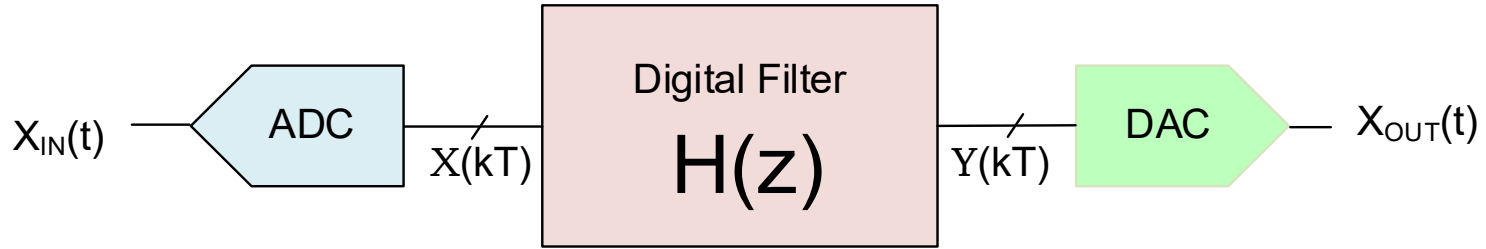


# EE 508 Lecture 42

Digital/Analog Filter Comparisons  
Some Recent Filter Structures

# Digital Filter Properties



Theorem: Any FIR filter is linear phase if the impulse response is symmetric or antisymmetric

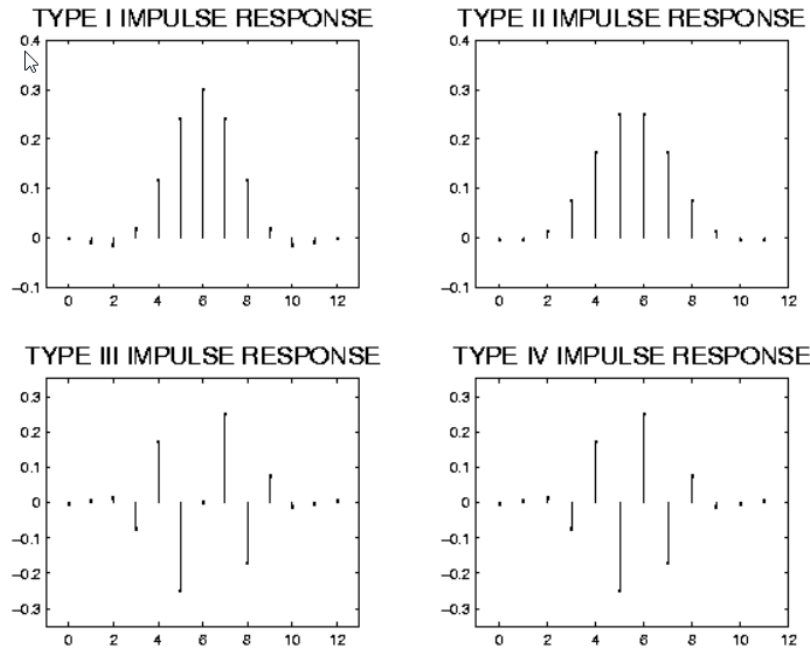
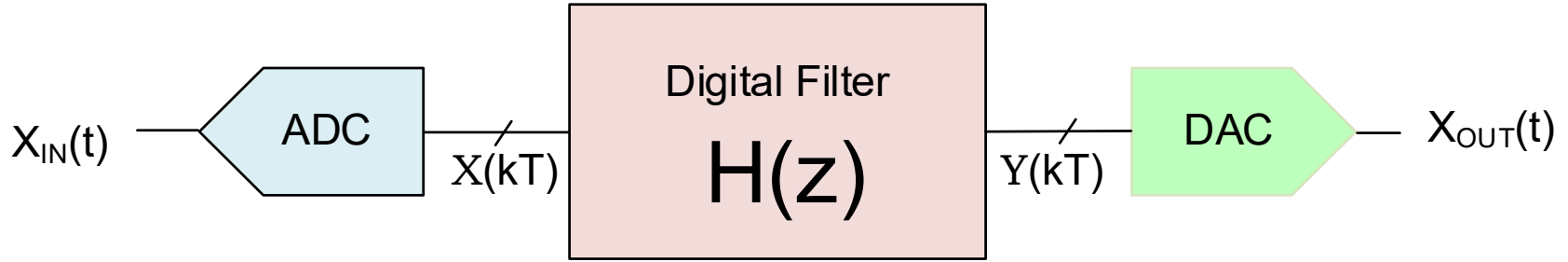


