R_L}{2R_E} \text{ (HALF CIRCUIT)}$$

$$OVR = \frac{g_m R_L}{2} \cdot \frac{2R_E}{R_L} = g_m R_E$$



$G_{d,max}$ limited by $I \cdot R_L = V_{L,max}$ ($M1$ & $M2$ may become diode)
 OVR_{max} limited by the maximum current I , which is limited by the maximum V_{EAS} .

② CURRENT GENERATORS

We may then to improve R_{out} without having to deal with high voltage drops by replacing R_L with current generators.

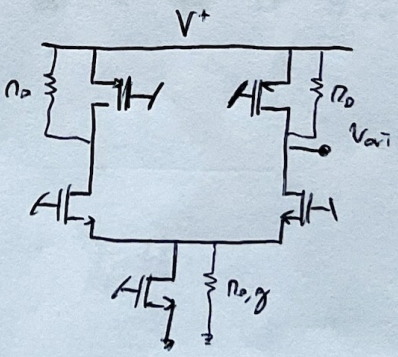
Similarly, we should also reuse R_E by substituting it with a current generator.

$$G_{in} = \frac{r_o}{2r_{o,g}}$$

$$G_d = \frac{g_m r_o}{2}$$

$$OVR = \frac{g_m r_o}{2} \cdot \frac{2r_{o,g}}{r_o} = g_m r_{o,g}$$

Maybe improve $r_{o,g}$ using CASCODES!

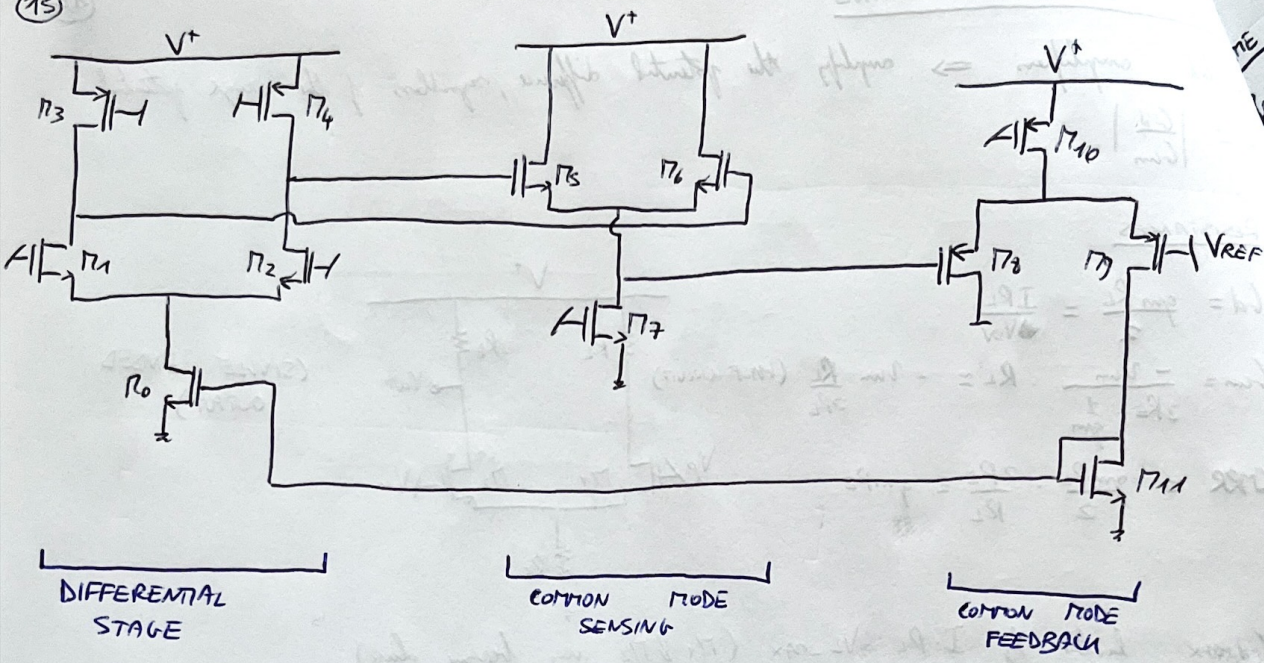


③ COMMON MODE FEEDBACK

The previous circuit indeed needs to have the MOS transistors perfectly matched: the CMOS current mirror, otherwise the intermediate drain nodes which act in order to push the upper or lower pair into diode region, until the units match. Diode transistors though have lower r_o , so lower $G_d \Rightarrow$ BAD!

We need feedback:

(15)



$$V_{out} = \frac{V_{in}}{2}$$
$$V_{out} = \frac{V_{in}}{2}$$

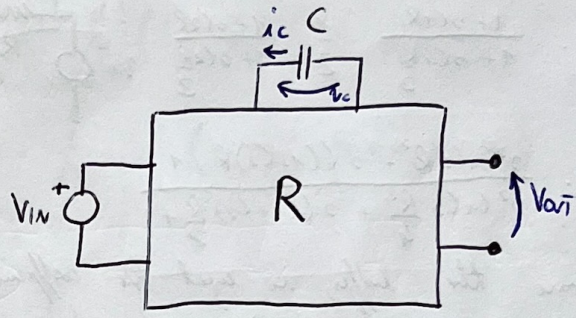
Common Mode Feedback

TIME CONSTANTS METHOD

Let's consider a linear time-invariant network with a sensitive loss of one capacitor:

INPUT VARIABLES: i_c & v_{in}

OUTPUT VARIABLES: v_{out} & v_c



$$v_{out} = A_0 v_{in} + R_m i_c$$

$$v_c = B_0 v_{in} + R_1 i_c$$

$(A_0, B_0, R_m, R_1) \in \mathbb{R}$ being the case sensitive.

Let's consider that $v_c = -\frac{i_c}{sC}$

$$v_{out} = A_0 v_{in} - sCR_m v_c$$

$$v_c = B_0 v_{in} - sCR_1 v_c$$

$$\Rightarrow v_c = \frac{B_0 v_{in}}{1 + sCR_1} ; v_{out} = A_0 v_{in} \left[1 - sC \frac{R_m B_0 / A_0}{1 + sCR_1} \right]$$

$$v_{out} = A_0 v_{in} \frac{1 + sC(R_1 - R_m B_0 / A_0)}{1 + sCR_1}$$

So we have a pole at $s = -\frac{1}{CR_1}$ and a zero at $s = -\frac{1}{C(R_1 - R_m B_0 / A_0)}$
For a zero to exist it must be

$$A_0 v_{in} + R_m i_c = 0$$

$$\downarrow$$
$$i_c = -\frac{A_0}{R_m} v_{in} \quad \text{or} \quad v_{in} = -\frac{R_m}{A_0} i_c$$

So

~~scribble~~

$$v_c = -\frac{B_0 R_m}{A_0} i_c + R_1 i_c$$

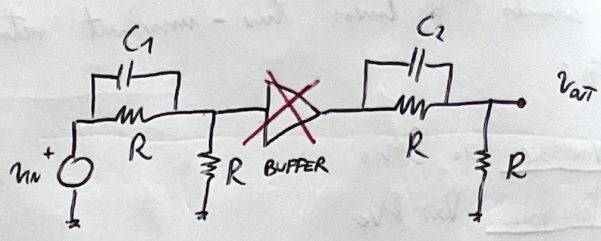
$$\downarrow$$
$$\frac{v_c}{i_c} = \left(R_1 - \frac{R_m B_0}{A_0} \right) = R_{01} \quad \text{[RESISTANCE SEEN ACROSS C when } v_{out} = 0]$$

$$\Rightarrow \frac{v_{out}}{v_{in}} = A_0 \frac{1 + sCR_{01}}{1 + sCR_1}$$

17) Let's consider now two RC networks decoupled by a buffer, the overall T.F. will be the cascade of the two:

$$T(s) = \frac{1}{2} \cdot \frac{1 + sC_1R}{1 + sC_1\frac{R}{2}} \cdot \frac{1}{2} \cdot \frac{1 + sC_2R}{1 + sC_2\frac{R}{2}}$$

$$= \frac{1}{4} \cdot \frac{s^2 C_1 C_2 R^2 + s((C_1 + C_2)R + 1)}{s^2 C_1 C_2 \frac{R^2}{4} + s((C_1 + C_2)\frac{R}{2} + 1)}$$



If we remove the buffer, we expect the coefficients to change, but not for the resistive part. Moreover, also the DC gain changes!

[$A_0 \triangleq$ GAIN WITH ALL THE CAPACITORS OPEN]

1) $C_2 \rightarrow \emptyset$

$$T(s) = A_0 \frac{sC_1 d_1 + 1}{sC_1 \beta_1 + 1}$$

and we know that $d_1 = R_{1,0}^{(0)}$ $\beta_1 = R_1^{(0)}$

~~_____~~

$$T(s) = A_0 \frac{s^2 C_1 C_2 d_{12} + s(C_1 d_1 + C_2 d_2) + 1}{s^2 C_1 C_2 \beta_{12} + s(C_1 \beta_1 + C_2 \beta_2) + 1}$$

2) $C_1 \rightarrow \emptyset$

$$T(s) = A_0 \frac{sC_2 d_2 + 1}{sC_2 \beta_2 + 1}$$

and we know that $d_2 = R_{2,0}^{(0)}$ $\beta_2 = R_2^{(0)}$

3) $C_1 \rightarrow \infty$ (a short)

$$T(s) = A_0 \frac{s^2 C_1 C_2 d_{12} + s(C_1 R_{01}^{(0)})}{s^2 C_1 C_2 \beta_{12} + s(C_1 R_1^{(0)})} = A_0 \frac{s C_1 C_2 d_{12} + (C_1 R_{01}^{(0)})}{s C_1 C_2 \beta_{12} + (C_1 R_1^{(0)})} =$$

$$= A_0 \frac{R_{01}^{(0)}}{R_1^{(0)}} \frac{s C_2 d_{12} / R_{01}^{(0)} + 1}{s C_2 \beta_{12} / R_1^{(0)} + 1}$$

and since the network can be seen as the one of a right capacitor C_2 and $C_1 \rightarrow$ a short, we may write

$$\begin{cases} \frac{d_{12}}{R_{01}^{(0)}} = R_{02}^{(1)} \\ \frac{\beta_{12}}{R_1^{(0)}} = R_2^{(1)} \end{cases} \Rightarrow \begin{cases} d_{12} = R_{01}^{(0)} R_{02}^{(1)} \\ \beta_{12} = R_1^{(0)} R_2^{(1)} \end{cases}$$

If we chose $C_2 \rightarrow \infty$ instead of C_1 , we'd get

$$\begin{cases} d_{12} = R_{02}^{(0)} R_{01}^{(1)} \\ \beta_{12} = R_2^{(0)} R_1^{(2)} \end{cases} \text{ which must be equal to the one obtained for } C_1 \rightarrow \infty, \text{ since the T.F. is unique!}$$

→ POLEZER RESULTS

Starting from a 3rd order T.F., we may write

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

$$b_1 \approx - \left(\frac{1}{p_1} + \frac{1}{p_2} + \frac{1}{p_3} \right) \approx - \frac{1}{p_1} \quad (\text{assuming } p_1 \ll p_2, p_3)$$

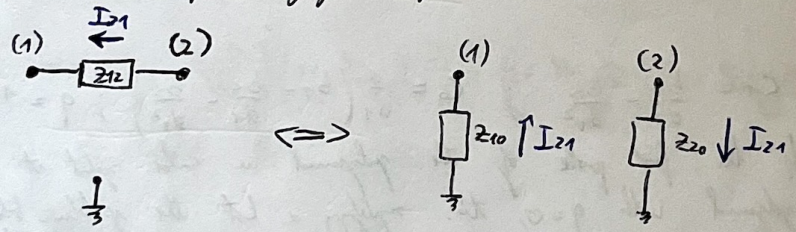
⊗ HF

$$D(s) \approx b_3 s^3 + b_2 s^2 = s^2 (b_3 s + b_2) = 0$$

$$p_3 \approx - \frac{b_2}{b_3} = \text{corner freq...}$$

→ MULLER THEOREM

We want to replace Z_{12} with two impedances Z_{10} and Z_{20} , we must set that the current entering or going out of the nodes remain unchanged



$$I_{21} = \frac{V_2 - V_1}{Z_{12}} = - \frac{V_1}{Z_{10}} \quad \left[\frac{V_2}{V_1} = K(s) \right]$$

$$\downarrow$$
$$V_2 - V_1 = - \frac{Z_{12}}{Z_{10}} V_1$$

$$\downarrow$$
$$Z_{10} = - \frac{V_1}{V_2 - V_1} \cdot Z_{12} = \frac{V_1}{V_1 - V_2} Z_{12}$$

$$\Rightarrow Y_{10} = \frac{V_1 - V_2}{V_1} \quad Y_{12} = [1 - K(s)] Y_{12}$$

$$I_{21} = \frac{V_2 - V_1}{Z_{12}} = \frac{V_2}{Z_{20}}$$

$$\downarrow$$
$$Z_{20} = \frac{V_2}{V_2 - V_1} Z_{12}$$

$$\downarrow$$
$$Y_{20} = \frac{V_2 - V_1}{V_2} Y_{12}$$

$$\Rightarrow Y_{20} = \frac{K(s) - 1}{K(s)} Y_{12}$$

→ APPROXIMATIONS AND CIRCUIT INSIGHTS

It's important to derive some estimates of the poles and zeros by using approximations in order to understand which parameters are limiting the stability. Let's consider a third order T.F. for example:

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

19) We may divide it for a first order term

$$(b_3 r^3 + b_2 r^2 + b_1 r + 1) \div (d_1 r + 1)$$

$$\overline{b_3 r^3 + b_2 r^2 + b_1 r + 1}$$

$$\underline{b_3 r^3 + \frac{b_3}{d_1} r^2}$$

$$\parallel (b_2 - \frac{b_3}{d_1}) r^2 + b_1 r + 1$$

$$\underline{(b_2 - \frac{b_3}{d_1}) r^2 + \frac{1}{d_1} (b_2 - \frac{b_3}{d_1}) r}$$

$$\parallel (b_1 - \frac{b_2}{d_1} - \frac{b_3}{d_1^2}) r + 1$$

$$(b_1 - \frac{b_2}{d_1} - \frac{b_3}{d_1^2}) r + \frac{1}{d_1} (b_1 - \frac{b_2}{d_1} - \frac{b_3}{d_1^2})$$

$$\overline{d_1 r + 1}$$

$$\frac{b_3}{d_1} r^2 + \frac{1}{d_1} (b_2 - \frac{b_3}{d_1}) r + \frac{1}{d_1} (b_1 - \frac{b_2}{d_1} - \frac{b_3}{d_1^2})$$

So we get

$$C_2 r^2 + C_1 r + C_0 + \frac{q}{d_1 r + 1}$$

where

$$C_2 = \frac{b_3}{d_1} ; C_1 = \frac{b_2}{d_1} - \frac{b_3}{d_1^2} ; C_0 = \frac{1}{d_1} (b_1 - \frac{b_2}{d_1} - \frac{b_3}{d_1^2}) ; q = 1 - \frac{1}{d_1} (b_1 - \frac{b_2}{d_1} - \frac{b_3}{d_1^2})$$

If $-\frac{1}{d_1}$ was exactly the first pole of the polynomial we could split it into a 2nd and a 1st order polynomial with $q=0$, this simplifying a lot the problem. However, even if we don't have exactly p_1 , we have that it is not by the Miller approximation, so we may say

$$p_1 \approx -\frac{1}{b_1}$$

so we get

$$C_2 = \frac{b_3}{b_1} ; C_1 = \frac{b_2}{b_1} - \frac{b_3}{b_1^2} ; C_0 = \frac{1}{b_1} [b_1 - \frac{b_2}{b_1} - \frac{b_3}{b_1^2}] ; q = \frac{1}{b_1} (\frac{b_2}{b_1} - \frac{b_3}{b_1^2})$$

and split the denominator like this

$$(C_2 r^2 + C_1 r + 1)(1 + \alpha b_1 r)$$

using $q \approx 0$. It can be shown that these results are really close to the ones obtained by choosing C_0 and equating the remaining poles using the extended time constant method.

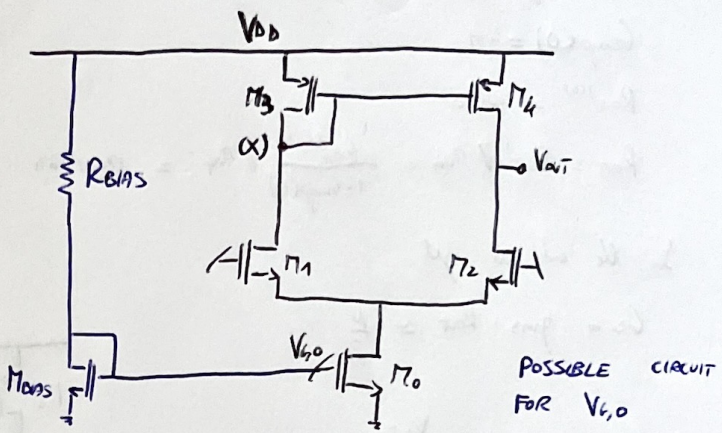
The single-ended configuration is another option to guarantee consistent bias of the stage. The adoption of the current mirror biases the current from M_4 to the current delivered by the tail, hence implementing the control loop directly in the differential stage.

In order for the bias to be consistent, we need

$$\begin{cases} I_1 = I_3 \\ I_2 = I_4 \end{cases}$$

(neglecting V_{ov} in $(V_{DS} - V_{ov})/V_A$)

$$\begin{cases} \frac{1}{2} \mu_n C_{ox} \left(\frac{W}{L}\right)_1 (V_{ov,1})^2 \left[1 + \frac{V_X - V_S}{V_A}\right] = I_1 \\ \frac{1}{2} \mu_p C_{ox} \left(\frac{W}{L}\right)_3 (V_{ov,3})^2 \left[1 + \frac{V_{DD} - V_X}{V_A}\right] = I_3 \\ \frac{1}{2} \mu_n C_{ox} \left(\frac{W}{L}\right)_2 (V_{ov,2})^2 \left[1 + \frac{V_{OUT} - V_S}{V_A}\right] = I_2 \\ \frac{1}{2} \mu_p C_{ox} \left(\frac{W}{L}\right)_4 (V_{ov,4})^2 \left[1 + \frac{V_{DD} - V_{OUT}}{V_A}\right] = I_4 \end{cases}$$



$$\frac{I_1}{I_2} = \frac{I_3}{I_4} \Rightarrow \left(1 + \frac{V_X - V_S}{V_A}\right) \left(1 + \frac{V_{DD} - V_{OUT}}{V_A}\right) = \left(1 + \frac{V_{DD} - V_X}{V_A}\right) \left(1 + \frac{V_{OUT} - V_S}{V_A}\right)$$

by inspection it turns out that $V_X = V_{OUT}$

→ VOLTAGE SWINGS

- $V_{OUT|MAX} = V_{DD} - V_{SG,3} + |V_T| = V_{DD} - V_{ov,3} - |V_T| + |V_T| = V_{DD} - V_{ov,3}$ (Saturation of M_3)
- $V_{OUT|MIN} = V_{ov,0} + V_{GS,1} = V_{ov,0} + V_{ov,1} + |V_T|$ (Saturation of M_0)
- Not symmetric! Towards ground it has to accommodate for a V_{GS} in addition to an overdrive voltage.
- $V_{OUT|MIN} = V_{OV,1MIN} - |V_T| = V_{ov,0} + V_{ov,1}$ (Saturation of M_2)
- $V_{OUT|MAX} = V_{DD} - V_{ov,4}$ (Saturation of M_4)

→ DIFFERENTIAL GAIN

Due to the introduction of the mirror, the circuit is not symmetric anymore, since the overdrives on the drain of M_1 and M_2 are different! In order to recover symmetry, we use the Norton theorem:

- in order to compute i_{cc} we need to short the drain of M_4 . In this way both the drain of M_1 and M_2 are connected to low- Z nodes, we can assume them to be σ with an error $\sim \frac{1}{g_{mcs}}$.

(21) Now we've restored symmetry and so any that diff splits only on the two transistors, this being the same state as small signal operation. Neglecting also the mirroring error, we get

$$i_{cc} = g_m v_{diff}$$

• R_{out} can be computed by noting that we have a loop acting to fix v_{out} to V_{DD} , so we may cut the loop, reconstruct the equivalent ($2R_{o2}$) and evaluate

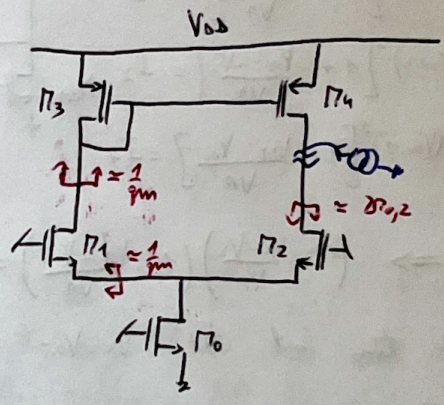
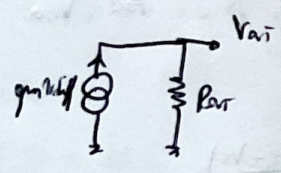
$$G_{loop}(0) = -1$$

$$R_{out}^{(0)} = 2R_{o2}$$

$$R_{out} = R_{out}^{(0)} \parallel R_{o4} = \frac{R_{out}^{(0)}}{1 - G_{loop}(0)} \parallel R_{o4} = R_{o2} \parallel R_{o4}$$

In the end we get

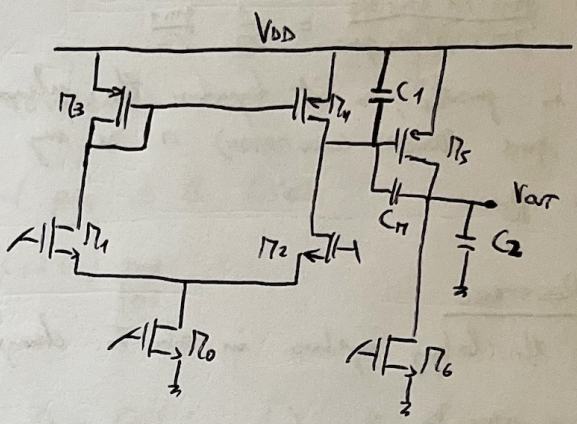
$$G_0 = g_{m3} \cdot R_{out} = \frac{\mu}{2}$$



MILLER

Split the two poles due to the two high impedance nodes by placing a C_1 bridging the gate-drain nodes of a transistor.