Figure 1

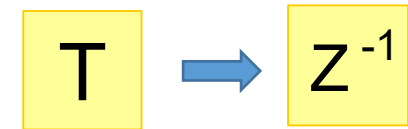
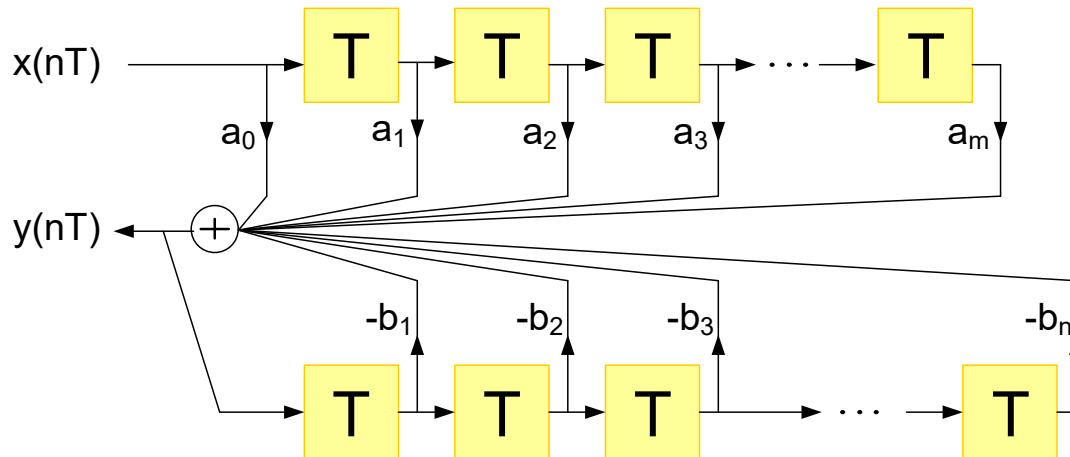
Table from Robert Novak book

# Digital Filter Properties

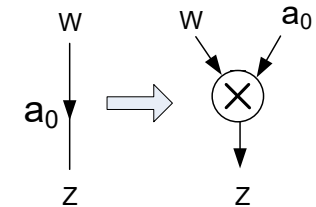


$$y(nT) = \sum_{i=0}^m a_i x(nT - iT) + \sum_{i=1}^n b_i y(nT - iT)$$

## An Implementation of a Digital Filter



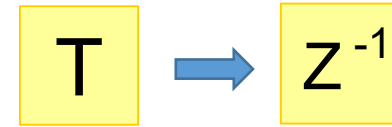
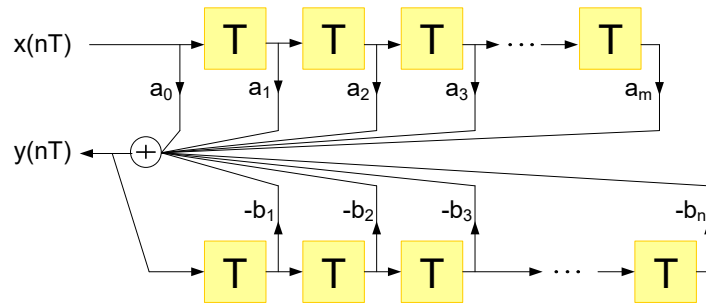
Delay Element



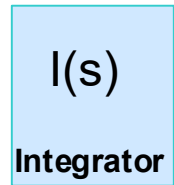
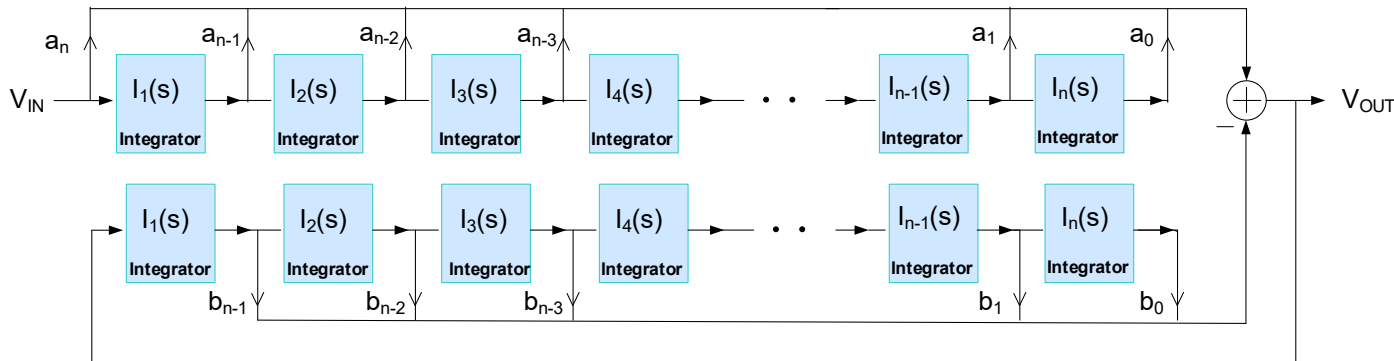
Multiply Element