The three capacitors are dependent (they form a loop) \Rightarrow we expect only two poles in the T.F.:



$$b_1 = C_1 \cdot R_1^0 + C_2 R_2^0 + C_3 R_1^0$$

$$\begin{cases} R_1^0 = R_{out,1} = R_1 \\ R_2^0 = R_{out,2} = R_2 \\ R_{M1}^0 = R_1 + g_{m5} R_2 R_1 + R_2 \quad (\text{Miller effect}) \end{cases}$$

$$\Rightarrow b_1 = C_1 R_{out,1} + R_2 (C_2 + C_1) + (1 + g_{m5} R_2) C_1 R_1 = z_1 + z_2 + z_3$$

$$b_2 = C_1 C_2 R_1^0 R_2^1 + C_1 C_3 R_1^0 R_{M1}^1 + C_2 C_3 R_2^0 R_1^2$$

$$\begin{cases} R_2^1 = R_2 \\ R_{M1}^1 = R_2 \\ R_1^2 = R_1 \end{cases}$$

$$\Rightarrow b_2 = C_1 R_1 R_2 (C_2 + C_1) + R_1 R_2 C_2 C_3 = z_1 z_2 + z_4$$

$$b_3 = C_1 C_2 C_3 R_1^0 R_2^1 R_{M1}^{1,2} = 0 \quad (\text{as expected})$$

We expect also a zero when

$$\begin{aligned} \omega_{z1} C_{M1} &= g_{m5} v_s \\ \downarrow \\ z &= \frac{g_{m5}}{C_{M1}} \quad (\text{POSITIVE}) \end{aligned}$$

If the roots of the denominator are split of more than a decade, we get

$$P_2 \approx -\frac{1}{b_1} \approx -\frac{1}{R_1 g_{m5} R_2 C_1}$$

$$P_H \approx -\frac{b_1}{b_2} \approx -\frac{R_1 g_{m5} R_2 C_1}{R_1 R_2 C_1 (C_2 + C_1) + R_1 R_2 C_2 C_3} = -\frac{g_{m5} C_{M1}}{C_1 C_2 + C_1 C_3 + C_2 C_3}$$

23

For large C_{T1} values we get

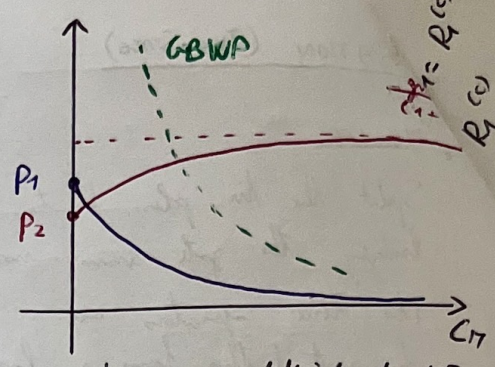
$$P_H \approx - \frac{g_{mS}}{C_1 + C_2}$$

and p_L moves to lower frequencies.

Finally, we have

$$GBWP = \frac{g_{m1} R_1 g_{mS} R_2}{2\pi C_{T1} R_1 R_2 g_{mS}} = \frac{1}{2\pi} \frac{g_{m1}}{C_{T1}}$$

The zero is positive, so it degrades the phase margin! We must shift it to HF by increasing g_{mS} (POWER CONSUMPTION) or we may increase C_{T1} in order to move p_L away from P_H .



NULLING RESISTANCE

We modify the loading impedance in order to change the zero:

$$\frac{v_S}{R_N + \frac{1}{sC_{T1}}} = g_{mS} v_S$$

$$s = - \frac{1}{C_{T1} [R_N - \frac{1}{g_{mS}}]}$$

We may size R_N in order to have a negative zero cancelling out with the second pole.

Due to R_N we have three poles

$$b_1 = C_1 R_1 + C_2 R_2 + C_C (R_1 + R_2 + g_{mS} R_1 R_2 + R_N)$$

being $R_N \approx \frac{1}{g_{mS}}$ we expect

$$b_1 \approx C_{T1} g_{mS} R_2 R_1 \quad (\text{UNCHANGED})$$

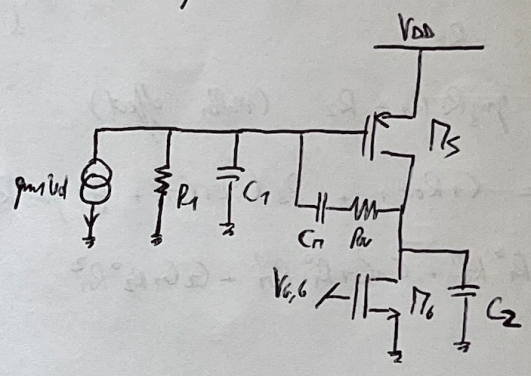
$$\Rightarrow \tau_2 \approx b_1 ; P_L = - \frac{1}{b_1}$$

We can compute τ_{M1} using the time constant method:

$$\begin{cases} \tau_1^\infty = (R_1 // R_N) C_1 \approx R_N C_1 \\ \tau_2^\infty = (R_2 // R_N) C_2 \approx R_N C_2 \\ \tau_C^\infty = R_N \end{cases}$$

$$P_H = - \left(\frac{1}{\tau_1^\infty} + \frac{1}{\tau_2^\infty} + \frac{1}{\tau_C^\infty} \right) \approx - \frac{1}{R_N (C_1 // C_2 // C_C)}$$

We can compute the second pole using that C_C is a short (this approximation is clearly justified in "APPROXIMATIONS AND CIRCUIT INSIGHTS")

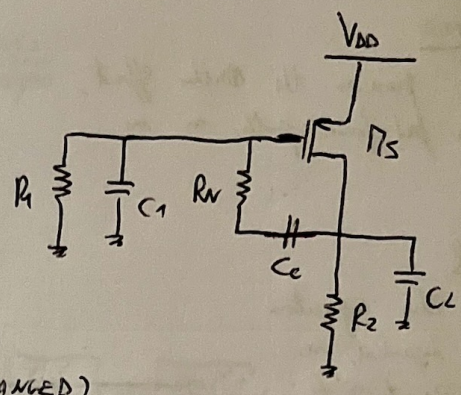


in series to the resistance across gate and drain of M_5

$b_1 = R_1^{(c)} C_1 + R_2^{(c)} C_2$

$R_1^{(c)} = R_1 \parallel \left[\frac{R_v + R_2}{1 + g_{mS} R_2} \right] \approx \frac{1}{g_{mS}}$

$R_2^{(c)} = R_2 \parallel \left[\frac{R_v + R_1}{1 + g_{mS} R_1} \right] \approx \frac{1}{g_{mS}}$



so

$P_2 \approx - \frac{1}{(C_1 + C_2) \frac{1}{g_{mS}}} = - \frac{g_{mS}}{C_1 + C_2}$ (UNCHANGED)

Therefore if we place the second pole at the GBWP:

$\frac{C_r}{g_{m1}} = \frac{C_1 + C_2}{g_{mS}} \Rightarrow C_r = (C_1 + C_2) \frac{g_{m1}}{g_{mS}}$

then

$\left[R_v - \frac{1}{g_{mS}} \right] g_{m1} = \frac{C_r}{g_{m1}} \Rightarrow R_v = \frac{C_r}{g_{m1}} \left(\frac{1}{g_{m1}} + \frac{1}{g_{mS}} \right) = \frac{2}{g_{mS}}$ (if $g_{m1} = g_{mS}$)

We may implement the resistor by using a MOSFET in diode config

$I_{DS} = \frac{1}{2} \mu_n C'_{ox} \frac{W}{L} \left[(V_{GS} - V_T) V_{DS} - V_{DS}^2 \right]$

$g_o = \left. \frac{\partial I_{DS}}{\partial V_{DS}} \right|_{V_{GS} = V} = \frac{1}{2} \mu_n C'_{ox} \frac{W}{L} (2 V_{GS} - V_{DS})$

[Just by chance is equal to the TRANSCONDUCTANCE of a MOSFET in saturation]

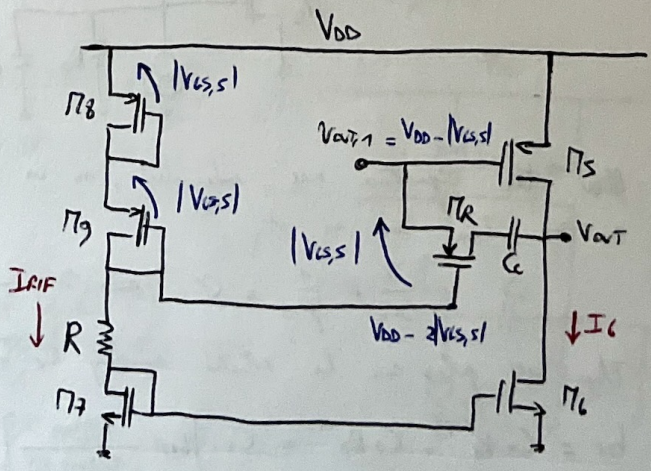
therefore we need

$\frac{g_{mS}}{g_o} = 2 \Rightarrow \frac{(W/L)_S}{(W/L)_R} \frac{V_{GS,S}}{V_{GS,R}} = 2$

If we set carefully $V_{GS,S} = V_{GS,R}$, then the accuracy depends on a ratio, which is optimal in integrated design. In order to achieve this result we need to REPLICATE the bias conditions of M_5 :

M_5 and M_6 are used in order to carry the same current with $V_{GS} = \frac{V_{DD}}{2}$. This current is set by the $\frac{1}{2}$ reference branch of M_7 , which can be set to be $\frac{1}{11}$ of I_6 by sizing the $\left(\frac{W}{L} \right)_7 = \frac{1}{11} \left(\frac{W}{L} \right)_6$. If now we choose $\left(\frac{W}{L} \right)_8 = \left(\frac{W}{L} \right)_9 = \frac{1}{11} \left(\frac{W}{L} \right)_7$, then

$|V_{GS,8}| = |V_{GS,9}| = |V_{GS,5}|$



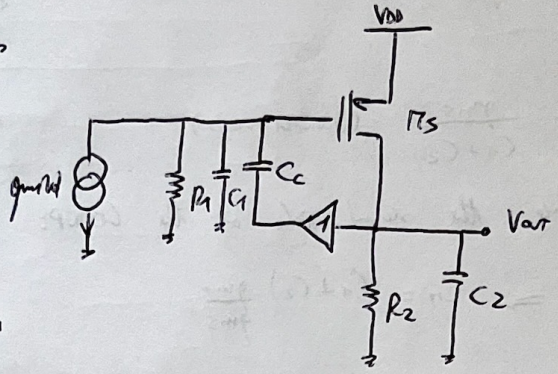
Therefore the position of the zero depends only on the ratio of $\frac{(W/L)_S}{(W/L)_R}$.

VOLTAGE BUFFER

This solution prevents the Miller effect, but kills the feed-forward path, \approx no z_{oo} !

Ideal

The three capacitors are dependent, \approx we expect just two poles. The dominant pole is still due to the Miller effect

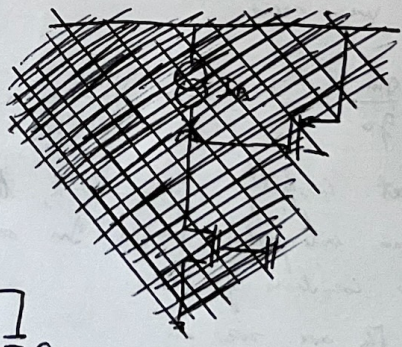
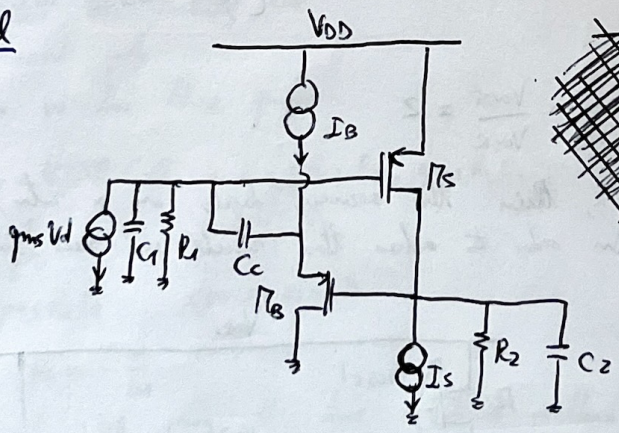


$$P_L \approx - \frac{1}{g_{mS} R_2 C_C R_1}$$

The second pole can be computed by considering C_c as a short. C_1 sees a z_{oo} resistor, which we have

$$P_H \approx - \frac{1}{C_2 \cdot \frac{1}{g_{mS}}} \quad (\text{higher, it doesn't depend on } C_1!)$$

Real



Now the capacitors are independent, so we have an additional HF pole and a z_{oo} . The z_{oo} is at

$$\frac{1}{z_c} + \frac{1}{g_{mS}} = 0 \Rightarrow z = - \frac{1}{C_c \cdot \frac{1}{g_{mS}}} \quad (\text{output unit of } r_{DS} \text{ toward ground})$$

The HF poles can be estimated considering C_c as a short

$$b_1 = C_1 R_1^{(c)} + C_2 R_2^{(c)} = C_1 \cdot \left[R_1 \parallel \frac{1/g_{mS}}{1+g_{mS} R_2} \right] + C_2 \cdot \left[\frac{R_2}{1+g_{mS} R_2} \right] \approx \frac{C_2}{g_{mS}}$$

$$b_2 = C_1 C_2 R_1^{(CO)} R_2^{(1)CO} = C_1 C_2 \frac{R_2}{g_{mB} g_{mS} R_2} = \frac{C_1 C_2}{g_{mB} g_{mS}}$$

So we get

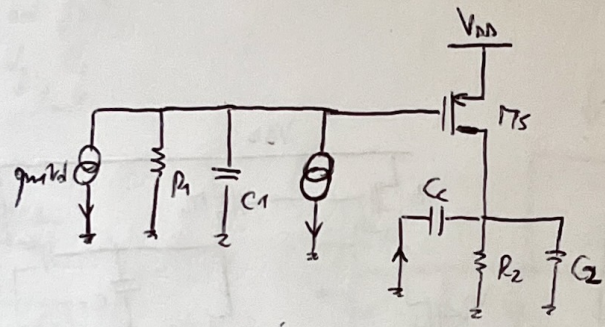
$$\left[s^2 \frac{C_1 C_2}{g_{mB} g_{mS}} + s \frac{C_2}{g_{mS}} + 1 \right] = 0$$

$$\downarrow$$

$$\omega_0 = \sqrt{\frac{g_{mB} g_{mS}}{C_1 C_2}} \quad Q = \frac{g_{mS}}{C_2} \sqrt{\frac{C_1 C_2}{g_{mB} g_{mS}}} = \sqrt{\frac{g_{mS} C_1}{g_{mB} C_2}} \quad (\text{POLE PAIR})$$

• AMUSA

→ Ideal



The Miller effect is still present ($v_c = g_{mS} v_s R_2$, $i_c = s C_c g_{mS} R_2 v_s$)
Same LF pole.

C_c and C_1 are in ||, \approx only two poles are expected. We can short C_c or a short and compute the HF pole:

$$b_1 = C_2 \cdot 0 + C_1 \cdot (R_1 \parallel \frac{1}{g_{mS}}) \approx \frac{C_1}{g_{mS}}$$

$$PH \approx - \frac{1}{C_1 \cdot \frac{1}{g_{mS}}} \quad (\text{load capacitance plays no role!})$$

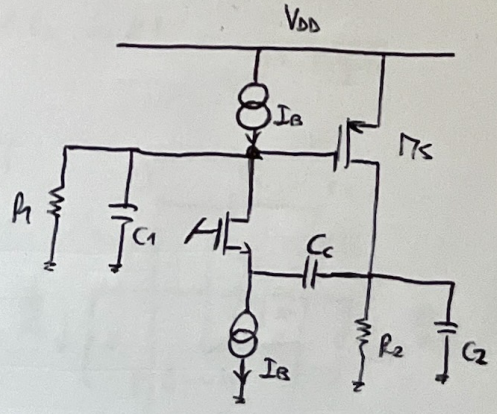
→ Real

We have a LHP zero at

$$\frac{1}{s C_c} + \frac{1}{g_{mB}} = 0$$

$$\downarrow$$

$$s = - \frac{1}{C_c \cdot \frac{1}{g_{mB}}} \quad (\text{output shorted to ground})$$



and three poles. The LF pole is still due to the Miller effect, which we can extract the HF ones by shorting C_c :

(27)

$$b_1 = C_1 \cdot [R_1 \parallel \frac{1}{g_{m3}}] + C_2 \cdot R_2 \parallel \left(\frac{1/g_{m3}}{1 + g_{m3}R_1} \right) \approx C_1 \frac{1}{g_{m3}}$$

$$b_2 = C_1 C_2 \cdot \frac{1}{g_{m3}} \cdot \frac{1}{g_{m3}}$$

So we get

$$\frac{C_1 C_2}{g_{m3} g_{m3}} \omega^2 + \omega \cdot \frac{C_1}{g_{m3}} + 1 = 0$$

$$\omega_0 = \sqrt{\frac{g_{m3} g_{m3}}{C_1 C_2}} \quad Q = \frac{g_{m3}}{C_1} \cdot \sqrt{\frac{C_1 C_2}{g_{m3} g_{m3}}} = \sqrt{\frac{C_2 g_{m3}}{C_1 g_{m3}}}$$

For the same C_2 , the Ahuja compensation has a higher Q factor than the voltage buffer compensation, so poles remain close and don't tend to split.

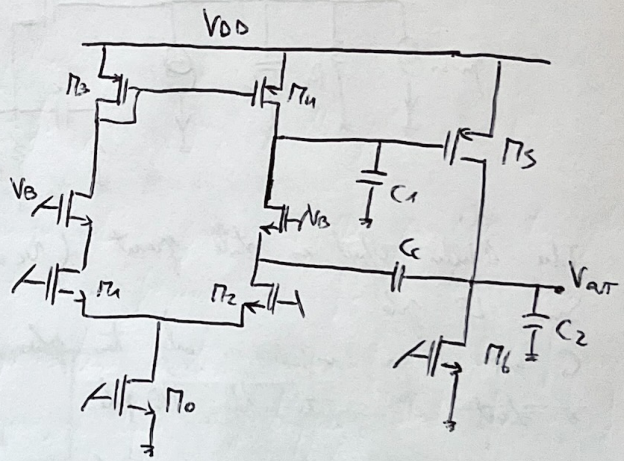
• ALTERNATIVE AHUSA (CASCODE)

Compute b_3, b_2, b_1 and a_3, a_1 and write

$$b_3 \omega^3 + b_2 \omega^2 + b_1 \omega + 1 = 0$$

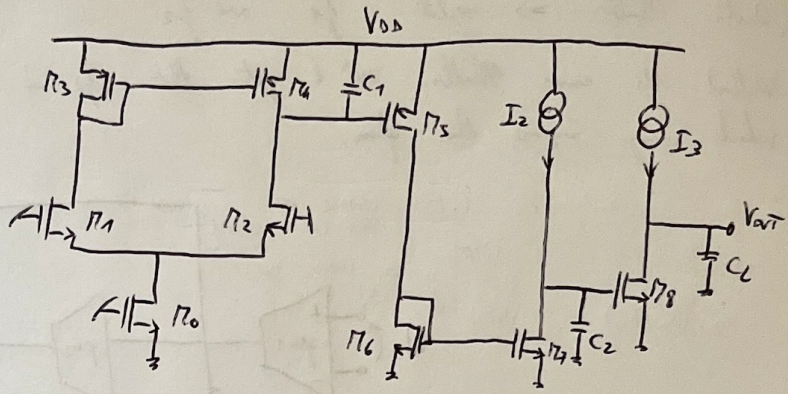
$\Downarrow \approx$

$$(1 + b_1 \omega) \left(1 + \frac{b_2}{b_1} \omega + \frac{b_3}{b_1} \omega^2 \right) = 0$$



We use three-stage OTAs any time we need a large gain and do not have enough voltage headroom to accommodate cascades:

now we have three high impedance nodes, so we need two regulated Miller compensations.