## Review from last lecture

# An Implementation of a Digital Filter



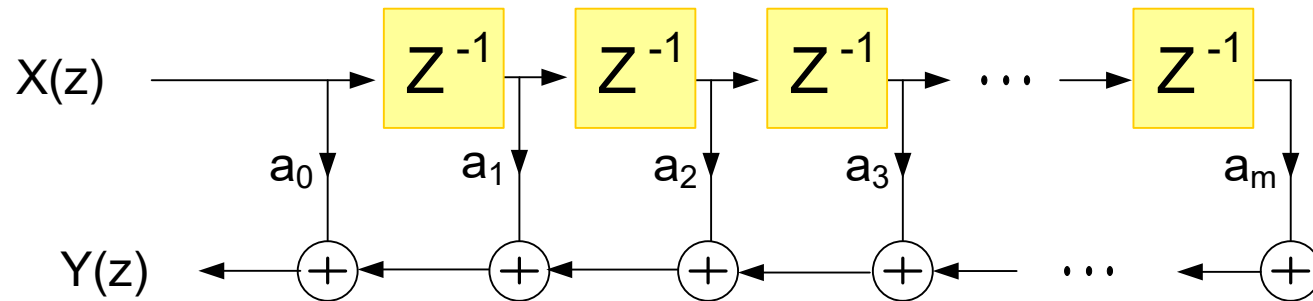
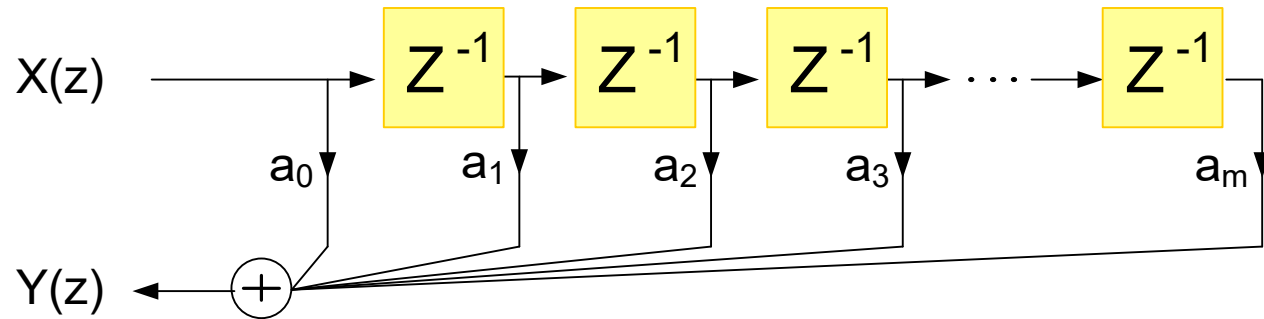
# An Implementation of an Analog Filter



- Can be viewed as analogous implementations
- Neither particularly practical
- Many other architectures for both analog and digital filters
- Approximately double the number of integrators or delay elements needed

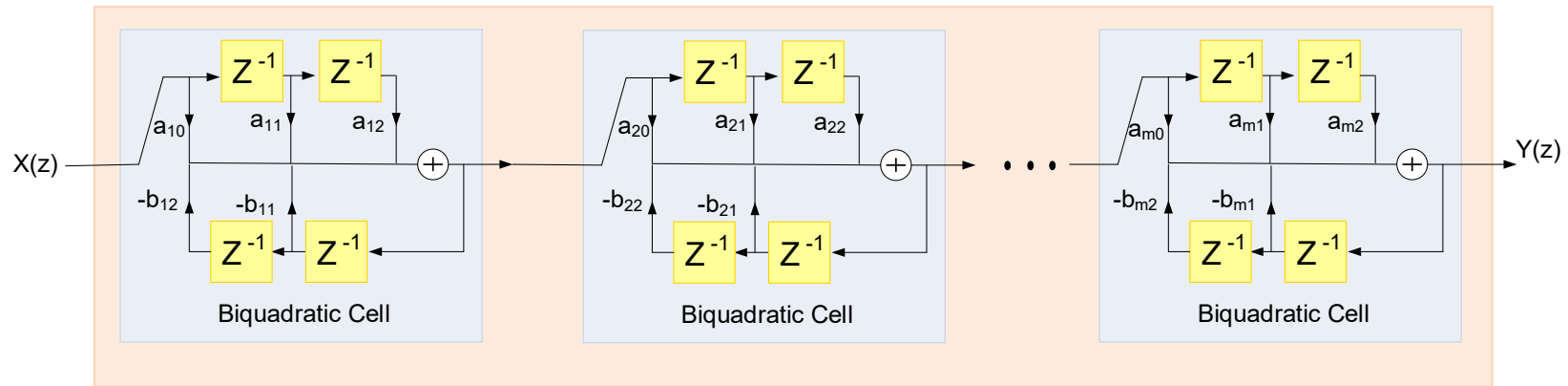
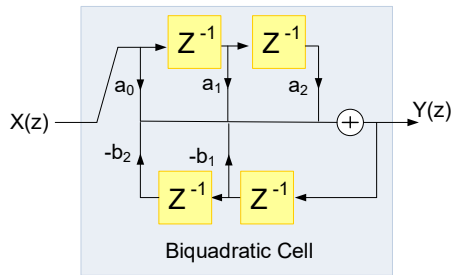
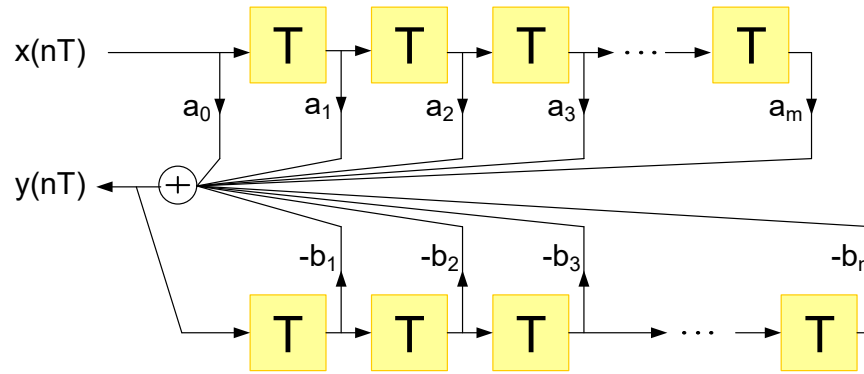
## Review from last lecture

# Alternate Implementations of an FIR Digital Filter



# Review from last lecture

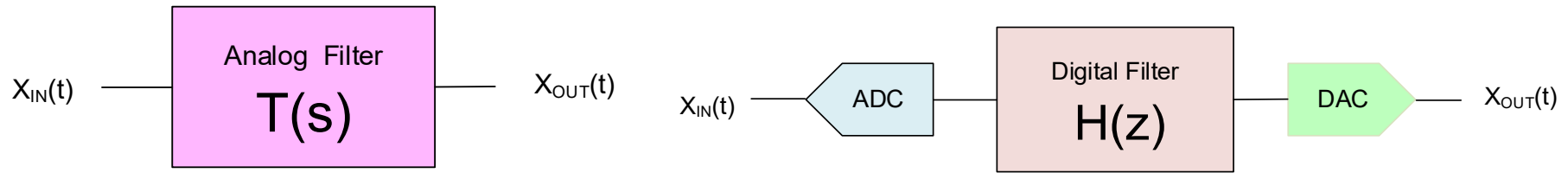
## Alternate Implementations of IIR Digital Filter



Excessive delay elements but not of as much concern as excessive Integrators

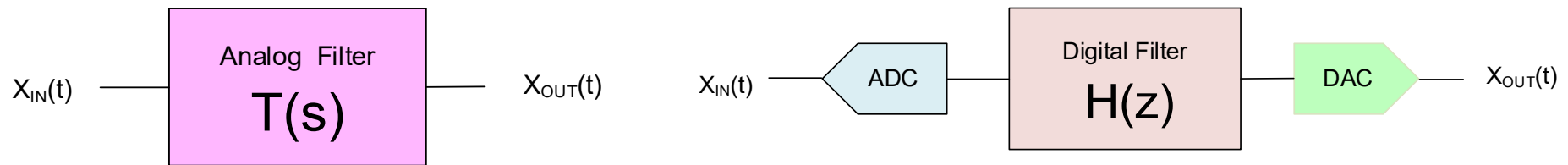
## Review from last lecture

# Does Digital Filter Overcome Limitations



- A - Transfer functions sensitive to component and process variations
- D - Transfer function part of  $H(z)$  not sensitive to process variations
  - Transfer function sensitive to coefficient quantization
  - ADC and DAC minimally sensitive to process variations but highly sensitive to mismatch
- A - Distortion inherent due to nonlinearities in components (particularly amplifiers)
- D - Transfer function part of  $H(z)$  not sensitive nonlinearity of components
  - ADC and DAC sensitive to nonlinearity of components
- A - Power dissipation can be large
- D - Power dissipation can be large due to a large number of arithmetic operations during each clock cycle
  - ADC and DAC dissipate considerable energy for high resolution or high speed

# Does Digital Filter Overcome Limitations



- A - Area gets large, often unacceptably so for very low frequency poles and even of concern for audio-frequency poles
- D - Area for DSP in Digital Filter can be large
  - ADC and DAC can become large if high resolution is required
  - No area penalty for low frequency operation of digital system
- A - Programmability introduces considerable complexity (with existing approaches)
- D - Programmability of filter characteristics is very efficient with digital filter approach
- A - Making minor changes in filter requirements often necessitates a major redesign effort
- D - Making minor or even major changes in filter requirements requires minimal effort with digital filter approach