① Place a Miller capacitor across the third stage, in order to split f_2 and f_3 . Let's consider only the cascade of the last two stages:

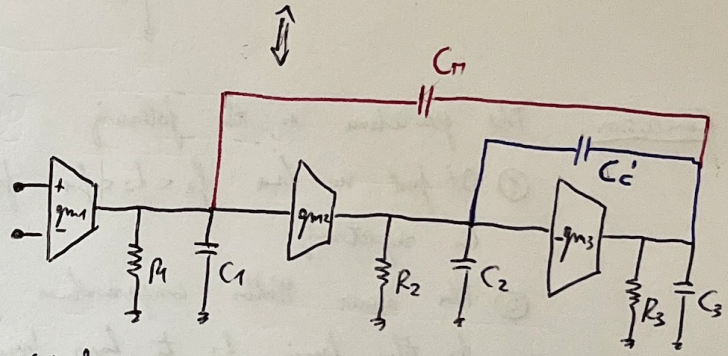
$$GBWP_{23} = \frac{gm_2}{2\pi C_c}$$

$$f_3' = \frac{gm_3}{2\pi(C_3 + C_2)}$$

$$f_2 = \frac{gm_3}{2\pi C_c} \text{ (POSITIVE)}$$

Usually $gm_3 \gg gm_2$, so we can neglect f_2 .

Set gm of the 2-3 stage by picking $f_3' = 2 GBWP_{23}$



② Place an outer Miller capacitor in order to further lower the frequency of the first pole

$$GBWP = \frac{gm_1}{2\pi C_1}$$

The second pole can be estimated by shorting C_1 :

$$R_3^{(2)} = \frac{R_1 // R_3}{1 + gm_2 R_2 gm_3 (R_1 // R_3)} \text{ (seen by } C_1 \text{ and } C_3 \text{ in //)} \approx \frac{1}{gm_2 R_2 gm_3}$$

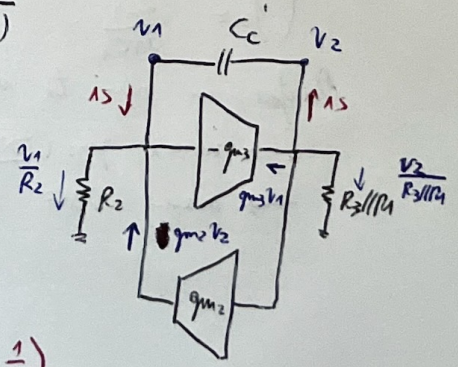
$$R_2^{(2)} = \frac{R_2}{1 + gm_2 R_2 gm_3 (R_1 // R_3)} \approx \frac{1}{gm_2 gm_3 (R_1 // R_3)}$$

$$R_{Cc}^{(2)} = \frac{1}{gm_2} - \frac{1}{gm_3}$$

$$\downarrow$$

$$= \frac{gm_3 - gm_2}{gm_3 gm_2}$$

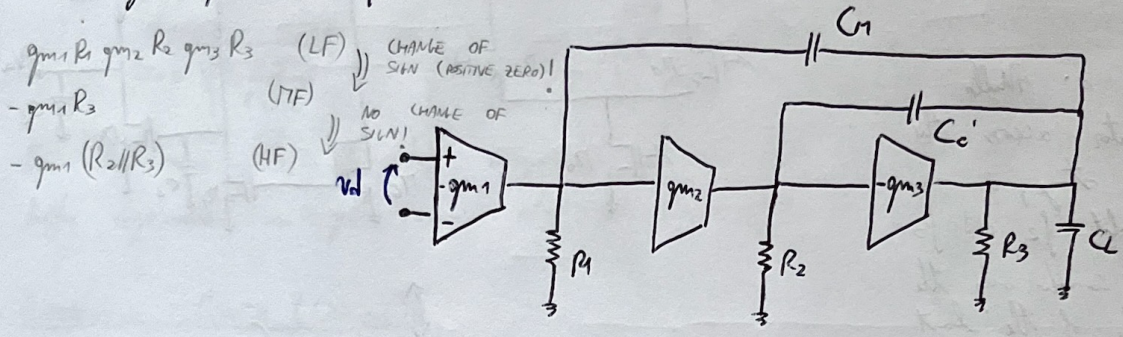
$$\begin{cases} i_3 = -gm_2 v_2 \\ i_3 = -gm_3 v_1 \\ \downarrow \\ v_1 = -\frac{i_3}{gm_3} \\ v_2 = -\frac{i_3}{gm_2} \\ v_1 - v_2 = i_3 \left(\frac{1}{gm_2} - \frac{1}{gm_3} \right) \end{cases}$$



29) A better estimate of the regulator can be achieved by neglecting C_1 and C_2 , via the capacitor loads at these nodes are dominated by the Miller capacitors.

Inner Miller \Rightarrow pre-split f_2 and f_3 (ROOT LOCUS)
 Outer Miller \Rightarrow split f_1 and f_2

Without the inner Miller we'd get that f_2 and f_3 may become a pole pair, which may degrade the ϕ_m .



CONCLUSION: The procedure is the following:

- 1) At first we have $f_3 < f_2 < f_1$ frequency of the poles due to C_3, C_2 and C_1 respectively
 - 2) The inner Miller compensation gives us the possibility to split f_3 and f_2 , this being f_2 to lower frequencies and f_3 to higher ones
 - 3) The outer Miller compensation instead moves f_1 to low frequencies and f_2 to high frequencies, so there may happen that f_2 and f_3 move closer to each other and become a pole pair.
- Neglecting the fact that the outer Miller compensation brings f_2 and f_3 closer to each other, we have that

$$\begin{cases} GBWP \approx \frac{gm_3}{2\pi C_M} \\ f_2 \approx \frac{gm_2}{2\pi C_c} \quad \left(\text{or } \frac{1}{2\pi} \frac{1}{C_c \left[\frac{gm_3 - gm_2}{gm_3 gm_2} \right]} \right) \\ f_3 \approx \frac{gm_3}{2\pi C_3} \end{cases}$$

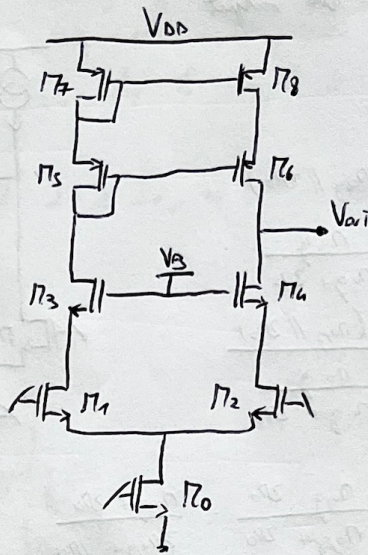
Therefore, in order to improve ϕ_m , we could either
 \rightarrow increase gm_2 (MORE POWER DISSIPATION) and gm_3
 \rightarrow decreasing GBWP by increasing C_1 (MORE SILICON AREA)

• TELESCOPIC CASCODE

This structure has a cascode both on the upper and lower side with respect to V_{ov} ,

$\Rightarrow g_{m,OUT} \approx \frac{\mu^2}{2}$

[DRAWBACK: reduced voltage swing]



$\rightarrow V_{out}/max = V_{DD} - V_{SG,8} - V_{ov,6}$

note that we may improve it by using an ENHANCED MIRROR

$\rightarrow V_{out}/min = V_B - |V_T|$

$\rightarrow V_{out}/max = V_B - V_{GS,4} + |V_T| = V_B - V_{ov}$

$\rightarrow V_{out}/min = V_{ov,0} + V_{GS,1} = 2V_{ov} + |V_T|$

Now, if we increase V_B we

INCREASE the common mode sig

DECREASE the output sig

and increase if we decrease V_B .

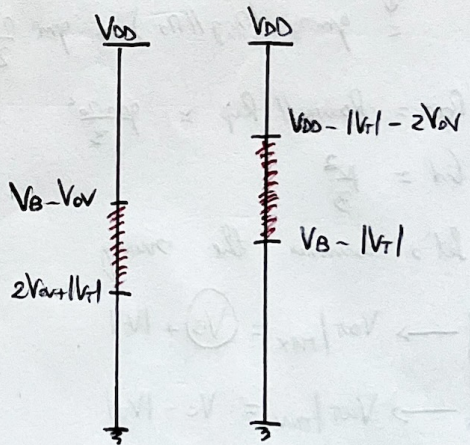
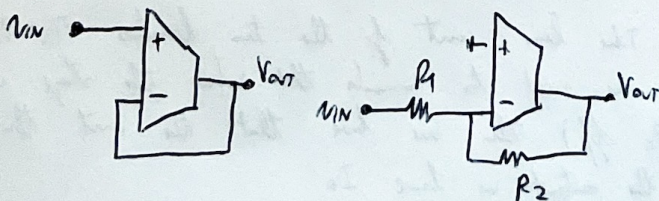
We have that

$V_{B}/max = V_{DD} - V_{SG,8} - V_{ov,6} = V_{DD} - 2V_{ov} - |V_T|$

$V_{B}/min = V_{out}/min - |V_T| + V_{ov,4} + |V_T| = 3V_{ov} + |V_T|$

We may have two cases:

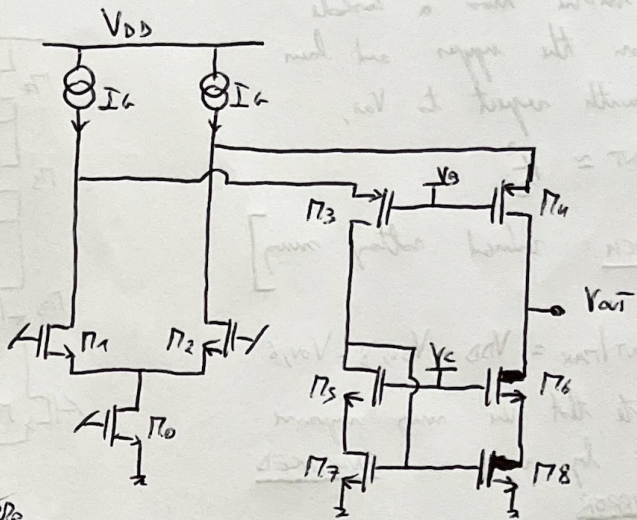
- BUFFERS, we need the input and output sig to swing as much as possible
- INERTIAL CONFIGURATION, the virtual ground doesn't pose requirements on the common mode range



31

FOLDED CASCODE

We avoid the series of many transistors in order to recover voltage swing
 Let's compute the output resistance:



$$R_{DOWN} = g_m r_o^2$$

$$R_{UP} \approx g_m r_o (r_{o3} \parallel 2r_o)$$

$$G_{loop}(\omega) = - \frac{g_m g}{g_m g + 2r_o}$$

$$R_{UP} = \frac{g_m r_o (r_{o3} \parallel 2r_o)}{1 + \frac{g_m g}{g_m g + 2r_o}}$$

$$= g_m r_o \frac{r_{o3} \cdot 2r_o}{g_m g + 2r_o} \cdot \frac{g_m g + 2r_o}{2r_o g + 2r_o} =$$

$$= g_m r_o (r_{o3} \parallel r_o) \approx g_m \frac{r_o^2}{2}$$

$$R_{out} = R_{DOWN} \parallel R_{UP} \approx \frac{g_m r_o^2}{3}$$

$$C_d \approx \frac{\mu^2}{3}$$

Let's consider the swing

$$\rightarrow V_{out} |_{MAX} = V_B + |V_T|$$

$$\rightarrow V_{out} |_{MIN} = V_C - |V_T|$$

BIG DEAL!

$$\rightarrow V_{out} |_{MAX} = V_B + |V_{S,C,3}| + |V_T|$$

$$\rightarrow V_{out} |_{MIN} = V_{D,0} + V_{C,S,1} - |V_T| + 2V_{ov}$$

Note that by increasing V_B we increase both the common mode and the output swing.

$$V_B |_{MAX} = V_{DD} - V_{ov,C} - |V_{C,S,4}|$$

$$V_B |_{MIN} = V_{out,MIN} - |V_T|$$

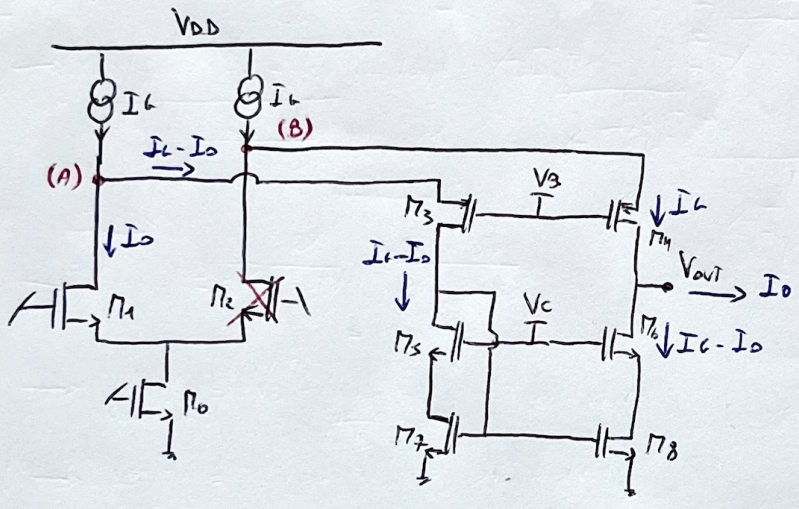
SLEW RATE PERFORMANCE

Let's consider I_0 current of M_0 . The bias current of the two branches $M_3 - M_4$ will be $I_C - \frac{I_0}{2}$. To set I_C we need to consider that when the stage is fully unbalanced (for example M_2 off) then we have that the current through M_3 will be $I_C - I_0$ and at the output we have I_0

If $I_L < I_O$, then M_3 and all his branch turn off, the potential of node (A) drops until when M_1 and M_0 go into linear, then matching $I_L = I_O'$. On the M_4 branch instead, I_L misses the potential of the output node, then switching off M_4 , or doing it into linear in order to match the needed output current, then doing also I_L linear.

In this way nodes (A) and (B) reach voltages close to supply or ground, then charging the stray capacitors and slowing down the circuit.

Therefore, not to degrade the performance, we need $I_L \geq I_O$, so more power limitations.



CMRR and offset performances of a differential amplifier are severely affected by components' tolerances. They depend on material properties variability and on dimensional parameter variability.

→ RESISTORS

$$R = \rho \frac{L}{W\Delta} = R_0 \cdot \frac{L}{W} \quad R_0 \in [10 \text{ m}\Omega/\square; 1 \text{ k}\Omega/\square]$$

$$\Delta R = \frac{L}{W} \Delta R_0 + \frac{R_0}{W} \Delta L + R_0 L \frac{\Delta W}{W^2} \quad (\text{We sum them up in abs to account for the worst case})$$

$$\downarrow$$

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

Usually negligible

We assume that ρ and Δ are subjected to variability over a characteristic spatial length $\Lambda \ll W, L$. From this conclusion, we can divide the resistor in elementary cubes of height Δ and area $A_0 = \Lambda \cdot \Lambda$. Each elementary volume will have a normal distribution of R_0 with \bar{R}_0 mean value and $\sigma(\bar{R}_0)$ standard deviation.

We can see the resistors as a matrix of $m \times n$ resistors, $m = \frac{L}{\Lambda}$ and $n = \frac{W}{\Lambda}$. Each of them being a sample of a Gaussian distribution.

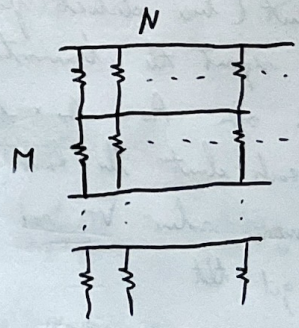
① ROWS $G_1 = \sum_{i=1}^N G_{1,i}$

$$E[G_1] = \sum_{i=1}^N E[G_{1,i}] = N \bar{G}_0$$

Max, being $G = \frac{1}{R}$ we may write

$$\frac{dG}{G} = \frac{dR}{R}, \quad \sigma^2(G) = \frac{\sigma^2(R)}{R^2}$$

assuming to deal with small variances, in abs not to have PERCOLATIVE PATHS. (higher total resistors, higher variability)



$$\sigma^2(G_1) = \sum_{i=1}^N \sigma^2(G_{1,i}) = N \sigma^2(\bar{G}_0)$$

$$\frac{\sigma(G_1)}{G_1} = \frac{\sqrt{N} \sigma(\bar{G}_0)}{N \bar{G}_0} = \frac{\sigma(\bar{G}_0)}{\bar{G}_0} \cdot \frac{1}{\sqrt{N}} \approx \frac{\sigma(R_0)}{R_0} \cdot \frac{1}{\sqrt{N}} = \frac{\sigma(R_0)}{R_0}$$

② COLUMN

$$R_{\text{row}} = \frac{1}{N \bar{G}_0} = \frac{R_0}{N}$$

$$\frac{\sigma(R_{\text{row}})}{R_{\text{row}}} = \frac{\sigma(R_0)}{R_0} \cdot \frac{1}{\sqrt{N}}$$

34) So we get that the mean value of the total resistance is

$$\bar{R} = \frac{17 R_0}{N}$$

and the standard deviation is

$$\frac{\sigma(R)}{\bar{R}} = \frac{\sqrt{17 \sigma^2(R_{row})}}{\frac{17 R_0}{N}} = \frac{\sigma(R_{row})}{\sqrt{17} R_{row}} = \frac{\sigma(R_{ij})}{R_{ij}} \cdot \frac{1}{\sqrt{17N}} = \frac{\sigma(R_{ij})}{R_{ij}} \cdot \frac{\sqrt{A_0}}{\sqrt{WL}}$$

$$\Rightarrow \frac{\sigma(R)}{R} = \frac{K' \Delta R/R}{\sqrt{WL}}$$

Let's consider now two nominally identical resistors R_1 and R_2 : if they are manufactured close to each other, their mismatch is mainly due to statistical variability, so we can write

$$\Delta R = \sigma(\Delta R) = \sqrt{\sigma^2(R_1) + \sigma^2(R_2)}$$

$$\frac{\sigma(\Delta R)}{R} = \sqrt{\frac{\sigma^2(R_1)}{R^2} + \frac{\sigma^2(R_2)}{R^2}} = \sqrt{\frac{2 K'^2 \Delta R/R}{WL}} = \frac{\sqrt{2} K' \Delta R/R}{\sqrt{WL}} = \frac{K' \Delta R/R}{\sqrt{WL}} \quad \checkmark$$

→ TRANSISTORS

As for resistors, also for transistors the variability due to threshold and conductivity coefficient (here electrical parameters) are dominant with respect to dimensional parameters. Let's represent the transistors area into a matrix of $M \times N$ dummy transistors with area $A_0 = L \times L$ on which the variability affects the V_T value. For each dummy the local V_T is a sample extracted from a gaussian distribution of mean value $V_{T,0}$ and standard deviation $\sigma(V_T)$

We get that

[The overall threshold is the average of all the average contributions of each dummy value]



$$V_T = E[E[V_{T,i,0}]] = V_{T,0} = \frac{1}{M \cdot N} \sum_{i=1}^M \sum_{j=1}^N V_{T,i}$$

$$\sigma^2(V_T) = E[(E[V_{T,i,0}])^2] = \frac{1}{(M \cdot N)^2} \cdot \sigma^2\left(\sum_{i=1}^M V_{T,i}\right) = \frac{1}{(M \cdot N)^2} \cdot M \cdot N \cdot \sigma^2(V_{T,0})$$

↓

$$\sigma(V_T) = \sqrt{\frac{A_0}{WL} \sigma^2(V_{T,0})} = \frac{K' \Delta V_T}{\sqrt{WL}}$$

$V_{T,i} \triangleq$ threshold of an dummy value
 $V_{T,0} \triangleq$ average threshold of an dummy value
 $\sigma(V_{T,0}) \triangleq$ standard deviation of an dummy value

and, considering two nominally equal transistors,

$$\sigma(\Delta V_T) = \frac{\sqrt{2} K' \Delta V_T}{\sqrt{WL}} = \frac{K' \Delta V_T}{\sqrt{WL}}$$

We can write, in general,

$$V_{out} = G_d V_d + G_m v_{cm} = G_d \left[v_d + \frac{G_m v_{cm}}{G_d} \right] = G_d \left[v_d + \frac{v_{cm}}{CIRR} \right]$$

so the common mode signal, together with the OVR, is significant for an input referred offset. Since OVR depends on frequency, @ HF this effect contribution may be relevant.