Review from last lecture

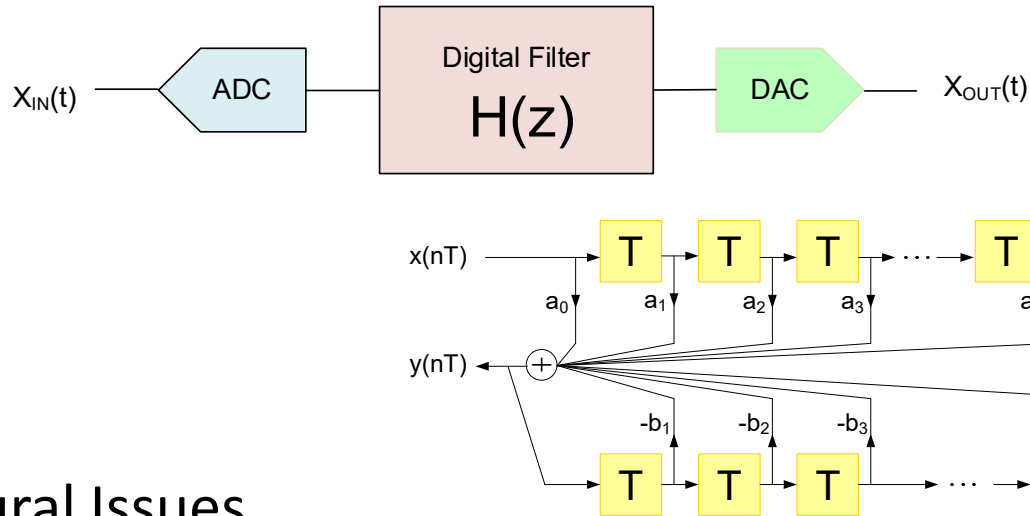
# Digital Filter Design Issues



## Order of Digital Filters Can be Large

- 128 or more delay elements are not uncommon
- Can achieve very steep transitions from passband to stop band
- High Q poles can be practically realized
- Particularly attractive for filtering low-frequency signals
- Large number of adds and multiplies slows response of the filter
- ARMA filters invariably are of lower order than FIR filters for given transition requirements
- FIR filters inherently stable

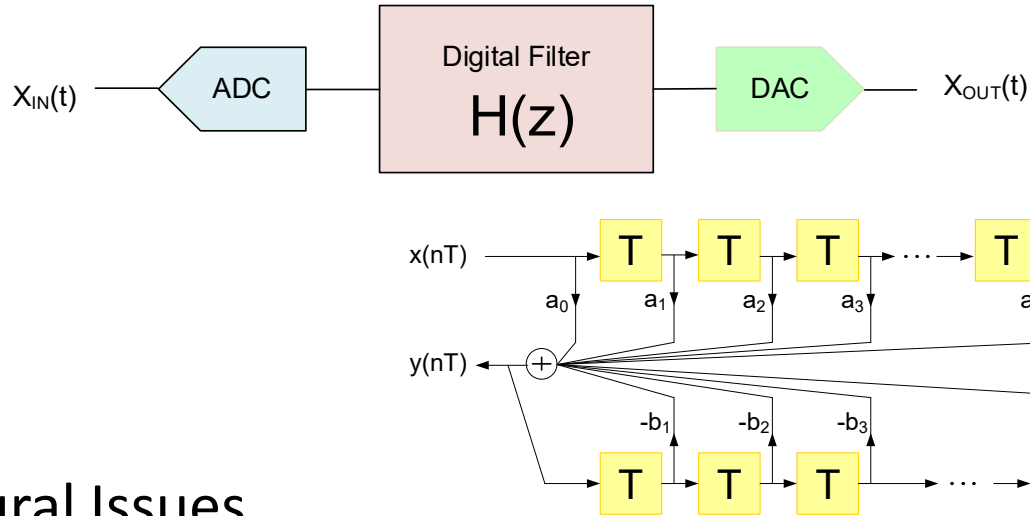
# Digital Filter Design Issues



## Architectural Issues

- Many different filter architectures
- Must be sure to not overflow registers during intermediate calculations
- Order of operations for given architecture can affect performance
- Coefficient sensitivity can be high
- Number of bits of resolution on coefficients affects multiply and add times
- Some work on filters where all coefficients are power of 2 (multiplies become simply shifts)
- Concerns about how many intermediate memory locations are required

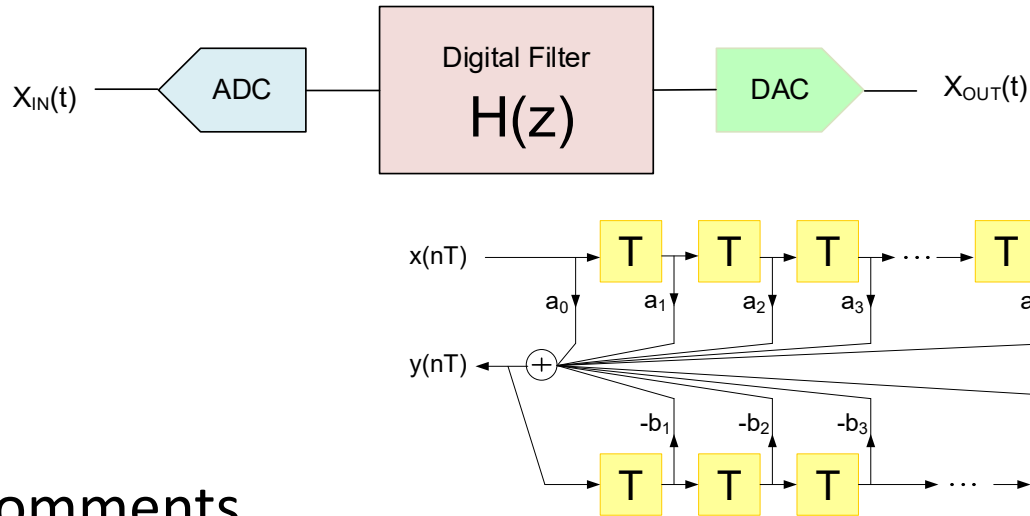
# Digital Filter Design Issues



## Architectural Issues

- May not be easy to assess overflow concerns without overdesign since intermediate totals dependent upon input
- Architecture affects number of arithmetic operations
- Large number of operations can introduce noise into substrate which of concern with systems with extreme SNR where ADC and DAC are on-chip
- Some architectures and some approximations naturally support parallel operations

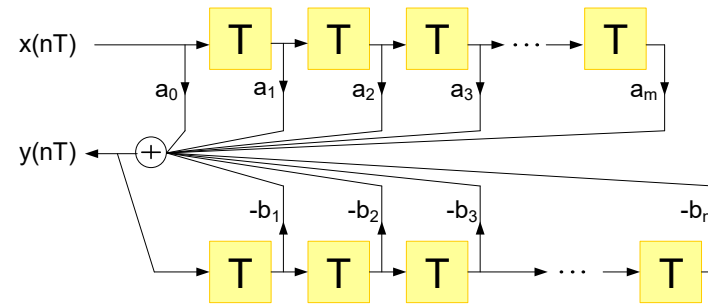
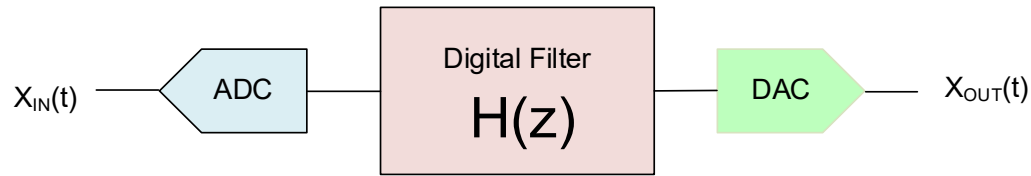
# Digital Filter Design Issues



## General Comments

- Extreme precision possible with right order and good implementation
- Time and amplitude quantization both affect performance
- Not practical for applications that have very high frequency poles (due to both data converter and filter limitations)
- Power dissipation can be large if many arithmetic operations are required
- ADC and DAC design efforts can be substantial
- ADC and DAC may require considerable area and power
- Significant effort in design of computer or DSP to drive the digital filter

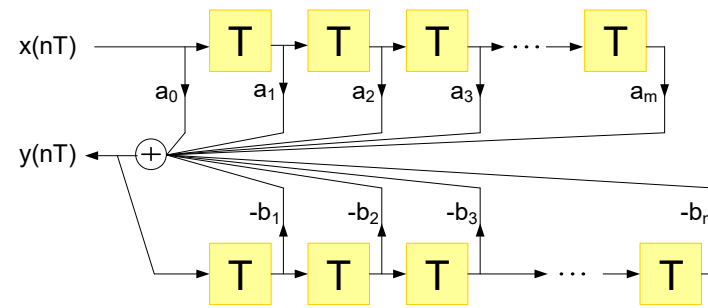
# Digital Filter Design Issues



## General Comments

- Though process variations in digital filter not of concern, they do affect the ADC and DAC designs beyond matching (e.g. clock skew)
- Big step in area and power to implement the DSP and filter
- Switched Capacitor filters have some properties of a digital filter (time-quantization and thus  $H(z)$  instead of  $T(s)$ ) and some of analog filters but overhead for implementing a lower-order filter with SC techniques is relatively small
- DAC often not required since decisions are often made in digital logic and no subsequent analog output is required
- One (of many) applications that favor use of digital filters is in output filtering and decimation in delta-sigma ADCs

# Digital Filter Design Issues




## General Comments

- Digital filters are vulnerable to aliasing
- Digital filters are expensive
- Digital filters limited to relatively low frequency operation (due to both the data converters and the adds/multiplies)
- Digital filter intermediate results can be stored for later analysis
- The  $H(z)$  portion of the digital filter benefits from technology scaling
- The  $H(z)$  portion does not drift with time or temperature
- $H(z)$  can be easily tweaked or even modified with software



**SECTION 6**  
**DIGITAL FILTERS**  
*Walt Kester*

**COMPARISON BETWEEN FIR AND IIR FILTERS**



<b>IIR FILTERS</b>	<b>FIR FILTERS</b>
<b>More Efficient</b>	<b>Less Efficient</b>
<b>Analog Equivalent</b>	<b>No Analog Equivalent</b>
<b>May Be Unstable</b>	<b>Always Stable</b>
<b>Non-Linear Phase Response</b>	<b>Linear Phase Response</b>
<b>More Ringing on Glitches</b>	<b>Less Ringing on Glitches</b>
<b>CAD Design Packages Available</b>	<b>CAD Design Packages Available</b>
<b>No Efficiency Gained by Decimation</b>	<b>Decimation Increases Efficiency</b>