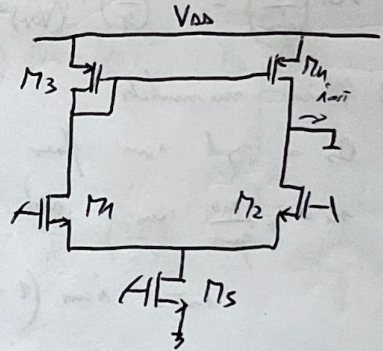
Let's assume to have an input common mode signal at the input, it is almost entirely copied at this node, being thus follows:

$$i_{cm} \approx \frac{v_{cm}}{2r_{o,g}} \quad [\text{common mode current flowing in the two branches}]$$

Due to mismatches we may have that $i_{out} = \epsilon i_{cm}$, so

$$G_m = \frac{\epsilon R_{out}}{2r_{o,g}}$$

$$CIRR = \frac{g_{mD} R_{out}}{\frac{\epsilon R_{out}}{2r_{o,g}}} = \frac{2g_{mD} r_{o,g}}{\epsilon}$$



→ DETERMINISTIC ϵ

① Mismatch error

i_{cm} in the left-hand branch is partially lost into $r_{o,M1}$ and not mirrored, thus giving



$$\epsilon = i_{cm} \frac{1/g_{mM1}}{\frac{1}{g_{mM1}} + r_{o,M1}} = i_{cm} \frac{1}{1 + g_{mM1} r_{o,M1}} \approx \frac{i_{cm}}{g_{mM1} r_{o,M1}}$$

② Unbalanced resistances at the input pair's drain nodes

$$R_{S,1} = \frac{\frac{1}{g_{mM1}} + r_{o,D}}{1 + g_{mD} r_{o,D}} =$$

$$R_{S,2} = \frac{\frac{r_{o,D}}{g_{mD}}}{\frac{1}{g_{mD}} + r_{o,D}} = \frac{r_{o,D}}{1 + g_{mD} r_{o,D}}$$

$$\epsilon i_{cm} = 2i_{cm} \frac{R_2}{R_1 + R_2} - 2i_{cm} \frac{R_1}{R_1 + R_2} = \frac{R_2 - R_1}{R_2 + R_1} 2i_{cm} = \frac{1}{2r_{o,D} + \frac{1}{g_{m1}}} 2i_{cm}$$

$$R_2 - R_1 = \frac{-\frac{1}{g_{m1}}}{1 + g_{mD} r_{o,D}}$$

$$R_2 + R_1 = \frac{\frac{r_{o,D}}{2r_{o,D} + \frac{1}{g_{m1}}}}{1 + g_{mD} r_{o,D}}$$

36

$$i_{in} \approx \frac{1}{2r_{oD} g_{m1}} \cdot 2i_{in} = \frac{i_{in}}{r_{oD} \cdot g_{m1}}$$

$$\Rightarrow \left[\epsilon_{DET} \approx \frac{1}{r_{oD} g_{m1}} + \frac{1}{r_{oT} g_{m1}} \right]$$

STATISTICAL OVER

$$d g_m = d(2k V_{ov}) = 2dk \cdot V_{ov} + 2k dV_T$$

$$\downarrow$$

$$\frac{d g_m}{g_m} = \frac{dk}{k} + \frac{dV_T}{V_{ov}}$$

Assuming that Δk and ΔV_T are statistically independent (they are not) we get

$$\sigma^2 \left(\frac{\Delta g_m}{g_m} \right) = \sigma^2 \left(\frac{\Delta k}{k} \right) + \left(\frac{1}{V_{ov}} \right)^2 \sigma^2(V_T)$$

Because for example a variation of T_{ox} multiplies C_{ox} , which is present both in k and V_T formulas

① Mismatch

As a signal i_{in} flows into n_3 , $r_{g,3} = -\frac{i_{in}}{g_{m3}}$, then we get

$$i_k = +\frac{g_{m6}}{g_{m3}} i_{in} \text{ and } \dots$$

$$i_{out} = i_{in} \left(1 - \frac{g_{m6}}{g_{m3}} \right) = \frac{g_{m3} - g_{m6}}{g_{m3}} i_{in} \approx \frac{\Delta g_{m3}}{g_{m1}} i_{in}$$

so

$$\epsilon_{IT}^{STAT} \approx \frac{\Delta g_{m3}}{g_{m1}}$$

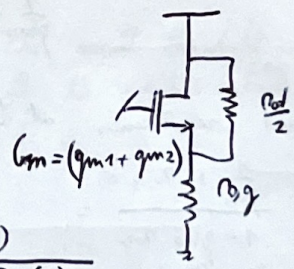
② Input mismatch

We need r_{s1} , so we FOLD our transistors on the other

$$r_{s1} = r_{in} \cdot \frac{r_{o1} \parallel \frac{r_{oD}}{2}}{r_{o1} \parallel \frac{r_{oD}}{2} + \frac{1}{2g_{m1}}}$$

$$\downarrow$$

$$= r_{in} \cdot \frac{2g_{mD} (r_{o1} \parallel \frac{r_{oD}}{2})}{1 + 2g_{m1} (r_{o1} \parallel \frac{r_{oD}}{2})}$$



$$i_2 - i_1 = i_{out} = r_{in} \frac{(g_{m1} - g_{m2})}{1 + 2g_{m1} (r_{o1} \parallel \frac{r_{oD}}{2})}$$

$$\downarrow$$

$$\approx \frac{\Delta g_{mD}}{2g_{mD}} \cdot \frac{r_{o1} + \frac{r_{oD}}{2}}{\frac{r_{o1} r_{oD}}{2}} r_{in} = \frac{\Delta g_{mD}}{g_{mD}} \frac{r_{in}}{2r_{o1}} \left(\frac{2r_{o1} + r_{oD}}{r_{oD}} \right)$$

$$\Rightarrow \epsilon_D^{STAT} \approx \frac{\Delta g_{mD}}{g_{mD}} \left(\frac{2r_{o1}}{r_{oD}} + 1 \right)$$

For $r_{o1} \rightarrow \infty$ we get OVER $\rightarrow \frac{2g_{mD} r_{o1}}{\left(\frac{\Delta g_{mD}}{g_{mD}} \right) \frac{2r_{o1}}{r_{oD}}} = \frac{g_{mD} r_{oD}}{\left(\frac{\Delta g_{mD}}{g_{mD}} \right)}$

In a single-ended OTA, in principle, the output voltage should be at $\frac{V_{DD} - V_{EE}}{2}$ if no differential input drives the two inputs. If the parameters varied deviate from nominal values, we may have that $V_{OS} \neq \frac{V_{DD} - V_{EE}}{2}$, thus causing an output voltage offset. We usually describe the component as an ideal one with an offset equivalent quantity placed between the input terminals, where

$$V_{OS} = \frac{V_{OS,OS}}{G(O)}$$

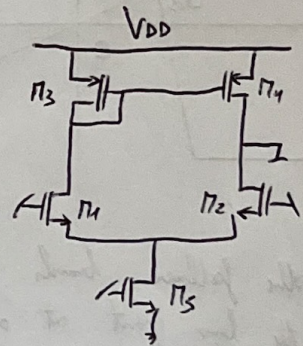
Mismatches between transistors of the input stage are more critical, being thus amplified by the total gain of the stage.

→ k MISMATCH M_1 & M_2

$$I_1 = \left(k + \frac{\Delta k}{2}\right) (V_{GS} - V_T)^2 ; \quad I_2 = \left(k - \frac{\Delta k}{2}\right) (V_{GS} - V_T)^2$$

$$|\Delta I| = |\Delta k (V_{GS} - V_T)^2| = |I_1 - I_2|$$

$$V_{OS|\Delta k} = \frac{|\Delta I|}{g_m} = \frac{V_{OV}^2 \Delta k}{g_m} = \frac{V_{OV}^2 \cdot \Delta k}{2k V_{OV}} = \frac{\Delta k}{k} \cdot \left(\frac{V_{OV}}{2}\right)$$



→ V_T MISMATCH M_1 & M_2

$$I_1 = k \left(V_{GS} - V_T - \frac{\Delta V_T}{2}\right)^2 ; \quad I_2 = k \left(V_{GS} - V_T + \frac{\Delta V_T}{2}\right)^2$$

$$|\Delta I| = \left| k \left[(V_{GS} - V_T)^2 + \frac{\Delta V_T^2}{4} - 2V_{OV} \cdot \frac{\Delta V_T}{2} - (V_{GS} - V_T)^2 + \left(-\frac{\Delta V_T^2}{4}\right) - 2V_{OV} \frac{\Delta V_T}{2} \right] \right| = 2k V_{OV} \cdot \Delta V_T$$

$$V_{OS|\Delta V_T} = \frac{|\Delta I|}{g_m} = \frac{2k V_{OV} \cdot \Delta V_T}{2k V_{OV}} = \Delta V_T$$

→ k MISMATCH M_3 & M_4

$$I_3 = \left(k_M + \frac{\Delta k_M}{2}\right) (V_{GS} - V_T)^2 ; \quad I_4 = \left(k_M - \frac{\Delta k_M}{2}\right) (V_{GS} - V_T)^2$$

$$|\Delta I| = \frac{\Delta k_M}{2} (V_{GS} - V_T)^2$$

$$V_{OS|\Delta k_M} = \frac{|\Delta I|}{g_m} = \frac{\Delta k_M}{2k_M} \frac{V_{OV,M}^2}{V_{OV,D}} = \frac{\Delta k_M V_{OV,M}^2}{2I} \frac{V_{OV,D}}{V_{OV,D}} = \frac{V_{OV,D}}{2} \left(\frac{\Delta k_M}{k_M}\right)$$

→ V_T MISMATCH M_3 & M_4

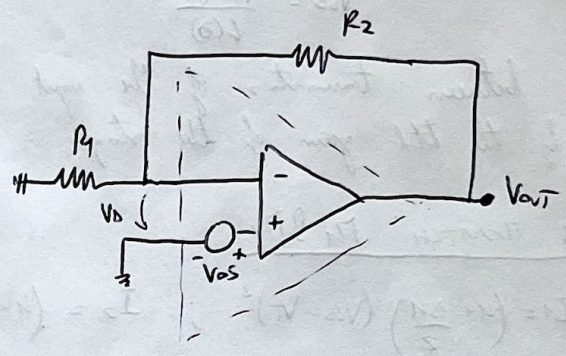
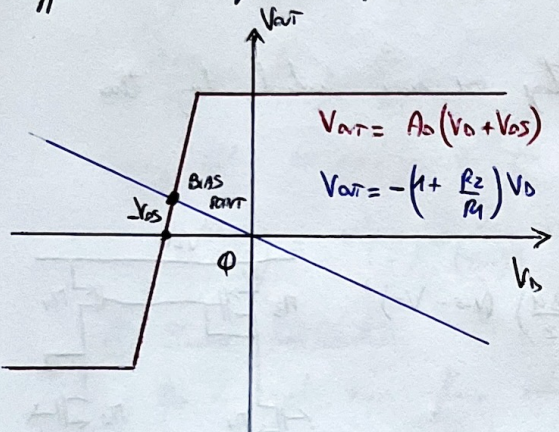
$$I_3 = k_M \left(V_{GS} - V_T - \frac{\Delta V_T}{2}\right)^2 ; \quad I_4 = k_M \left(V_{GS} - V_T + \frac{\Delta V_T}{2}\right)^2$$

$$|\Delta I| = 2k_{n1} V_{ov1,n} \cdot \Delta V_T$$

$$V_{os}|_{\Delta V_T} = \frac{2k_{n1} V_{ov1,n}}{g_m} \Delta V_T = \frac{2k_{n1} V_{ov1,n}}{2I} \cdot V_{ov,D} \cdot \Delta V_T = \frac{2I}{V_{ov1,n}} \cdot \frac{V_{ov,D} \Delta V_T}{2I} = \frac{V_{ov,D}}{V_{ov1,n}} \Delta V_T$$

$$\Rightarrow \sigma^2(V_{os}) = \sigma^2(\Delta V_{T,D}) + \sigma^2(\Delta V_{T,n}) \left(\frac{V_{ov,D}}{V_{ov1,n}} \right)^2 + \left[\sigma^2 \left(\frac{\Delta k_{n1}}{k_{n1}} \right) + \sigma^2 \left(\frac{\Delta k_{n2}}{k_{n2}} \right) \right] \left(\frac{V_{ov,D}}{2} \right)^2$$

Effect is compensated by FEEDBACK:



the feedback branch introduces a new relationship in the system, the setting the bias point at a value lower than the power supply. In order to have this situation, it must be

$$\left(1 + \frac{R_2}{R_1}\right) \ll A_D$$

$$\downarrow$$

$$A_D \frac{R_1}{R_1 + R_2} \gg 1$$

$$\underbrace{\hspace{1cm}}_{|Loop\ Gain|}$$

[Therefore in order to properly act on the feedback nodes, the loop gain must be properly sized and $\gg 1$.]

OUTPUT STAGES

→ CLASS - A (TRANSISTORS ARE ALWAYS ACTIVE)

The simplest buffer stage is the source follower, where we set the output node at $\frac{V_{DD} - V_{EE}}{2}$ in order not to have a current into R_L when no input signal is applied ² ($V_{REF} = \frac{V_{DD} - V_{EE}}{2}$)

Let's assume to have a positive signal at the gate of M_7 . The gain of the stage is

$$\frac{V_{OUT}}{V_{i,7}} = \frac{R_L}{R_L + \frac{1}{g_{m7}}}$$

where g_{m7} is NOT constant, since it depends on the current value. As $V_{i,7} \uparrow$, the current $I_{7,1} \uparrow$, so $g_{m7} \uparrow$ and overall $\left(\frac{V_{OUT}}{V_{i,7}}\right) \uparrow$. On the negative sig instead, $I_{7,1} \downarrow$ and M_7 may turn off. In order to avoid the output clipping we need to have

$$R_L \cdot I_B \geq \Delta$$

being Δ the peak (negative) voltage of the output.

This stage is also adopted in POWER AMPLIFIERS, so we may compute the

POWER EFFICIENCY $\eta = \frac{V_p^2 / 2R_L}{I_B V_{DD}} \triangleq \frac{\text{POWER DELIVERED TO THE LOAD}}{\text{AVERAGE POWER DRAWN FROM SUPPLY}}$

Assuming that no changing transes phase, and that

$$\begin{cases} V_{DD} = 2V_{REF} \\ V_p \leq V_{REF} \end{cases}$$

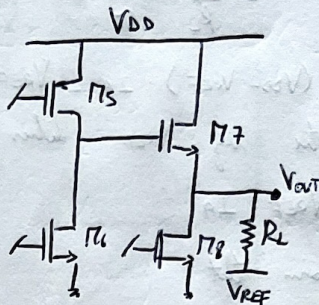
$$\Rightarrow \eta \leq \frac{V_{REF}^2}{2R_L I_B 2V_{REF}} \leq \frac{V_{REF}^2}{4 \cdot V_{REF} \cdot V_{REF}} = \frac{1}{4}$$

→ CLASS - B (PUSH - PULL)

In order to avoid static power consumption, we may substitute the current generator with a PNP follower, so that

- on the positive sig the NPNs turn on while the PNP is off
- on the negative sig the PNP turn on while the NPNs is off
- for $-V_T \leq V_{i,7} - V_{REF} \leq V_T$ we have a dead zone that causes CROSSOVER DISTORTION

therefore we have MORE DISTORTION, but BETTER POWER EFFICIENCY.



(40)

- The power delivered to the load is again $\frac{V_p^2}{2R_L}$ *neglect distortion*
- On the positive half cycle a current flows from V_{DD} to V_{REF} , whose mean value is

$$I_A = \frac{1}{T/2} \int_0^{T/2} \frac{V_p}{R_L} \sin\left(\frac{2\pi}{T} t\right) dt = \frac{V_p}{R_L} \cdot \frac{2}{T} \frac{T}{2\pi} \left[\cos\left(\frac{2\pi}{T} t\right) \right]_{T/2}^0$$

$$= \frac{V_p}{R_L} \cdot \frac{1}{\pi} = \frac{2}{\pi} \frac{V_p}{R_L}$$

$\therefore \bar{P} = (V_{DD} - V_{REF}) \cdot I_A = V_{REF} \cdot I_A$

On the negative half cycle V_{REF} delivers the same amount towards ground, \therefore again

$\bar{P} = V_{REF} \cdot I_A$

Remembering again that

$$\begin{cases} V_p \leq V_{REF} \\ V_{DD} = 2V_{REF} \\ \text{Distortion} \end{cases}$$

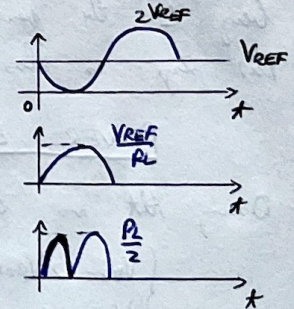
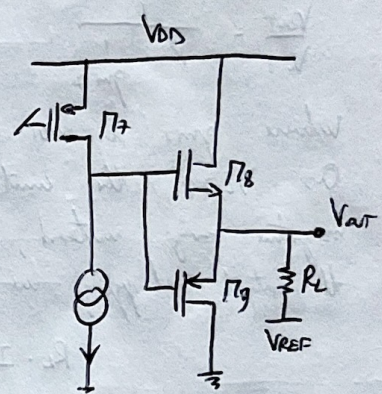
$$\eta \leq \frac{\frac{V_{REF}^2}{2R_L}}{V_{REF} \cdot \frac{2}{\pi} \cdot \frac{V_{REF}}{R_L}} = \frac{\pi}{4} \approx 78\%$$

We have that across each transistor

$$\begin{cases} V_{DS,max} = 2V_{REF} \\ I_{max} = \frac{V_{REF}}{R_L} \end{cases} \quad (\text{neglecting the overdrive})$$

$P(t) = \frac{V_{REF}}{R_L} \sin(\theta) V_{REF} [1 - \sin(\theta)]$

\downarrow
 $P_{max} = \frac{V_{REF}^2}{4R_L} \quad \left[\theta = \frac{\pi}{6} \right] \quad (\text{half of the maximum power delivered to the load!})$



Class A-B

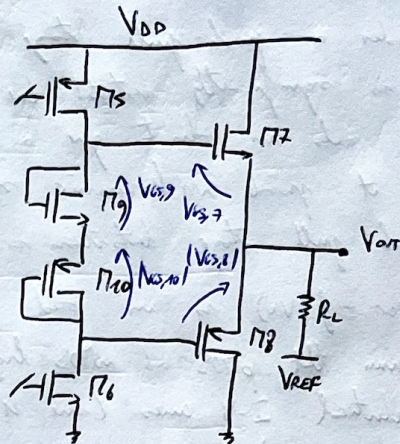
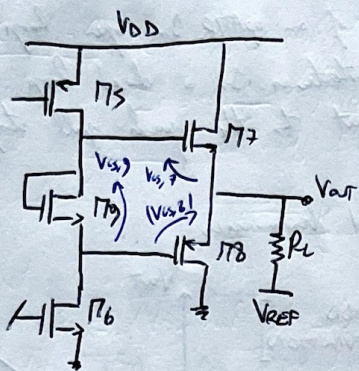
To limit distortion due to the dead-zone of the push-pull stage, we shall bias the transistors M_8 and M_9 at the edge of their full conduction.

We may implement this voltage divider by using a transdiode:

$V_{GS,7} + V_{GS,8} = V_{GS,9}$

{ DEPENDENT ON ABSOLUTE VALUES (BAD)! }

$$\sqrt{\frac{I_7}{K_7}} + V_{TH} + \sqrt{\frac{I_7}{K_8}} + V_{TP} = \sqrt{\frac{I_C}{K_C}} + V_{TH} \Rightarrow I_7 = \left(\frac{\sqrt{I_C/K_C} - V_{TP}}{\sqrt{1/K_7} + \sqrt{1/K_8}} \right)^2$$



We may use the TRANSLINEAR PRINCIPLE by adding two complementary transistors, in such a way that the KVL in the M77-M78-M79-M76 loop is depending only on V_{GS} terms and we can control more carefully the current

$$\sqrt{\frac{I_7}{k_7}} + \sqrt{\frac{I_7}{k_8}} + V_{GS7} + V_{GS79} = \sqrt{\frac{I_6}{k_9}} + \sqrt{\frac{I_6}{k_{10}}} + V_{GS77} + V_{GS76}$$

$$\downarrow$$

$$I_7 = I_6 \left(\frac{\sqrt{1/k_9} + \sqrt{1/k_{10}}}{\sqrt{1/k_7} + \sqrt{1/k_8}} \right)^2$$

So we can precisely tailor the current (static one) in M77 and M78 by sizing the transistors M79 and M76.

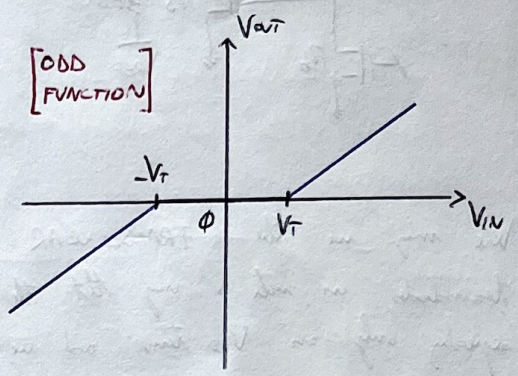
- { BIAS OF THE OUTPUT STAGE \Rightarrow set by R_L, V_{IN} (avoid clipping)
- { SIZING OF M79 AND M76 \Rightarrow set by I_7

DISTORTION AND FEEDBACK

Let's consider a push-pull stage as output stage of an opAMP. In open-loop operation, the output is significantly distorted due to crossover distortion. In a closed loop configuration, the feedback acts in order to give a pre-distorted input to the push-pull stage and thus making it provide an harmonic signal.

In a push-pull stage distortion is dominated by odd harmonics:

→ $V_{out} = f(V_{in})$ is an odd function,
 given that pMOS and nMOS have identical I-V curves (same μ and same V_T)



→ if we drive the stage with

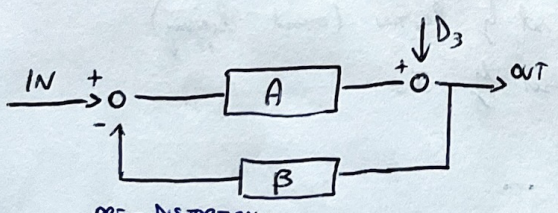
$V_{in} = A_0 \sin(\omega t)$

the signal went through M_1 is a replica of the signal went through M_2 but with a $\frac{T}{2}$ shift, $\omega T = \frac{2\pi}{\omega T}$

$i_1(t) = A_0 + A_1 \sin(\omega t + \phi_1) + A_2 \sin(\omega t + \phi_2) + A_3 \sin(\omega t + \phi_3) + \dots$

$i_2(t) = A_0 + A_1 \sin(\omega t + \phi_1 + \pi) + A_2 \sin(\omega t + \phi_2 + 2\pi) + A_3 \sin(\omega t + \phi_3 + 3\pi) + \dots$

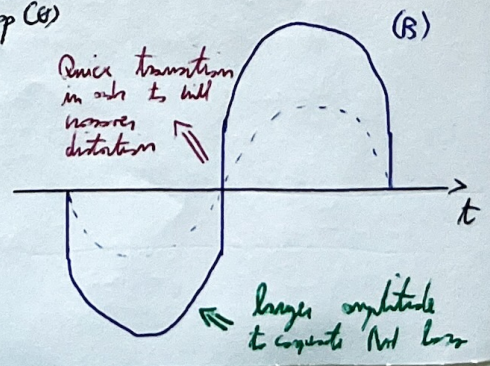
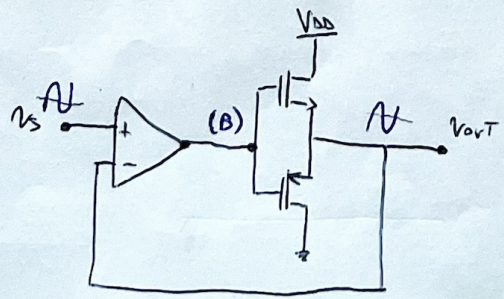
$i_L = i_1 - i_2 = 2A_1 \sin(\omega t + \phi_1) + 2A_3 \sin(\omega t + \phi_3) + \dots$



PRE-DISTORTION

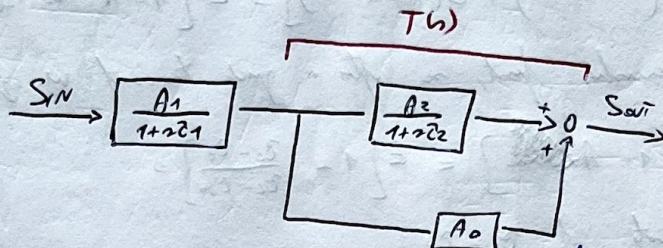
$-D_3^{out} \cdot A \cdot \beta + D_3 = D_3^{out}$

$D_3^{out} = \frac{D_3}{1 + A\beta} \Rightarrow \frac{D_3^{out}}{D_3} = \frac{1}{1 - \text{loop gain}}$



IN-BAND ZERO-POLE DOUBLETS

In-band zero-pole doublets appear any time we need for example to drive a large load capacitance with a multi-compensated OTA stage, or when we use a FEED-FORWARD compensation technique:



$$T(s) = \frac{A_2}{1+s\tau_2} + A_0 = \frac{1 + s\left(\frac{A_0\tau_2}{A_2+A_0}\right)}{1+s\tau_2} \quad (A_2+A_0)$$

high-pass block with broad band and low gain (GBWP trade-off)

so we get a zero @ $\tau_z = \frac{A_0}{A_2+A_0} \tau_2 < \tau_2$ (so $f_z > f_p$).

These doublets increase ϕ_m but worsen the transient response of the amplifier.

Let's consider

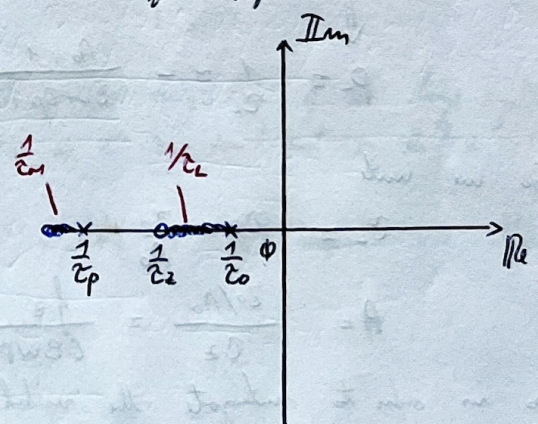
$$A(s) = \frac{A_0(1+s\tau_z)}{(1+s\tau_0)(1+s\tau_p)} \quad (\text{T.F. of a multi-compensated OTA})$$

and $\tau_p < \tau_z < \tau_0$. The closed loop T.F. in a buffer configuration will be

$$H(s) = \frac{(1+s\tau_z)}{(1+s\tau_L)(1+s\tau_H)}$$

Let's consider the response to a step function E

$$\begin{aligned} \text{Var}(s) &= \frac{E}{s} H(s) \\ &\downarrow \\ &= \frac{E}{s} \frac{1+s\tau_z}{(1+s\tau_L)(1+s\tau_H)} \\ &\downarrow \\ &= \frac{E}{s} \left[\frac{A}{1+s\tau_L} + \frac{B}{1+s\tau_H} \right] \end{aligned}$$



where

$$A = \lim_{s \rightarrow -\frac{1}{\tau_L}} H(s) (1+s\tau_L) = \frac{1+s\tau_z \left(-\frac{1}{\tau_L}\right)}{1+s\tau_H \left(-\frac{1}{\tau_L}\right)} = \frac{\tau_L - \tau_z}{\tau_L - \tau_H}$$

$$B = \lim_{s \rightarrow -\frac{1}{\tau_H}} H(s) (1+s\tau_H) = \frac{1+s\tau_z \left(-\frac{1}{\tau_H}\right)}{1+s\tau_L \left(-\frac{1}{\tau_H}\right)} = \frac{\tau_H - \tau_z}{\tau_H - \tau_L}$$

44

We get that

$$H(s) = \frac{A}{s - \tau_L} + \frac{B}{s - \tau_H}$$

so, by integration, we get:

$$\text{var}(t) = E \left[A(1 - e^{-\frac{t}{\tau_L}}) + B(1 - e^{-\frac{t}{\tau_H}}) \right] = \overset{A+B}{E} \left[1 - Ae^{-\frac{t}{\tau_L}} - Be^{-\frac{t}{\tau_H}} \right]$$

The first exponential vanishes first, so after a quick transient we get

$$\text{var}(t) \approx E \left[1 - Ae^{-\frac{t}{\tau_L}} \right] = E \left[1 - \frac{\tau_L - \tau_H}{\tau_L - \tau_H} e^{-\frac{t}{\tau_L}} \right] =$$

We need an estimate of $\tau_L - \tau_H$:

$$\zeta_{loop}(s) = -A_0 \frac{1 + \tau_0 \tau_z}{(1 + \tau_0 s)(1 + \tau_p s)}$$

$$-\zeta_{loop}(s) + 1 = 0$$

↓

$$-A_0 - A_0 \tau_0 \tau_z \cdot s = 1 + \tau_0 s + \tau_p s + \tau_0 \tau_p s^2$$

↓

$$\tau_0 \tau_p s^2 + s(\tau_0 + \tau_p + A_0 \tau_0 \tau_z) + A_0 + 1 = 0$$

↓

$$p_L \approx -\frac{1}{\tau_L} \approx -\frac{A_0 + 1}{\tau_0 + \tau_p + A_0 \tau_0 \tau_z} \approx -\frac{1}{\tau_z + \frac{\tau_0}{A_0}} \quad (\text{big } \tau_0 \gg \tau_p)$$

So we write

$$\tau_L = \tau_z + \frac{\tau_0}{A_0}$$

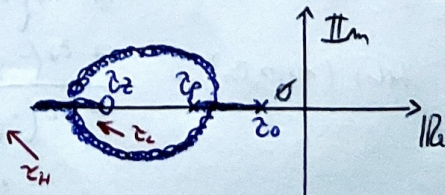
$$A = \frac{\tau_0 / A_0}{\tau_z} = \frac{1}{\text{GBWP}}$$

so in order to mitigate the residual gap that has to be equalized with the slow time constant, we must shift τ_z to LF, but this must also slow τ_L down!

For the case of feed-forward compensation, $\tau_0 > \tau_p > \tau_z$ so the net loop is stiff and we expect

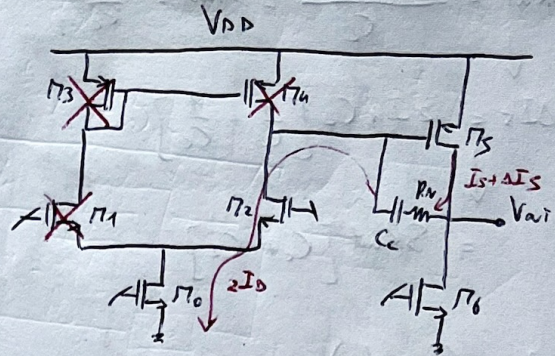
$$\tau_L < \tau_z$$

so the response has an overshoot that is then recovered with the slow time constant τ_L



The slew-rate performance is related to the finite current available to charge capacitors.

If we wish a large signal swing for example V_{O2} , it fully mobilizes the differential output, thus switching $M_1-M_3-M_4$ off and letting I_{D0} flow into C_c .
As a result, $V_{O1} \downarrow$ and M_5 comes



$$I_S + \frac{\Delta I_S}{2I_D}$$

The initial response is a linear ramp with $SR = \frac{I_D}{C_c}$. If we come to have the OTA in a buffer configuration, then we recover the expected behavior when

$$\begin{cases} SR = \frac{\Delta V}{\tau} \\ \Delta V = V_{(oo)} - SR \cdot t_{SLEW} \end{cases}$$

$$\begin{cases} \Delta V = SR \cdot \tau \\ t_{SLEW} = \frac{V_{(oo)} - \tau}{SR} \end{cases}$$

$$\Rightarrow v_{out}(t) = \begin{cases} SR \cdot t & (t \leq t_{SLEW}) \\ V_{(oo)} - \Delta V \cdot e^{-\frac{t-t_{SLEW}}{\tau}} & (t \geq t_{SLEW}) \end{cases}$$

This discussion is valid under the assumption that the ring time of the large signal signal is $< \frac{\tau_L}{GBW} = \frac{1}{GBW(PN)}$ of the OTA, so that v_{out} doesn't react to the large signal at first.

If we consider an harmonic signal input, whose amplitude is the maximum one allowed by the output dynamics, then

$$2\pi A_{max} f_{max} = \left. \frac{dv_{out}}{dt} \right|_{max} \leq SR$$

$$f_{max} \leq \frac{SR}{2\pi A_{max}} \stackrel{D}{=} \text{POWER BANDWIDTH}$$

Finally, we can point out that

$$SR = \frac{I_D}{C_c} = \frac{V_{O1,1} \cdot g_{m1}}{C_c} = 2\pi V_{O1,1} \cdot GBWP$$

Once GBWP is set, we need to raise $V_{O1,1}$ in order to raise SR, but this ~~is~~ beyond E_{in}^2

→ EXTERNAL SR

Let's consider now a load capacitance. We must distinguish now two cases

① large V_p (positive)

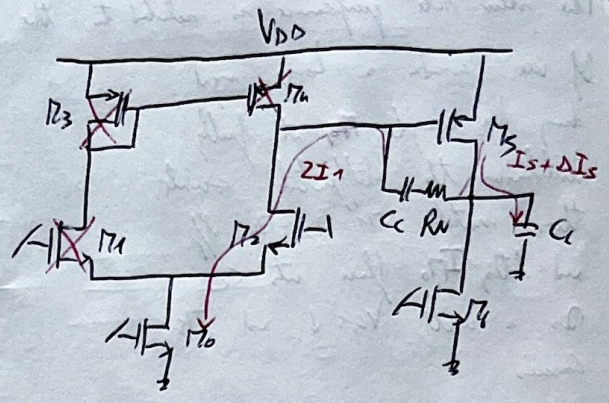
$M_1-M_3-M_4$ are off and $2I_{D1}$ flows into C_c . Under the assumption that V_A reaches a steady state, the ~~signal~~ ramp rate across C_c must be equal to the

④⑥ one across C_L , being it equal to $S_{\text{int}} = \frac{2I_1}{C_L}$ imposed by the first stage, so it must be

$$S_{\text{out}}^+ = \frac{\Delta I_S - 2I_1}{C_L} = S_{\text{int}} = \frac{2I_1}{C_L}$$

$$\Delta I_S = 2I_1 C_L \left[\frac{1}{C_L} + \frac{1}{C_C} \right]$$

$$= S_{\text{int}} [C_C + C_L]$$



② V_p large (negative)

M_2 off, $2I_1$ flows into $M_1 - M_3 - M_4$, M_5 's current decreases

$$S_{\text{out}}^- = \frac{\Delta I_S - 2I_1}{C_L} = S_{\text{int}} = \frac{2I_1}{C_L} \quad (\text{if } S_{\text{int}} \text{ is more limiting than the output load})$$

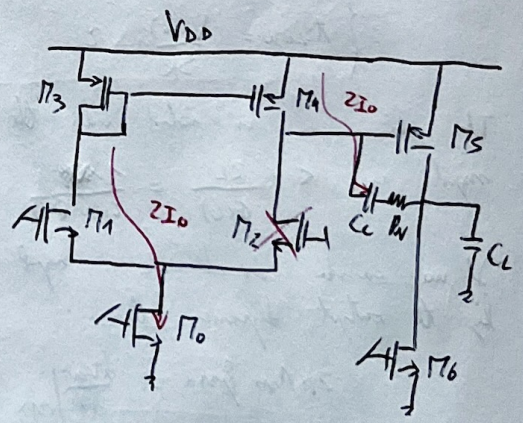
$$\Delta I_S = 2I_1 C_L \left[\frac{1}{C_L} + \frac{1}{C_C} \right] = S_{\text{int}} (C_C + C_L)$$

But if $\Delta I_S > I_S$, then M_5 turns off and M_6 has to sink both C_C and C_L . In this case, at steady state the ramp rate across C_C matches the one across C_L , so

$$\frac{I_6 - I_4}{C_L} = \frac{I_6}{C_C} = S_{\text{out}}^-$$

$$\begin{cases} I_4 = I_6 \cdot \frac{C_C}{C_L + C_C} < 2I_0 \\ S_{\text{out}}^- = \frac{I_6}{C_L + C_C} \end{cases}$$

M_6 is driver!



In this case

$$S_{\text{out}}^- = \min \left\{ S_{\text{int}}; \frac{I_6}{C_L + C_C} \right\}$$

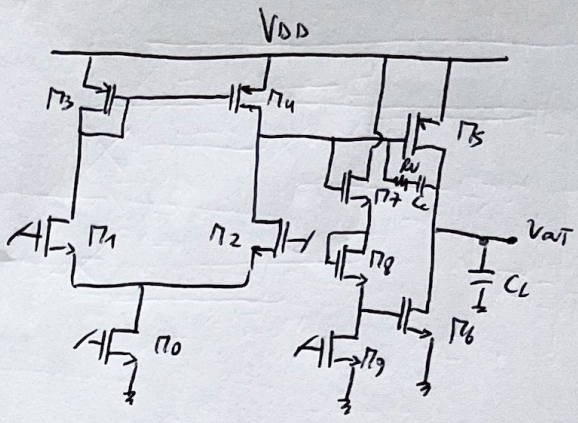
Have to have that S_{int} is the limiting one on both polarities?

→ increase I_6

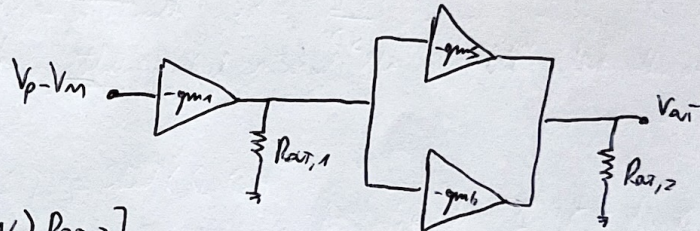
→ use a class A-B, raise M_6 current only when needed!

$M_2 - M_3$ out or settings shifts to properly set M_6 's bias. We use them in order to provide a shift of $0.5V + 0.8V$.

In this way M_6 reacts in order to increase its current only when needed, so we do not need static power dissipation!



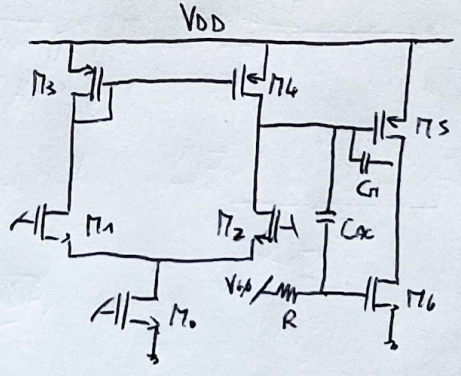
NOTE: the gain changes



$$G_d = \underbrace{[g_{m1} \cdot R_{out,1}]}_{G_1} \cdot \underbrace{[(g_{m5} + g_{m6}) R_{out,2}]}_{G_2}$$

A possible alternative is the AC coupling:

the capacitance C_{ac} makes it possible to properly bias the M_6 transistor and also to let it come into action when needed.



FILTERS

→ Filter synthesis procedure:

- ① FILTER MASK
- ② FILTER TRANSFER FUNCTION
- ③ NETWORK IMPLEMENTATION (ideal)
- ④ NON-IDEALITIES OPTIMIZATION

→ Filters are divided (classified) into

LOW PASS; HIGH PASS; BAND PASS; STOP-BAND; EQUALIZERS (all pass)

The ideal filter should have

- 1) constant in band amplitude
- 2) linear phase shift proportional to ω

DEF: Let's consider a signal fully in the pass-band of a filter. It is affected by a finite propagation delay τ , so

$$x(t) \rightarrow \boxed{\text{FILTER}} \rightarrow y(t) = Ax(t - \tau)$$

τ must affect any harmonic component, so each harmonic suffers from a shift

$$\Delta\phi = -\frac{2\pi}{T} \cdot \tau = -\omega\tau$$

These conditions cannot be fulfilled by a real filter, since

DEF: Let's consider an ideal LP on a bilateral frequency axis, so a rectangle. Its inverse Fourier transform is a

$$h(t) = \frac{1}{2\pi} \int_{-\omega_c}^{\omega_c} H_0 e^{j\omega t} d\omega = \frac{H_0}{2\pi j t} \left[\frac{e^{j\omega t}}{j} \right]_{-\omega_c}^{\omega_c} = \frac{H_0}{\pi t} \sin(\omega_c t)$$

$$= \frac{\omega_c H_0}{\pi} \frac{\sin(\omega_c t)}{\omega_c t} = h_0 \cdot \text{sinc}(\omega_c t)$$

$h(t) \neq 0$ for $t < 0 \Rightarrow$ ANTI-CAUSAL, NOT FEASIBLE!

→ Important definitions for the FILTER MASK are

- A_{BP} ATTENUATION IN BAND-PASS (maximum)
- A_{SB} ATTENUATION IN STOP-BAND (minimum)
- ω_{BP} CUT-OFF FREQUENCY
- ω_{SB} LOWER LIMIT OF THE STOP-BAND

$\omega_0 = \sqrt{\omega_{BP}^+ \omega_{BP}^-}$ CENTRAL FREQUENCY OF A BAND-PASS FILTER

$Q = \frac{\omega_0}{\Delta\omega}$ Q FACTOR OF A BAND-PASS FILTER

(49)

$K = \frac{W_{BP}}{W_{SB}} < 1$ SELECTIVITY INDEX (Lower = worse)

$E_{BP} = \sqrt{10^{\frac{A_{BP}}{10}} - 1}$ MAXIMUM ATTENUATION INDEX IN BAND-PASS

$E_{SB} = \sqrt{10^{\frac{A_{SB}}{10}} - 1}$ MINIMUM ATTENUATION INDEX IN STOP-BAND

$K_E = \frac{E_{BP}}{E_{SB}} < 1$ DISCRIMINATION INDEX (Lower = better)

$\tau_{GD} = - \frac{d\phi}{d\omega}$ GROUP DELAY (ideally constant)

→ ALL POLES FUNCTIONS

• BUTTERWORTH

They require maximum flatness and stop-band flatness

$\left. \frac{d^n |H(j\omega)|^2}{d\omega^n} \right|_{\omega=0} = 0$ for $n=1, 2, \dots, 2m-1$

$\left. \frac{d^n |H(j\omega)|^2}{d\omega^n} \right|_{\omega=\infty} = 0$ for $n=1, 2, \dots, 2m-1$

For a response $\omega_0 = 1$ rad/s angular frequency, these conditions are fulfilled by

$|H(j\omega)|^2 = \frac{1}{1 + \omega^{2m}}$

All poles are placed on a circle of radius $\omega_0 = 1$ rad/s on the complex plane, which

$Q = \frac{1}{2\xi} = \frac{|p|}{2|Re\{p\}} = \frac{1}{2}$ if REAL

Poles are equally spaced one from the next on the circle by $\frac{\pi}{m}$ and the ones with higher Q factor are $\frac{\pi}{2m}$ from the $\pm j\omega$ axis.