# SECTION 6

## DIGITAL FILTERS

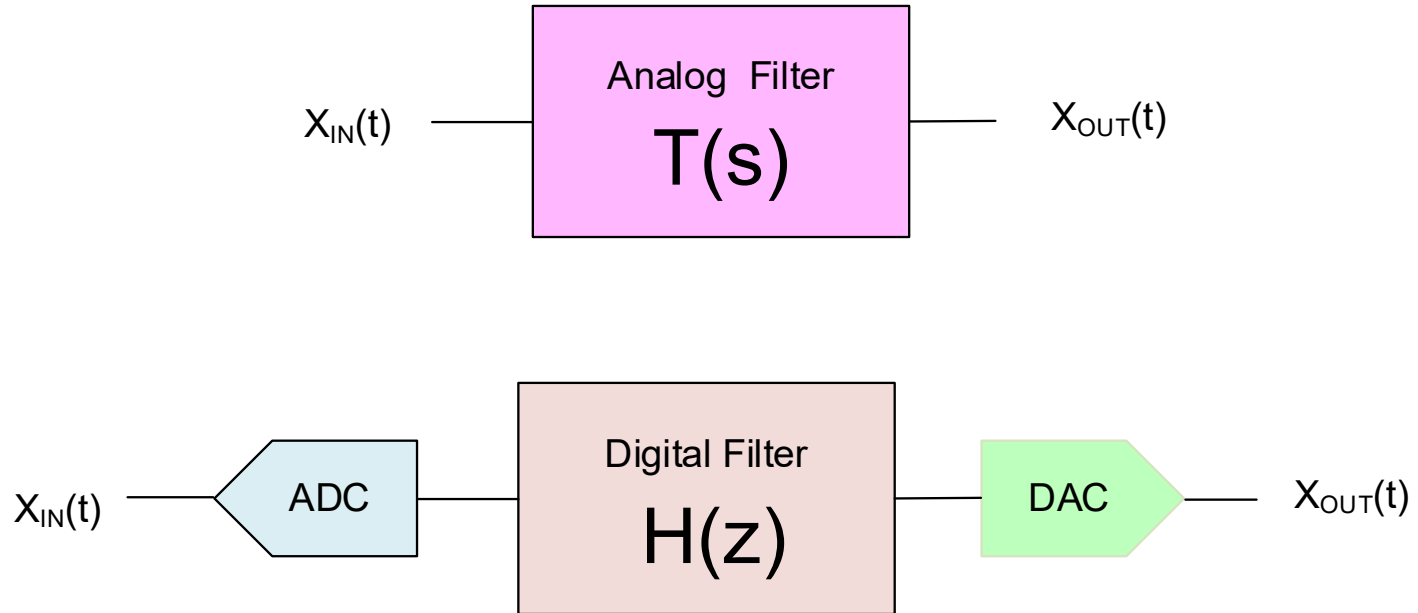
*Walt Kester*

### DIGITAL VERSUS ANALOG FILTERING

DIGITAL FILTERS	ANALOG FILTERS
High Accuracy	Less Accuracy - Component Tolerances
Linear Phase (FIR Filters)	Non-Linear Phase
No Drift Due to Component Variations	Drift Due to Component Variations
Flexible, Adaptive Filtering Possible	Adaptive Filters Difficult
Easy to Simulate and Design	Difficult to Simulate and Design
Computation Must be Completed in Sampling Period - Limits Real Time Operation	Analog Filters Required at High Frequencies and for Anti-Aliasing Filters
Requires High Performance ADC, DAC & DSP	No ADC, DAC, or DSP Required