In order to achieve a lower A_{BP} than 3dB we may choose a different ω_0 , then having

$|H(j\omega)|^2 = \frac{1}{1 + \left(\frac{\omega}{\omega_0}\right)^{2m}}$

Let's fulfill the specs:

$\frac{1}{1 + \left(\frac{\omega_{BP}}{\omega_0}\right)^{2m}} \geq \frac{1}{1 + E_{BP}^2} \Rightarrow \omega_0 \geq \frac{\omega_{BP}}{E_{BP}^{1/m}}$

$\frac{1}{1 + \left(\frac{\omega_{SB}}{\omega_0}\right)^{2m}} \leq \frac{1}{1 + E_{SB}^2} \Rightarrow \omega_0 \leq \frac{\omega_{SB}}{E_{SB}^{1/m}}$

DEF: $\rho_0 \geq \frac{\rho_{BP}}{\epsilon_{BP}^{1/m}} \wedge \rho_0 \leq \frac{\rho_{SB}}{\epsilon_{SB}^{1/m}}$, by ~~the~~ multiply the
 just times from $\epsilon_{SB}^{1/m}$ we get

$$\frac{\rho_{BP}}{\epsilon_{BP}^{1/m}} \cdot \epsilon_{SB}^{1/m} \leq \rho_0 \cdot \epsilon_{SB}^{1/m} \leq \rho_{SB}$$

$$\downarrow$$

$$\rho_{BP} \cdot \epsilon^{-1/m} \leq \rho_{SB}$$

$$\downarrow$$

$$\epsilon^{-1/m} \leq \epsilon^{-1}$$

$$\downarrow$$

$$\frac{1}{m} \ln(\epsilon^{-1}) \leq \ln(\epsilon^{-1})$$



$$m \geq \frac{\ln(\epsilon^{-1})}{\ln(\epsilon^{-1})}$$

And $\frac{\rho_{BP}}{\epsilon_{BP}^{1/m}} \leq \rho_0 \leq \frac{\rho_{SB}}{\epsilon_{SB}^{1/m}}$

• BESSEL

They require the maximum phase linearity, so

$$\left. \frac{d^k \rho_0}{d\omega^k} \right|_{\omega=0} = 0 \quad \text{for } k=1, 2, \dots, M$$

They have poles placed outside the ρ_0 characteristic frequency circle, in order to reach a better group delay, this also a lower selectivity.

In order to select this, we use tables referred to $\rho_0 = 1$ rad/s and we try to see if a specific order with a specific ρ_0 fulfill the specs.

• CHEBYSHEV - I

They require the maximum selectivity, but with an in-band ripple.

$$\max \left\{ 1 - |H(j\omega)|^2 \right\}_{\omega \leq 1} \leq \epsilon_{BP}^2 \quad \text{maximum ripple requirement}$$

$$\left. \frac{d^k |H(j\omega)|^2}{d\omega^k} \right|_{\omega \rightarrow \infty} = 0 \quad \text{for } k=1, 2, \dots, 2n-1 \quad \text{stop-band flatness}$$

The DC value must be set to 1 for ~~the~~ odd order T.F. and to

$\frac{1}{\sqrt{1+\epsilon_{BP}^2}}$ for ~~the~~ even order T.F., in order to have $|H(j\omega)| \leq 1$ in band.

USEFUL CHEBYSHEV FORMULAS

$$m \geq \frac{\operatorname{arCh}(M\epsilon^{-1})}{\operatorname{arCh}(M^{-1})}$$

$$\Gamma = \left(\frac{1 + \sqrt{1 + \epsilon_{sp}^2}}{\epsilon_{sp}} \right)^{1/m}$$

Consider only ~~pass~~ the one with $\Re\{s\} < 0$

$$s_k = -\sin \left[(2k-1) \frac{\pi}{2m} \right] \frac{\Gamma^2 - 1}{2\Gamma} + j \cos \left[(2k-1) \frac{\pi}{2m} \right] \frac{\Gamma^2 + 1}{2\Gamma}$$

(for $k = 1, \dots, 2m$)

$m = \#$ of half oscillations in band-pass

→ POLES AND ZEROS

• CHEBYSHEV II

They have imaginary conjugated zeros in stop band in order to be a ripple free and wide HF distributions.

$$\max_{|w| \leq 1} \{1 - |H(jw)|^2\} \geq \epsilon_{SB}^2 \quad \text{min attraction in stop band } (E = \frac{1}{\epsilon_{SB}})$$

$$\frac{d^n |H(jw)|^2}{dw^n} \Big|_{w=0} = 0 \quad \text{for } n = 1, 2, \dots, 2m-1$$

• COVER

They are a constant ripple both in stop-band and in band-pass. Once u and the in-band ripple is set, the stop-band ripple is determined

• GENERALIZED ELLIPTICAL

They give an additional degree of freedom, so we can independently set u and both ripples, but we get a band edge.

→ ADDITIONAL DEMONSTRATION

It can be shown that for a Chebyshev-type-I filter it is

$$|H(j\omega)|^2 = \frac{1}{1 + \epsilon^2 C_n^2(\omega)}$$

$\left\{ \begin{array}{l} \epsilon \cong \text{ripple factor} \\ C_n = \text{Chebyshev polynomial} \end{array} \right.$

where

$$C_n(\omega) = \begin{cases} \cos \left[n \operatorname{arCos} \left(\frac{\omega}{\omega_{sp}} \right) \right] & \left(\frac{\omega}{\omega_{sp}} \leq 1 \right) \\ \operatorname{Ch} \left[n \operatorname{arCh} \left(\frac{\omega}{\omega_{sp}} \right) \right] & \left(\frac{\omega}{\omega_{sp}} \geq 1 \right) \end{cases} \quad (=1 \text{ for } \omega = \omega_{sp})$$

so

$$|H(j\omega_{sp})|^2 = \frac{1}{1 + \epsilon^2} \implies \epsilon = \sqrt{10 \frac{\Delta_{sp}}{\Delta_0} - 1} = \epsilon_{sp}$$

In order to derive $m = \frac{\operatorname{arCh}(M\epsilon^{-1})}{\operatorname{arCh}(M^{-1})}$ with $|H(j\omega_{sp})|^2 = \frac{1}{1 + \epsilon_{sp}^2 C_n^2(\omega_{sp})}$...

Filter synthesis is performed in a domain Ω referred to a low pass filter with $\Omega_{cp} = 1$ rad/s. Therefore we need suitable transformations.

→ LOW PASS

$\Omega = \frac{\omega}{\omega_{cp}}$ in order to get $\Omega_{cp} = 1$ rad/s

Starting from the normalized Butterworth polynomials referred to $\Omega_0 = 1$ rad/s we need first to set

$\hat{S} = \Omega_0 p$

p normalized low-pass with $\Omega_0 = 1$ rad/s
 \hat{S} " " " " " $\Omega_0 \neq 1$ rad/s

Then, we get

$S = \omega_{cp} \hat{S}$ S low-pass

→ HIGH PASS

$\Omega = \frac{\omega_{cp}}{\omega}$ in order to have $\Omega_{cp} = 1$ rad/s and to flip the axis

$\hat{S} = \frac{\omega_{cp}}{S}$

→ BAND-PASS

DEM: let's consider this transformation

$\hat{S} = \hat{p} + \frac{1}{\hat{p}}$

$\begin{cases} \hat{S} = \Lambda + j\Omega \\ \hat{p} = d + j\omega \end{cases}$

$\hat{p}^2 - \hat{S}\hat{p} + 1 = 0$

$\hat{p} = \frac{\hat{S} \pm \sqrt{\hat{S}^2 - 4}}{2} \Rightarrow d + j\omega = \frac{\Lambda + j\Omega \pm \sqrt{\Lambda^2 - \Omega^2 + 2j\Omega\Lambda - 4}}{2}$

Now, the $\text{Im}[\hat{S}]$ axis of \hat{S} ($\Lambda = 0$) is mapped into

$d + j\omega = \frac{j\Omega \pm j\sqrt{\Omega^2 - 4}}{2} = j \frac{\Omega}{2} \pm j \sqrt{1 - \left(\frac{\Omega}{2}\right)^2}$



as $d = 0$ and the $\text{Im}[\hat{S}]$ axis of \hat{S} is mapped into two parts on the $\text{Im}[\hat{p}]$ axis of \hat{p} .

(53)

In particular,

$$\Omega = 0 \Rightarrow \omega_{1,2} = \pm 1$$

$$\Omega = 1 \Rightarrow \omega_{1,2} = +1.62; -0.62$$

Being $\Omega = 1$ the cut-off of the low pass filter, this is mapped into s_{cut} of s_{cut} of the low pass, it sets the BW to 1. So, to have a BW smaller than 1, we shall apply the transformation to a LP with $s_{\text{cut}} = \frac{1}{Q}$. Finally, to shift the cut frequency, we shall expect \hat{p} , this being $s = \hat{p} \cdot \omega_0$

$$\Downarrow$$

$$\hat{s} = Q\tilde{s} = Q\left[\hat{p} + \frac{1}{\hat{p}}\right] = Q\left[\frac{s}{\omega_0} + \frac{\omega_0}{s}\right] = Q\left[\frac{s^2 + \omega_0^2}{s\omega_0}\right]$$

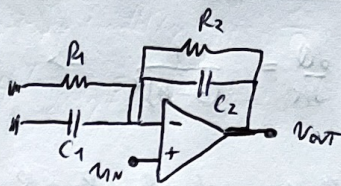
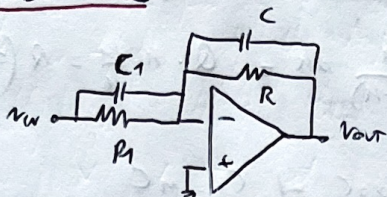
which in terms of real frequency they correspond to

$$\Omega = Q\left(\frac{\omega^2 - \omega_0^2}{\omega\omega_0}\right)$$

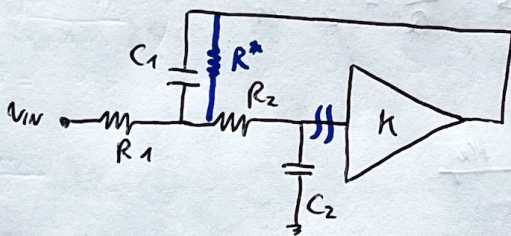
ACTIVE CELLS

We can always split the T.F. in the product of 1st and 2nd order terms, so we can consider each of them as a block and then cascade all of them (carefully decoupled) in order to reach the final T.F. We can use OPAMPS!

→ 1ST ORDER CELL



→ 2ND ORDER CELL (Sallen - Key)



DEF :

$$G_{loop}(s) = +K \cdot \frac{R_1}{R_1 + R^*}$$

$$a_1 = R^* C_1$$

$$b_1 = C_2 (R_2 + R_1 // R^*) + C_1 \cdot (R^* // R_1)$$

$$b_2 = C_1 C_2 (R^* // R_1) R_2$$

$$G_{loop}(s) = +K \cdot \frac{R_1}{R_1 + R^*} \cdot \frac{1 + s C_1 R^*}{1 + [C_2 (R_2 + R_1 // R^*) + C_1 (R^* // R_1)] s + s^2 C_1 C_2 R_2 (R^* // R_1)}$$

$R^* \rightarrow \infty$

$$= +K \frac{s C_1 R_1}{s^2 C_1 C_2 R_1 R_2 + s [C_1 R_1 + C_2 (R_1 + R_2)] + 1}$$

By setting $G_{loop}(s) - 1 = 0$ we get the char. equation:

$$s^2 (R_1 R_2 C_1 C_2) + s [R_1 (1-K) C_1 + (R_1 + R_2) C_2] + 1 = 0$$

$$\omega_0 = \frac{1}{\sqrt{C_1 C_2 R_1 R_2}} \quad \text{and} \quad Q = \frac{\sqrt{R_1 R_2 C_1 C_2}}{C_1 R_1 + C_2 (R_1 + R_2) - K C_1 R_1}$$

$$\downarrow$$

$$= \frac{(R_2 C_1) \sqrt{R_1 R_2 C_1 C_2}}{(1-K) + \frac{C_2 (R_1 + R_2)}{R_1 C_1}} = \frac{\sqrt{R_2 C_2}}{\sqrt{R_1 C_1}} \cdot \frac{1}{(1-K) + \frac{C_2 (R_1 + R_2)}{R_1 C_1}}$$

SS

Q depends on ratios \Rightarrow affected by on chip variations, mitigated by using
 no. depends on absolute values \Rightarrow affected by chip-to-chip variations

① $k \neq 1$ and $R_1 = R_2 = R$ and $C_1 = C_2 = C$

$$w_0 = \frac{1}{RC} \quad Q = \frac{1}{3-k} \Rightarrow k = 3 - \frac{1}{Q} \text{ dependent on } Q!$$

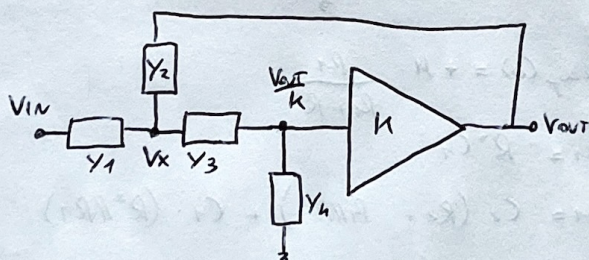
$$\frac{dQ}{dk} = \frac{1}{(3-k)^2} \Rightarrow \frac{dQ}{Q} = \frac{1}{(3-k)} dk = Q \cdot k \frac{dk}{k} = Q \left(\frac{3-k}{Q} \right) \frac{dk}{k}$$

② $k=1$ and $R_1 = R_2 = R$ and $C_2 = C$ and $C_1 = mC$

$$w_0 = \frac{1}{\sqrt{m}RC} \quad Q = \frac{\sqrt{m}}{2} \Rightarrow m = 4Q^2$$

$$\frac{dQ}{dm} = \frac{1}{4\sqrt{m}} \Rightarrow \frac{dQ}{Q} = \frac{1}{2} \frac{dm}{m}$$

\rightarrow HIGH-PASS Sallen-Key



DEF:

$$\begin{cases} (V_{in} - V_x) Y_1 + (V_{out} - V_x) Y_2 = (V_x - \frac{V_{out}}{k}) Y_3 \\ V_x = \frac{V_{out}}{k} \left(1 + \frac{Y_4}{Y_3} \right) \end{cases}$$

\downarrow

$$V_{out} \left(Y_2 + \frac{Y_3}{k} \right) - V_x \left[Y_1 + Y_2 + Y_3 \right] = -V_{in} Y_1$$

\downarrow

$$V_{out} \left(Y_2 + \frac{Y_3}{k} \right) - \frac{V_{out}}{k} \left[1 + \frac{Y_4}{Y_3} \right] \left[Y_1 + Y_2 + Y_3 \right] = -V_{in} Y_1$$

\downarrow

$$V_{out} \left[Y_2 + \frac{Y_3}{k} - \frac{Y_1}{k} - \frac{Y_2}{k} - \frac{Y_3}{k} - \frac{Y_1 Y_4}{k Y_3} - \frac{Y_2 Y_4}{k Y_3} - \frac{Y_4}{k} \right] = -V_{in} Y_1$$

\downarrow

$$\frac{V_{out}}{V_{in}} = \frac{Y_1}{\frac{1}{k} \left[Y_1 + Y_2 + Y_3 + \frac{Y_1 Y_4}{Y_3} + \frac{Y_2 Y_4}{Y_3} + Y_4 - k Y_2 \right]} = \frac{k Y_1 Y_3}{Y_4 \left[Y_1 + Y_2 + Y_3 + \frac{Y_1 Y_4}{Y_3} + \frac{Y_2 Y_4}{Y_3} + Y_4 - k Y_2 \right]}$$

→ UNIVERSAL CELL

Also called Kevin - Blawie - Mueser (KBM)

DEF: let's consider a high-pass T.F.

$$\frac{V_{HP}(s)}{V_{IN}} = \frac{\gamma s^2}{(s^2 + \frac{\omega_0}{Q}s + \omega_0^2)}$$

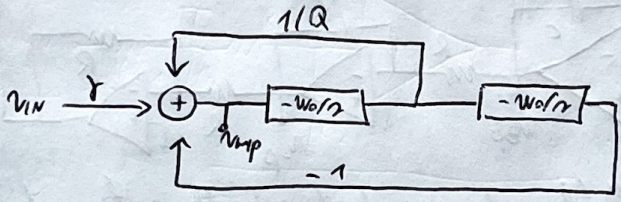
In order to let the integrators appear, let's divide for the highest power of s :

$$V_{HP} \left[1 + \frac{\omega_0}{Qs} + \frac{\omega_0^2}{s^2} \right] = \gamma V_{IN}$$

↓

$$V_{HP} = \gamma V_{IN} - \frac{\omega_0}{s} \frac{V_{HP}}{Q} - \frac{\omega_0^2}{s^2} V_{HP}$$

BLOCK DIAGRAM:

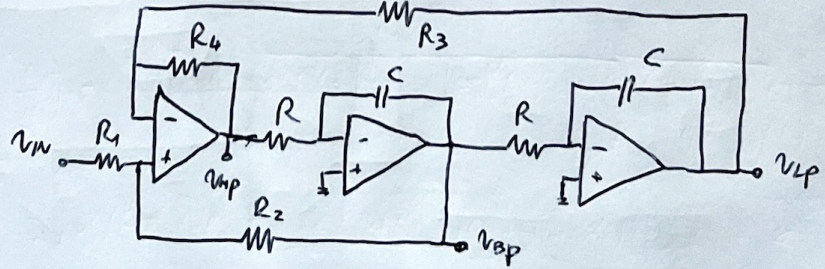


After the first integrator we get a BAND-PASS

$$\frac{V_{BP}(s)}{V_{HP}(s)} = - \frac{\gamma \omega_0}{(s^2 + \frac{\omega_0}{Q}s + \omega_0^2)}$$

and after the second we get a LOW-PASS

$$\frac{V_{LP}(s)}{V_{HP}(s)} = \frac{\gamma \omega_0^2}{(s^2 + \frac{\omega_0}{Q}s + \omega_0^2)}$$



$$\rightarrow V_{HP}(s) = \frac{R_2}{R_1 + R_2} \cdot \frac{R_3 + R_4}{R_3} \cdot V_{IN}(s) + \frac{R_4}{R_1 + R_2} \cdot \frac{R_3 + R_4}{R_3} V_{BP}(s) + \left(- \frac{R_4}{R_3} \right) V_{LP}(s)$$

So we get

$$R_4 = R_3 \quad (\text{and } -1) \quad \gamma = \frac{2R_2}{R_1 + R_2} \quad \text{and} \quad \frac{1}{Q} = \frac{2R_4}{R_1 + R_2}$$

(57)

$\frac{P_2}{P_1} = 2Q - 1$

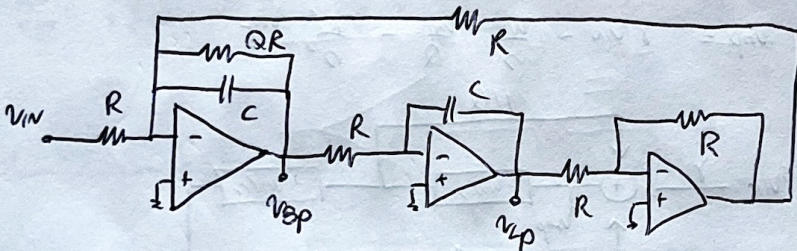
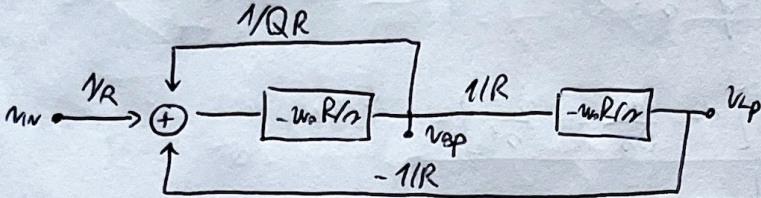
and $\gamma = (2 - \frac{1}{Q})$ dependent of each other!

If we combine the two stages in a initial gain we may get a T.F. with also zeros

ADDER
Tolu

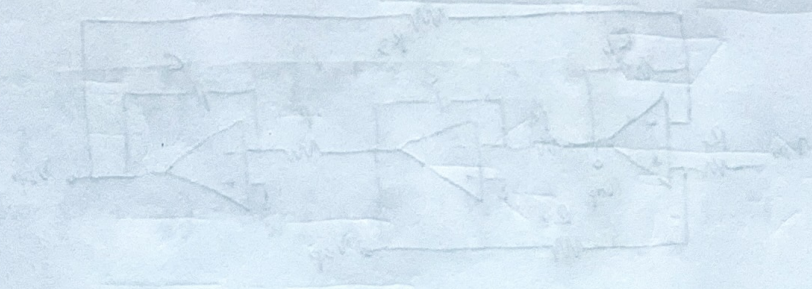
→ TOU-THOMAS AND FLEISCHER-THOMAS

We may run unit signals instead of voltages by using an integrator V.b. To this aim we need to divide by $\frac{1}{R}$ and to multiply integrator T.F. by R.



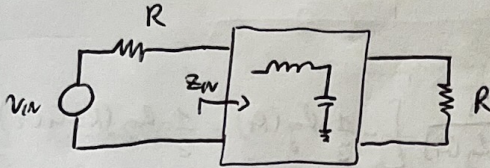
In a fully differential mode we don't need the inverters, it's enough to flip the colors.

By using with v_{in} in all three V.b. we get a Fleischer-Thomson cell with also two zeros.



The idea is to exploit interactions between different components in the same cell in order to compensate non-idealities. In active cells instead, each cell contributes independently to non-ideality.

DEF: These networks use PASSIVE COMPONENTS, like this:



DOUBLY TERMINATED NETWORK

At the maximum of the T.F. $Z_{IN} = R$. At the real frequency ω the power delivered to the load can be written as

$$P_L = \frac{V_{out}^2}{R} = \frac{V_{in}^2}{2R} |T(j\omega, X_0)|^2 \quad X_0 \triangleq \text{value when } \omega = \omega^* \text{ of } L \text{ or } C$$

Let's assume to have a maximum @ $\omega = \omega^*$, then

$$\left. \frac{\partial P_L}{\partial \omega} \right|_{\omega = \omega^*} = \frac{V_{in}^2}{2R} \cdot 2 |T(j\omega^*, X_0)| \cdot \left. \frac{\partial |T(j\omega, X_0)|}{\partial \omega} \right|_{\omega = \omega^*} = 0$$

$$\downarrow \quad \left. \frac{\partial}{\partial \omega} |T(j\omega, X_0)| \right|_{\omega = \omega^*} = 0 = \left. \frac{\partial |T(j\omega^*, X)|}{\partial X} \right|_{X = X_0} \quad \text{big an absolute maximum}$$

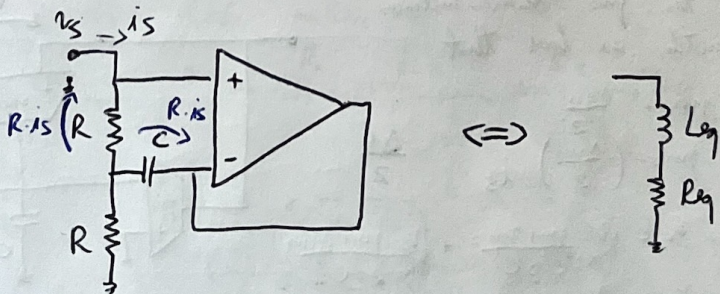
So the sensitivity of the T.F. to variations of L and C is nil at the passband and rises nil around it due to continuity (ORCHARD THEOREM)

How to implement inductors?

→ GYRATORS

They use active elements to generate an inductive impedance

①



$$v_s = R \cdot i_s + (i_s + R \cdot i_s \cdot \omega C) \cdot R$$

$$\downarrow = i_s [2R + \omega CR^2]$$

5) SENSITIVITY ANALYSIS

$$S_X^Y = \frac{\frac{\partial Y}{\partial X}}{\frac{Y}{X}} = \frac{\partial \log(Y)}{\partial \log(X)}$$

$X =$ circuit parameter
 $Y =$ filter parameter
 $\log =$ natural log \ln

EX: $\omega_0 = \frac{1}{\sqrt{R_1 R_2 C_1 C_2}}$

$$S_{R_1}^{\omega_0} = \frac{\partial \log(\omega_0)}{\partial \log(R_1)} = \frac{0}{\partial \log(R_1)} \left[-\frac{1}{2} \log(R_1) - \frac{1}{2} \log(R_2 C_1 C_2) \right] = -\frac{1}{2}$$

same for $S_{R_2}^{\omega_0}$, $S_{C_1}^{\omega_0}$, $S_{C_2}^{\omega_0}$, and for $S_H^{\omega_0} = 0$

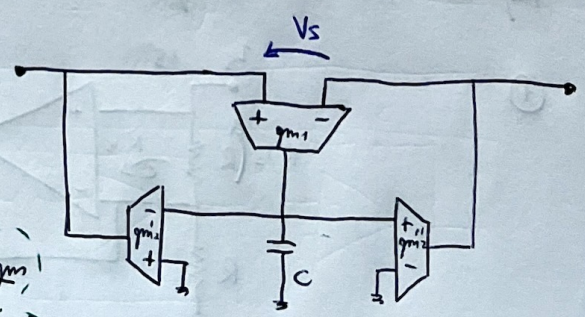
EX: $Q = \frac{1}{[R_1(1-k)C_1 + (R_1+R_2)C_2] \omega_0}$

$$\begin{aligned}
 S_{R_1}^Q &= \frac{0}{\partial \log(R_1)} \left[-\log(\omega_0) - \log(R_1(1-k)(C_1 + (R_1+R_2)C_2)) \right] = \\
 &= \frac{1}{2} - \frac{R_1}{\partial R_1} \left[\log(R_1(1-k)(C_1 + (R_1+R_2)C_2)) \right] = \\
 &= \frac{1}{2} - R_1 \frac{(1-k)(C_1 + C_2)}{R_1(1-k)(C_1 + (R_1+R_2)C_2)} = \\
 &= \frac{R_1[1-k](C_1 + (R_1+R_2)C_2) - 2C_1R_1(1-k) - 2C_2R_1}{2[R_1(1-k)(C_1 + (R_1+R_2)C_2)]} = \\
 &= -\frac{C_1R_1(1-k) + (R_1 - R_2)C_2}{2[R_1(1-k)(C_1 + (R_1+R_2)C_2)]}
 \end{aligned}$$

The others are performed similarly in the same way...

MISMATCH ISSUE IN CYRATORS

If the two g_{m2} transconductances do not match, we have that



$$g_{m2}' = g_{m2} + \frac{\Delta g_{m2}}{2}$$

$$g_{m2}'' = g_{m2} - \left(\frac{\Delta g_{m2}}{2}\right) \text{ is } \frac{\Delta i}{2}$$

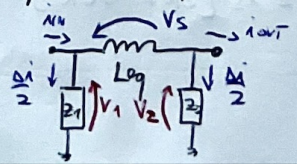
$$i_{in} = g_{m2}' v_c = \frac{v_s g_{m1} g_{m2}'}{2C} + \frac{v_s g_{m1} \Delta g_{m2}}{2C}$$

$$i_{out} = g_{m2}'' v_c = \frac{v_s g_{m1} g_{m2}''}{2C} - \frac{v_s g_{m1} \Delta g_{m2}}{2C}$$

$$i_{in} - i_{out} = \Delta i = \frac{v_s g_{m1}}{2C} \cdot \frac{\Delta g_{m2}}{g_{m2}} g_{m2} = \text{is } \frac{\Delta g_{m2}}{g_{m2}} = \frac{v_s}{\partial \log} \frac{\Delta g_{m2}}{g_{m2}}$$

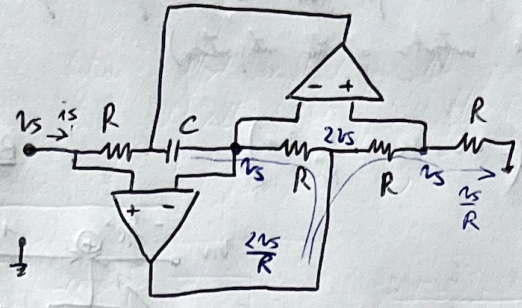
We basically get the return:

$$z_2 = \frac{v_2}{\Delta i / 2} ; z_1 = \frac{v_1}{\Delta i / 2}$$



$$\begin{cases}
 z_1 = \frac{v_s}{2} + v_{cm} \\
 z_2 = -\frac{v_s}{2} + v_{cm}
 \end{cases}$$

② Anomalous Circuit

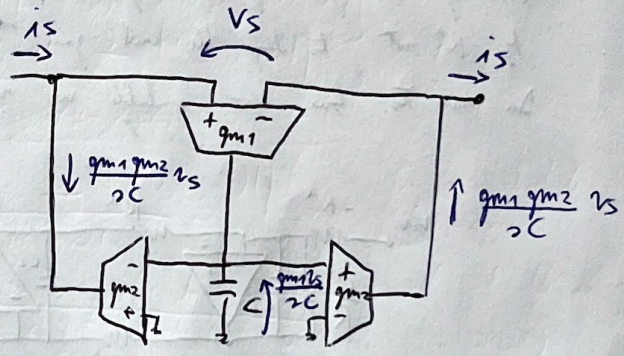
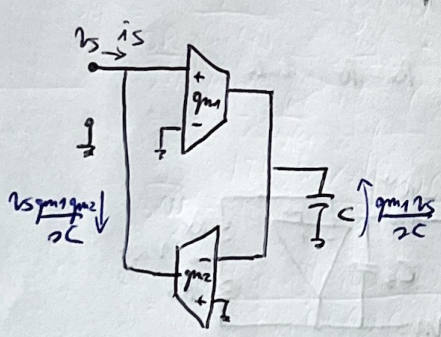


$$\frac{2V_s}{R} \cdot \frac{1}{2C} = i_s \cdot R$$

$$\downarrow$$

$$\frac{2V_s}{i_s} = 2CR^2$$

③ Other solutions

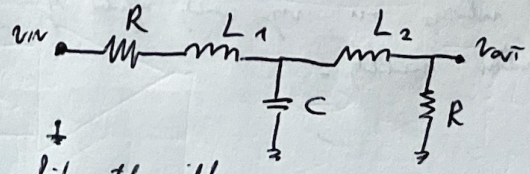


Problems of op-amps are
 - limited GBWP of active elements
 - noise

→ INTEGRATORS (DEF:)

Let's consider this 3rd order ladder network:

$$\begin{cases} \underline{MVL}: v_{in} - v_C = (2L_1 + R) i_1 \\ \underline{MVL}: v_C - v_{out} = R i_2 \\ \underline{MCL}: i_1 - i_2 = 2C v_C \end{cases}$$



find the nulls
 (inductors short, capacitors open)

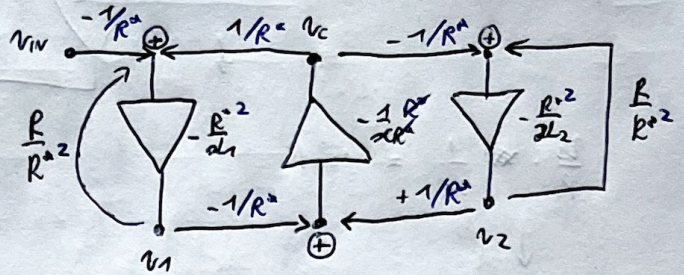
Now write the cuts as auxiliary voltages divided by an auxiliary resistance R'

$$\begin{cases} v_{in} - v_C = \frac{2L_1 + R}{R'} \cdot v_1 \\ v_C - v_{out} = \frac{R L_2}{R'} v_2 \\ \frac{v_1 - v_2}{R'} = 2C v_C \\ v_{out} = \frac{R}{R'} v_2 \end{cases}$$

61) Place limits such equation for the highest power of s in order to let the integrators appear: (write the outputs of the integrators)

$$v_1 = -\frac{R^2}{sL_1} (-v_{in} + v_c + \frac{R}{R^2} v_1)$$

BLOCK DIAGRAM



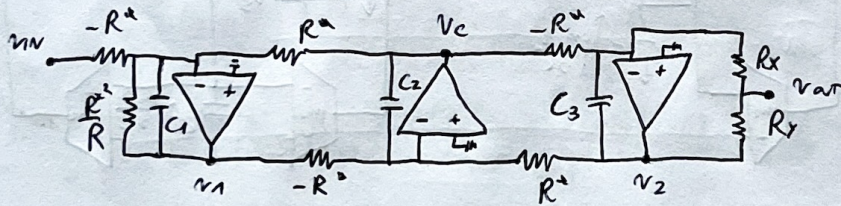
$$v_c = -\frac{1}{sCR^2} (-v_1 + v_2)$$

$$v_2 = -\frac{R^2}{sL_2} (-v_c + v_{out})$$

$$v_{out} = \frac{R}{R^2} v_2$$

We may use initial gains in order to sum signals, so we need to divide each T.F. of the inputs of the integrator by $\frac{1}{R^2}$ and then to multiply each integrator T.F. by R^2 , not to change the overall gain.

In order to have v_{out} available we need an $R_x - R_y$ divider



$$\begin{cases} R_x + R_y = \frac{R^2}{R} \\ v_{out} = \frac{R_x}{R_x + R_y} v_2 = \frac{R}{R^2} v_2 \Rightarrow R_x = R^2 \text{ and } R_y = \frac{R^2}{R} - R^2 \end{cases}$$

Their responses are very much affected by feedback components' variability, but they are affected by resistor variability.

Dynamic Range \geq ratio between maximum and minimum signal levels handled by the circuit.

DEF: Let's consider this circuit

The maximum output signal is $2V_{DD}$, so a signal

with $V_{PEAK} = \frac{2V_{DD}}{2}$, so

$$\langle v_{out}^2 \rangle = \frac{2^2 V_{DD}^2}{8}$$

while the RMS value of the noise due to just the resistor R is

$$k_B T R \cdot \frac{1}{4RC} = \frac{k_B T}{C}$$

and the additional noise due to the amplifier is $F \cdot \frac{k_B T}{C}$, so we get

$$DR = \sqrt{\frac{2^2 V_{DD}^2 / 8}{(1+F) k_B T / C}} = 2V_{DD} \sqrt{\frac{C}{8k_B T (1+F)}}$$

So, to increase the DR (which means to increase the number of bits of an A/D Converter, by $DR = \frac{FSR}{LSB}$) we need to increase both C (more area) and V_{DD}

(more power dissipation)

On each half-cycle the capacitance is charged up to $2V_{DD}$ and then discharged, so it dissipates

$$\frac{2V_{DD} \cdot C \cdot V_{DD}}{T} = \frac{\Delta Q}{\Delta t}$$

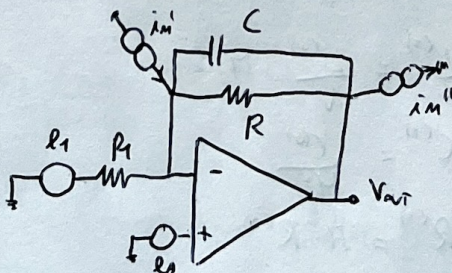
→ NOISE

Let's consider this amplifier

$$\textcircled{1} \text{ } S_{out, R} = R_1 \cdot \frac{R}{R_1} \frac{1}{4RC}$$

↓

$$S_{out, R} = S_{v, R} \left(\frac{R}{R_1} \right)^2 \frac{1}{4RC} = \frac{k_B T}{C} \frac{R}{R_1}$$



② $v_{out,2} \Rightarrow S_{I,R2} \cdot R^2 \cdot \frac{1}{4RC} = \frac{kT}{C}$

③ $v_{out,A} \Rightarrow S_A \cdot \frac{w_M}{4} + S_A \left(1 + \frac{R}{R_1}\right)^2 \frac{w_0}{4}$

we take into account also the finite GBWP of the amplifier, so that we consider the NF transfer to be limited by the $ENBW = \frac{w_M}{4}$

To be more accurate we should consider that the finite GBWP of the amplifier affects the small T.F., so

$$L_{out,A} = L_A \left[1 + \frac{R}{R_1} \frac{1}{1 + sRC} \right] \cdot \frac{1}{\left(1 + \frac{s}{w_M}\right)} = L_A \left(1 + \frac{R}{R_1}\right) \frac{\left(1 + \frac{s}{w_2}\right)}{\left(1 + \frac{s}{w_0}\right) \left(1 + \frac{s}{w_M}\right)}$$

$w_2 \text{ high} = \frac{1}{(R||R_1)}$

If we write it in this way, we can use the tabulated integral

$$\int_0^{+\infty} \left| \frac{1 + \frac{s}{w_2}}{\frac{s^2}{w_0 w_M} + s \frac{(w_0 + w_M)}{w_0 w_M} + 1} \right|^2 df = \frac{w_0 w_M}{(w_0 + w_M)} \frac{1}{4} \left[1 + \frac{w_0 w_M}{w_2^2} \right] = ENBW$$

$w_M \gg w_0$
 $\therefore L_{out,A} = S_A \cdot \left(1 + \frac{R}{R_1}\right)^2 \cdot ENBW \rightarrow S_A \cdot \left(1 + \frac{R}{R_1}\right)^2 \cdot \frac{w_0}{4} \left[1 + \frac{w_0 w_M}{w_2^2} \right]$

which compares to the approximated result. \downarrow
 $= S_A \left(1 + \frac{R}{R_1}\right)^2 \cdot \left[\frac{w_0}{4} + \frac{w_M}{4} \left(\frac{R_1}{R+R_1}\right)^2 \right]$

④ $S_{out} = \overbrace{S_{v,IN} G^2 \cdot \frac{w_0}{4}}^{\text{INPUT}} + \overbrace{kT R \frac{w_0}{4}}^{\text{R}} + \overbrace{S_{v,A} \cdot \left[\eta + (1+G)^2 \right] \frac{w_0}{4}}^{\text{AMPLIFIER}} + \overbrace{kT R_1 G \frac{w_0}{4}}^{\text{R}_1}$
 \downarrow
 $= S_{v,IN}^2 \frac{w_0}{4} \left[1 + \frac{kT R}{S_{v,IN}} \left(\frac{1+G}{G^2}\right) + \frac{S_{v,A}}{S_{v,IN}} \frac{\eta + (1+G)^2}{G^2} \right] = S_{v,IN} \cdot G^2 \cdot \frac{w_0}{4} \left[1 + F \right]$

S_v , in order to minimize F , we need to

- 1) have $G > 1$
- 2) resistor whose noise is $< S_{v,IN}$
- 3) amplifier whose noise is $< S_{v,IN}$
- 4) limit the loaded noise by a suitable GBWP

$$\left[\begin{aligned} G &= \frac{R}{R_1} \\ \eta &= \frac{w_M}{w_0} \end{aligned} \right]$$

63 LADDER NETWORKS SUMMARY

→ HL PROCEDURE

- ① Use HL in order to write relations among STATE VARIABLES (i_L, v_C)
- ② Divide each member by the highest power of s
- ③ Isolate integrators
- ④ Write each unit on an auxiliary network divided by R^a

→ FLOWCHART PROCEDURE

- ① Write all networks and units IN SIDE COMPONENTS
- ② Draw integrators
- ③ Use HL to give the inputs of integrators (i_C, v_L)
- ④ Divide the ladder going into v.c. by R^a and multiply the integrators T.F. by R^a
- ⑤ Multiply units on the down side by R^a , divide ladder going up by R^a and multiply ladder going down by R^a
- ⑥ Multiply by (-1) inputs of integrators and integrators T.F.

→ DENORMALIZATION

Remember N_0 a $\frac{1}{\sqrt{LC}}$ and Q a $N_0 RC$, as to shift us to a value N times higher \sqrt{LC} we need to set

$$\begin{cases} L^{(1)} = \frac{L^{(0)}}{N} \\ C^{(1)} = \frac{C^{(0)}}{N} \end{cases}$$

This does change Q .

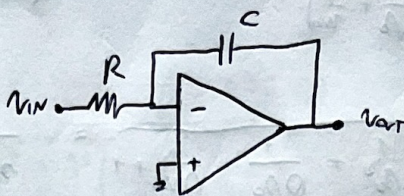
Moreover, to further decrease C without changing us and Q we need to multiply L and R by M :

$$\begin{cases} L^{(1)} = \frac{L^{(0)}}{N} \cdot M \\ C^{(1)} = \frac{C^{(0)}}{MN} \\ R^{(1)} = M \cdot R^{(0)} \end{cases}$$

OP-AMP Non IDEALITIES

Let's consider an integrator:

We have that the forward gain is the opAMP, while the $\beta(s)$ is



$$\beta(s) = \frac{R}{R + \frac{1}{sC}} = \frac{sCR}{1 + sCR}$$

$$\begin{cases} CR = \tau_0 \\ \frac{1}{CR} = \omega_0 \end{cases}$$

$$\frac{1}{\beta} = \frac{1 + \frac{\tau_0}{s}}{\tau_0} = 1 + \frac{\omega_0}{s}$$

The loop gain is equal to $G_{loop}(s) = -\frac{A_0}{(1+s\tau_0)} \frac{s\tau_0}{(1+s\tau_0)}$, it has a low frequency pole due to the opAMP, a zero in DC and a pole at ω_0 , so the loop gain @ middle frequency is

$$|G_{loop}(MF)| = A_0 \frac{\tau_0}{\tau} = \frac{GBWP}{f_0} \quad (\text{we need } GBWP \geq 100 f_0 \text{ to have a null loop gain})$$

and the closed loop transfer function has a pole at

$$\omega_L = \frac{\omega_0}{A_0} = \frac{1}{A_0 RC} \quad (\text{MILLER POLE})$$

and another pole at

$$\omega_H = GBWP$$

So, in the end,

$$H_{int}(s) = \frac{-A_0}{(1 + \frac{s}{\omega_L})(1 + \frac{s}{\omega_H})}$$

$\begin{cases} \text{DC gain} \\ \text{LOW FREQUENCY POLE} \\ \text{HIGH FREQUENCY POLE} \end{cases}$

DC GAIN AND LF POLE EFFECT

$$T(s) = \frac{\omega_0^2 \cdot \frac{1}{s^2}}{\left(s^2 + \frac{\omega_0}{Q} s + \frac{1}{\tau^2}\right)} = \frac{\left(\frac{-\omega_0}{\tau}\right)^2}{\left(\frac{-\omega_0}{\tau}\right)^2 - \frac{1}{Q} \left(\frac{-\omega_0}{\tau}\right) + 1} = \frac{H_{int}^2(s)}{H_{int}^2(s) - \frac{1}{Q} H_{int}(s) + 1}$$

Let's consider now $H_{int}(s) = \frac{-A_0}{1 + \frac{s}{\omega_L}}$ at LOW FREQUENCY, so we can write it as

$$H_{int}(s) = -\frac{1}{\left(\frac{1}{A_0} + \frac{s}{\omega_0}\right)}$$

and replace it into $T(s)$

(66)

$$\begin{aligned}
 T(s) &= \frac{1}{\left(\frac{1}{A_0} + \frac{2}{\omega_0}\right)^2} = \frac{1}{1 + \frac{1}{Q} \left(\frac{1}{A_0} + \frac{2}{\omega_0}\right) + \left(\frac{1}{A_0} + \frac{2}{\omega_0}\right)^2} \\
 &= \frac{1}{\left(\frac{1}{A_0} + \frac{2}{\omega_0}\right)^2 + \frac{1}{Q} \left(\frac{1}{A_0} + \frac{2}{\omega_0}\right) + 1} \\
 &= \frac{1}{\frac{\omega^2}{\omega_0^2} + \frac{2\omega}{\omega_0 A_0} + \frac{1}{A_0^2} + \frac{1}{Q A_0} + \frac{2}{Q \omega_0} + 1} = \frac{\omega_0^2}{\omega^2 + \frac{2\omega \omega_0}{A_0} + \frac{\omega_0^2}{A_0^2} + \frac{\omega_0^2}{Q A_0} + \omega_0^2 + \frac{2\omega_0}{Q}} \\
 &= \frac{\omega_0^2}{\omega^2 + \omega \omega_0 \left(\frac{1}{Q} + \frac{2}{A_0}\right) + \omega_0^2 \left(1 + \frac{1}{Q A_0} + \frac{1}{A_0^2}\right)}
 \end{aligned}$$

$$\Rightarrow \omega_0' = \omega_0 \sqrt{1 + \frac{1}{A_0 Q} + \frac{1}{A_0^2}} \quad \text{RADIAL FREQUENCY SHIFT}$$

$$\Rightarrow \frac{1}{Q'} \approx \left(\frac{1}{Q} + \frac{2}{A_0}\right) \rightarrow Q \left(\frac{1}{Q} - \frac{1}{Q'}\right) = \frac{Q' - Q}{Q' Q} \cdot Q \approx \frac{\Delta Q}{Q} = -\frac{2Q}{A_0} \quad \text{IN-BAND DROP}$$

→ FINITE CBWP EFFECT

Let's write now the real integrator T.F. at high frequency:

$$\text{Hint}''(s) = -\frac{\omega_0}{s} \frac{1}{\left(1 + \frac{2}{\omega_M}\right)}$$

so we get

$$\begin{aligned}
 T(s) &= \frac{\left(\frac{\omega_0}{s}\right)^2 \cdot \frac{1}{\left(1 + \frac{2}{\omega_M}\right)^2}}{\left(\frac{\omega_0}{s}\right)^2 \frac{1}{\left(1 + \frac{2}{\omega_M}\right)^2} + \frac{\omega_0}{Qs} \frac{1}{\left(1 + \frac{2}{\omega_M}\right)} + 1} \\
 &= \frac{1}{1 + \frac{\omega^2}{Q \omega_0} \left(1 + \frac{2}{\omega_M}\right) + \frac{2^2}{\omega_0^2} \left(1 + \frac{2}{\omega_M}\right)^2}
 \end{aligned}$$

This T.F. has two complex conjugate poles + the additional poles due to the integrator. Assuming to work at $|\omega| < \omega_M$ we may write

$$\left(1 + \frac{2}{\omega_M}\right)^2 \left[\frac{\omega^2}{\omega_0^2} + \frac{2}{Q \omega_0} \left(1 + \frac{2}{\omega_M}\right) + \frac{1}{\left(1 + 2/\omega_M\right)^2} \right] = 0$$

$$\left(1 + \frac{2}{\omega_M}\right)^2 \left[\frac{\omega^2}{\omega_0^2} + \frac{2}{Q \omega_0} \left(1 - \frac{2}{\omega_M}\right) + \left(1 - \frac{2}{\omega_M}\right)^2 \right] = 0$$

so we get

$$\frac{\omega_0^2}{\omega_0^2} \left[1 - \frac{\omega_0}{Q\omega_m} + \left(\frac{\omega_0}{\omega_m}\right)^2 \right] + \frac{2}{\omega_0} \left(\frac{1}{Q} - 2\frac{\omega_0}{\omega_m} \right) + 1 = 0$$

↓

$$\omega_0^2 + \frac{2\omega_0}{\left[1 - \frac{\omega_0}{Q\omega_m} + \left(\frac{\omega_0}{\omega_m}\right)^2 \right]} \frac{(1 - 2\frac{\omega_0}{\omega_m} Q)}{Q} + \frac{\omega_0^2}{\left[1 - \frac{\omega_0}{Q\omega_m} + \left(\frac{\omega_0}{\omega_m}\right)^2 \right]} = 0$$

$$\Rightarrow \omega_0' = \frac{\omega_0}{\sqrt{1 - \frac{\omega_0}{Q\omega_m} + \left(\frac{\omega_0}{\omega_m}\right)^2}} \approx \omega_0 \left(1 + \frac{\omega_0}{2Q\omega_m} \right) \quad (\omega_m \gg \omega_0)$$

$$\Rightarrow \frac{1}{Q'} \approx \frac{1}{Q} \frac{(1 - 2\frac{\omega_0}{\omega_m} Q)}{\left[1 - \frac{\omega_0}{Q\omega_m} + \left(\frac{\omega_0}{\omega_m}\right)^2 \right]} \approx \left(\frac{1}{Q} - 2\frac{\omega_0}{\omega_m} \right) \quad (\omega_m \gg \omega_0)$$

So $\frac{Q' - Q}{Q'} \approx \frac{\Delta Q}{Q} = 2\frac{\omega_0}{\omega_m} Q$

CONCLUSION : the real require of the opAMP affects the ω_0 and the Q of the real integrator, so actually it affects the gain of the filter, which becomes more sensitive if the Q factor is higher! This happens because the $\frac{\Delta Q}{Q}$ term are proportional to Q .
 Although the positive Q shifts due to HF imperfections may compensate the negative shift due to the DC gain and the LF poles, we consider the worst case, setting a constant on the maximum absolute value for both the positive and negative margin.
 $\left| \frac{\Delta Q}{Q} \right| \leq \text{Error}\%$

As a final result, we may consider the contribution arising from the positive zero in the open loop op-AMP T.F., so

$$\text{Hint}(s) = -\frac{\omega_0}{2} \left(1 - \frac{2}{\omega_z} \right)$$

$$T(s) = \frac{\text{Hint}^2(s)}{\text{Hint}^2(s) - \frac{1}{Q} \text{Hint}(s) + 1} = \frac{\left(\frac{\omega_0}{2}\right)^2 \left(1 - \frac{2}{\omega_z}\right)^2}{\left(\frac{\omega_0}{2}\right)^2 \left(1 - \frac{2}{\omega_z}\right)^2 + \frac{\omega_0}{Q\omega} \left(1 - \frac{2}{\omega_z}\right) + 1}$$

The poles will be given by the roots of

$$\frac{\omega_0^2}{\omega_0^2} + \frac{2}{Q\omega_0} \left(1 - \frac{2}{\omega_z} \right) + \left(1 - \frac{2}{\omega_z} \right)^2 = 0$$

(68) So we get

$$\frac{\omega^2}{\omega_0^2} \left[1 - \frac{\omega_0}{Q\omega_2} + \left(\frac{\omega_0}{\omega_2} \right)^2 \right] + \frac{2}{Q\omega_0} \left(1 - \frac{2\omega_0}{\omega_2} Q \right) + 1 = 0$$

$$\downarrow$$

$$\omega^2 + \frac{2\omega_0}{\left[1 - \frac{\omega_0}{Q\omega_2} + \left(\frac{\omega_0}{\omega_2} \right)^2 \right]} \frac{\left(1 - \frac{2\omega_0}{\omega_2} Q \right)}{Q} + \frac{\omega_0^2}{\left[1 - \frac{\omega_0}{Q\omega_2} + \left(\frac{\omega_0}{\omega_2} \right)^2 \right]^2} = 0$$

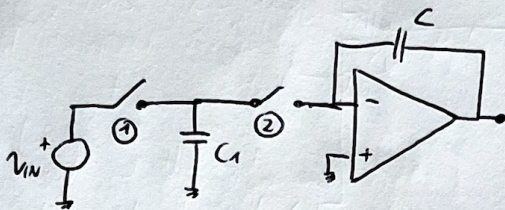
So

$$\omega_0' = \sqrt{1 - \frac{\omega_0}{Q\omega_2} + \left(\frac{\omega_0}{\omega_2} \right)^2} \approx \omega_0 \left(1 + \frac{2\omega_0}{Q\omega_2} \right) \quad (\omega_2 \gg \omega_0)$$

$$\frac{1}{Q'} = \frac{1}{Q} \frac{\left(1 - \frac{2\omega_0}{\omega_2} Q \right)}{\sqrt{1 - \frac{\omega_0}{Q\omega_2} + \left(\frac{\omega_0}{\omega_2} \right)^2}} \approx \frac{1}{Q} \left(1 - \frac{2\omega_0}{\omega_2} \right) \quad (\omega_2 \gg \omega_0)$$

$$\left(\frac{1}{Q} - \frac{1}{Q'} \right) Q \approx \frac{\Delta Q}{Q} = \frac{2\omega_0}{\omega_2} Q \quad \text{POSITIVE SHIFT, line for the GBWP}$$

In the low frequency range we'd require very large resistors in order to have suitable values of capacitors, but they are hard to be made in integrated technology, so we use switched capacitors



Phase ① \Rightarrow ① is closed and C_1 stores $Q = C_1 \cdot v_{IN}$

Phase ② \Rightarrow ② is closed and C_1 charges C , so $v_C = \frac{C_1 v_{IN}}{C}$

At the output we get a staircase with slope $\frac{C_1 v_{IN}}{C \cdot T}$

The output of an integrator has a slope of

~~$\frac{v_{IN}}{RC}$~~

so we can say that the switched capacitor provides $R_{eq} = \frac{I}{C_1}$

ADVANTAGES:

- make large resistors by using small capacitors
- no need of buffer output stages in ladder networks made of integrators
- no delays on the ratio between two capacitors, which can be controlled very well

DISADVANTAGES: we have to deal with a discrete signal, so Shannon must be respected

Therefore we get

$$R_c^{(0)} \approx g_{mS} R_1 R_2$$

and we can safely say that $b_1 \approx C_c g_{mS} R_1 R_2$

Then, we have that

$$b_2 = C_1 C_2 R_1 R_2 + C_1 C_c R_1 R_2 + C_2 C_c R_2 \cdot \frac{1}{g_{mB}} \approx C_1 R_1 R_2 (C_2 + C_c)$$

$$b_3 = C_1 C_2 C_c R_1 R_2 \frac{1}{g_{mB}}$$

Let's consider now the numerator of the T.F.:

~~at~~

$$a_1 = C_1 R_{o1}^{(0)} + C_c R_{oc}^{(0)}$$

$$a_2 = C_1 C_c R_{o1}^{(0)} R_{oc}^{(1)} = C_c (C_1 R_{oc}^{(0)} R_{o1}^{(0)})$$

$$R_{o1}^{(0)} = 0 \quad (\text{nil output} \Rightarrow \text{nil current through } \Gamma_S \Rightarrow \text{nil } v_{gs} = v_S)$$

$$R_{oc}^{(0)} = \frac{1}{2g_{mB}} \left(1 - \frac{1}{g_{mS} R_1} \right)$$

$$\downarrow \\ \approx \frac{1}{2g_{mB}}$$

We substitute C_c with a current source. If the output is i_S , then i_S flows into Γ_S , so $v_{gs} = -\frac{i_S}{g_{mS}}$. The current from i_S merges with the one due to the input pair (split), so we have i_S id through Γ_{B2} and thus

$$i_S - i_d - (i_d) = i_S - 2i_d$$

through R_1 . So we get

$$-\frac{i_S}{g_{mS} R_1} = i_S - 2i_d$$

\downarrow

$$i_d = \frac{i_S}{2} \left[1 + \frac{1}{g_{mS} R_1} \right]$$

so the voltage at the node of Γ_{B2} will be

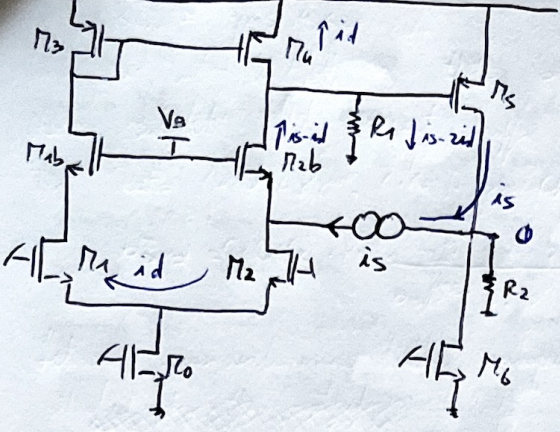
$$\frac{(i_S - i_d)}{g_{mB}} = \frac{1}{g_{mB}} \left[\frac{i_S}{2} - \frac{1}{g_{mS} R_1} \frac{i_S}{2} \right] = v_S$$

$$\Rightarrow \frac{v_S}{i_S} = \frac{1}{2g_{mB}} \left[1 - \frac{1}{g_{mS} R_1} \right]$$

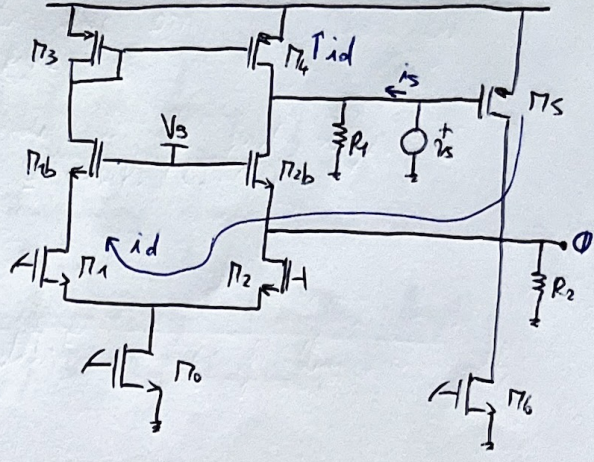
Now, since $R_{o1}^{(0)} = 0$, it's better to use the second form, in order not to get indeterminate results:

$$i_d = i_S = -g_{mS} v_S$$

COMPUTATION OF $R_{oc}^{(c)}$



COMPUTATION OF $R_{oi}^{(c)}$ (A3)



Then $i_S = i_d + \frac{v_S}{R_1} \approx i_d = -g_{m5} v_S$

$R_{oi}^{(c)} \approx -\frac{1}{g_{m5}}$

in the end we have

$a_1 \approx \frac{C_c}{2g_{mB}}$

$a_2 = C_c C_1 \frac{1}{2g_{mB}} \left(-\frac{1}{g_{m5}}\right)$

and

$T(s) \approx g_{m1} g_{m5} R_1 R_2 \cdot \frac{-s^2 \frac{C_c C_1}{2g_{mB} g_{m5}} + s \frac{C_c}{2g_{mB}} + 1}{C_1 C_2 C_c \frac{R_1 R_2}{g_{mB}} s^3 + C_1 R_1 R_2 (C_2 + C_c) s^2 + g_{m5} C_c R_1 R_2 s + 1}$

- The common mode swing sets overdrive constraints for the input stage
- The output swing sets constraints for the overdrive of the output stage
- SR, GBWP and E_{in} set constraints on the current of the input stage
- The gain sets constraints on the output transistors' length
- The offset sets constraints on the input transistors' length
- The noise corner sets constraints on the noise input transistors' length
- CTRR sets constraints on the length of the tail transistor

ROOT LOCUS RULES

- # BRANCHES = # POLES OF $G_{loop}(s)$;
- Branches proceed from poles to zeros (finite or infinite);
- $\gamma < 0$ regions on the $\text{Re}[s]$ axis on the ~~right~~ left of an odd number of singularities;
- $\gamma > 0$ regions of the $\text{Re}[s]$ axis on the left side of an even number of singularities;
- # ASYMPTOTES = (# POLES) - (# ZEROS @ FINITE FREQUENCY)
- ASYMPTOTES = branches going to ∞ that always split the complex plane into even parts

Write the T.F. in the Evans form.

The switched capacitor (SC) mimics the behavior of a large resistance, as it is widely used in filter design in audio range. We can make integrators based on a SC, who basically integrates current pulses, hence providing voltage steps at the output. The result is a staircase output waveform with an equivalent ramp-rate

$$\frac{\Delta V}{T} = \frac{E \cdot C_1}{C} \cdot \frac{1}{T} = \frac{E}{C \cdot R_{eq}}$$

$$R_{eq} = \frac{T}{C_1}$$

The discrete approximation is good enough provided that the clock frequency is much larger than the BW of the input signal, hence it respects Shannon's theorem. This can be easily done, since typical freq are ≈ 1 MHz, while the audio range is ≈ 20 kHz at the most.

We've already discussed the advantages, let's see now what's the bad part of dealing with discrete time systems:

We can write the output waveform like a staircase

$$v_{out}(t) = \sum_{m=0}^{100} v_{out}(mT) \times \left\{ \text{rect} \left[\frac{t-mT}{T} \right] \right\}$$

where

$$\text{rect}(x) = \begin{cases} 1 & |x| < \frac{1}{2} \\ 0 & |x| > \frac{1}{2} \end{cases}$$

and

$$\mathcal{F}[\text{rect}(t)](f) = \int_{-\frac{1}{2}}^{\frac{1}{2}} e^{-j2\pi f t} dt = -\frac{1}{2\pi f j} \left[e^{-j2\pi f t} \right]_{-\frac{1}{2}}^{\frac{1}{2}} =$$

~~$$= \frac{1}{\pi f} \cdot \frac{1}{2j} \left[e^{j\pi f} - e^{-j\pi f} \right] = \frac{\sin(\pi f)}{\pi f} = \text{sinc}(f)$$~~

$$= \frac{1}{\pi f} \cdot \frac{1}{2j} \left[e^{j\pi f} - e^{-j\pi f} \right] = \frac{\sin(\pi f)}{\pi f} = \text{sinc}(f)$$

Let's see now what is the link between the Fourier transform of the continuous-time signal and the discrete-time one, from input to output:

$$\mathcal{F}[v_{out}(t)] = \sum_{m=0}^{100} v_{out}(mT) \cdot \underbrace{e^{-j\omega mT} \text{sinc}\left(\frac{f}{T}\right) \cdot T}_{\mathcal{F}\left\{ \text{rect}\left[\frac{t-mT}{T}\right] \right\}} = \sum_{m=0}^{100} v_{out}(mT) z^{-m} \Big|_{z=e^{j\omega T}} \cdot \underbrace{T \text{sinc}\left(\frac{f}{T}\right)}_{\text{ZETA TRANSFORM}}$$

So we come to this result:

$$V_{out}(f) = V_{in}(z) \Big|_{z=e^{j2\pi fT}} \cdot T \operatorname{sinc}\left(\frac{f}{f_s}\right)$$

Now, from the circuit we have that

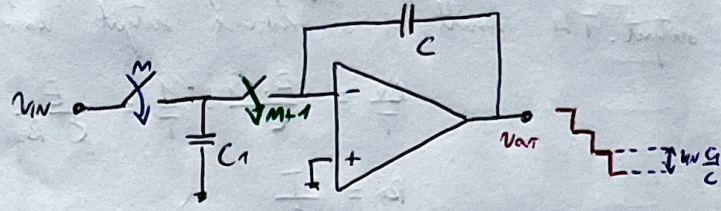
$$V_{out}(m+1) = V_{out}(m) - V_{in}(m) \frac{C_1}{C}$$

↓ ZETA TRANSFORM

$$V_{out}(z) \cdot z = V_{out}(z) - V_{in}(z) \frac{C_1}{C}$$

↓

$$\frac{V_{out}(z)}{V_{in}(z)} = -\frac{C_1}{C} \frac{1}{z-1} = H(z) \Rightarrow \text{Transfer function of the discrete-time filter}$$



So we can write

$$V_{out}(f) = V_{in}(z) \cdot H(z) \Big|_{z=e^{j2\pi fT}} \cdot T \operatorname{sinc}\left(\frac{f}{f_s}\right)$$

where

$$\begin{aligned} V_{in}(z) \Big|_{z=e^{j2\pi fT}} &= \sum_{m=0}^{+\infty} v_{in}(mT) e^{-j2\pi fTm} = \int_{-\infty}^{+\infty} v_{in}(t) \cdot \sum_{m=0}^{+\infty} \delta(t-mT) e^{-j2\pi fTm} dt \\ &= \int [v_{in}(t) \sum_{m=0}^{+\infty} \delta(t-mT)] dt \\ &= V_{in}(f) * \sum_{k=-\infty}^{+\infty} \frac{1}{T} \delta\left(f - \frac{k}{T}\right) \end{aligned}$$

and, in the end, we have

$$V_{out}(f) = \underbrace{\left[V_{in}(f) * \sum_{k=-\infty}^{+\infty} \frac{1}{T} \delta\left(f - \frac{k}{T}\right) \right]}_{\text{aliasing due to sampling}} \cdot \underbrace{\left[-\frac{C_1}{C} \frac{1}{e^{j2\pi fT} - 1} \right]}_{\text{action of the integrator, is a peaking filter}} \cdot \underbrace{\left[T \operatorname{sinc}\left(\frac{f}{f_s}\right) \right]}_{\text{Spectrum cardinal sine due to sampling}}$$

In order to have a reliable output, we need:

- a RECONSTRUCTION FILTER that kills all the HF harmonics due to aliasing;
- an ANTI-ALIASING FILTER before the the SC filter in order to kill HF disturbances that may be brought to base-band due to aliasing;
- an EQUALIZING FILTER that compensates the spectrum shape alteration due to the cardinal sine term.

All these filters can be easily implemented by continuous-time networks