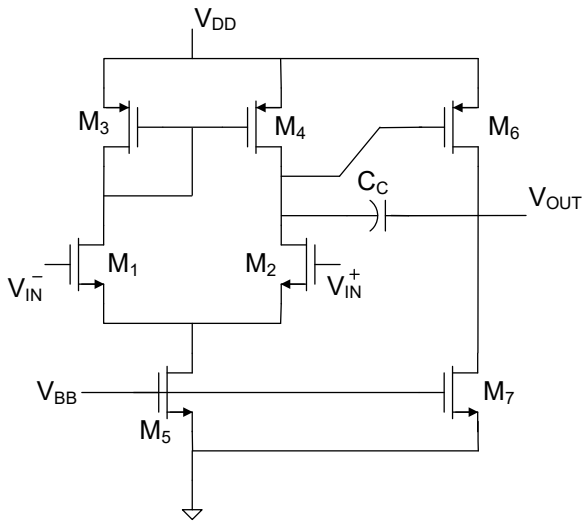
Figure 6.2

# Analog vs Digital Filter

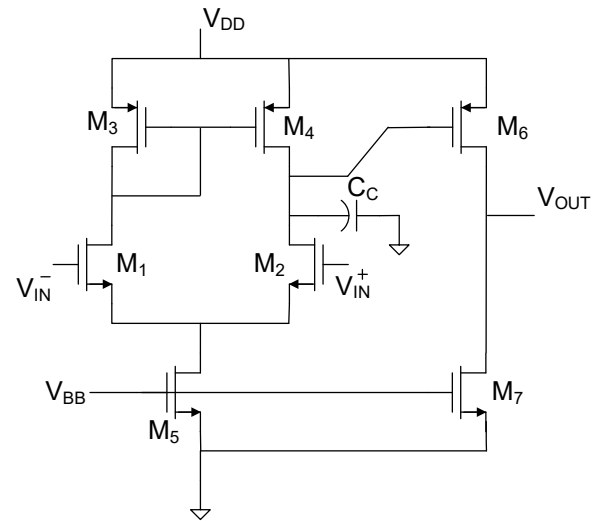


- Both approaches have advantages and limitations
- Digital filters particularly attractive if DSP already available and if ADC and DAC are necessary for other purposes or if decisions in system must be made in the digital domain
- Digital filters also attractive if much of the signal processing will occur in the digital domain of a system
- Digital filters have replaced analog filters in many applications

# Towards Active R Filters



Internally Compensated 7T Op Amp

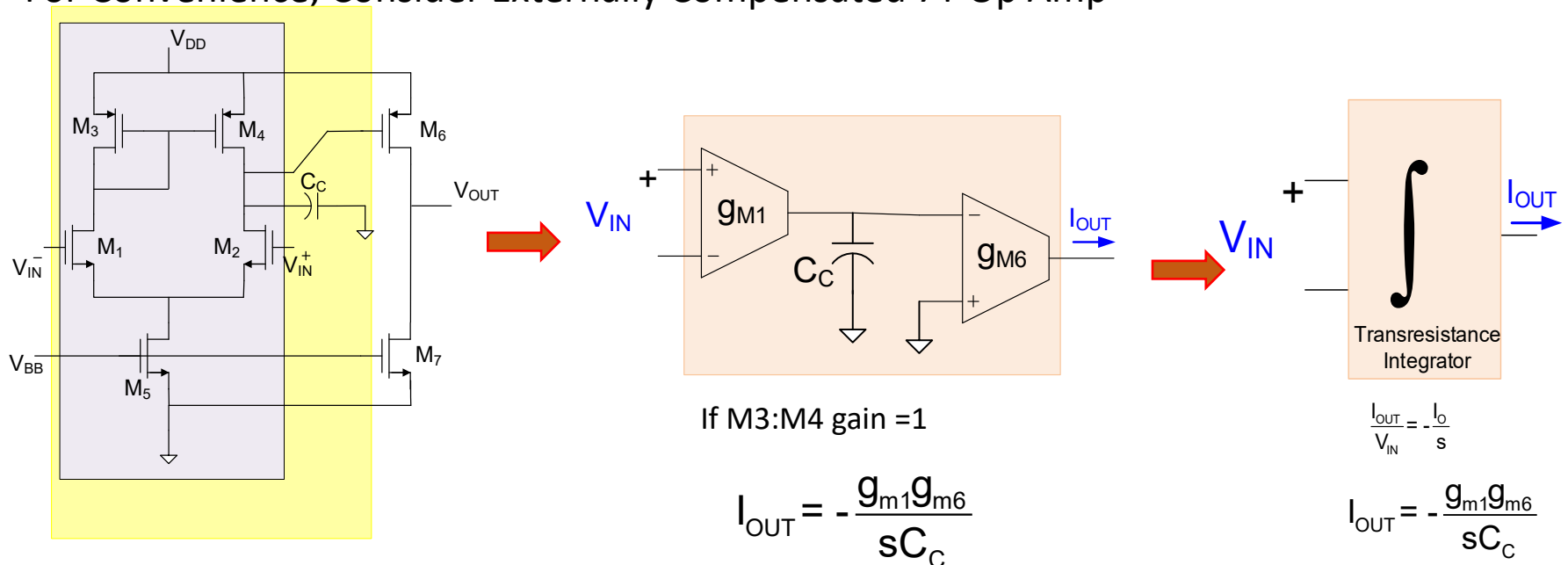


Externally Compensated 7T Op Amp

- Electrical Characteristics are Similar
- Have a single energy storage element,  $C_C$

# Towards Active R Filters

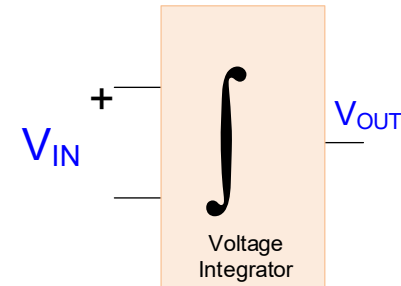
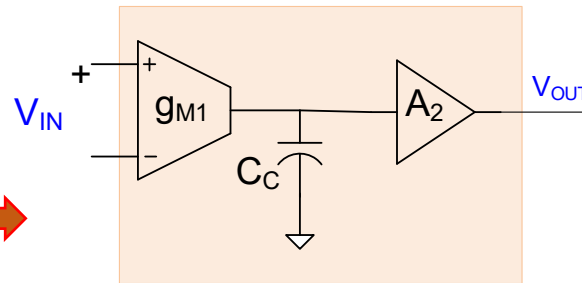
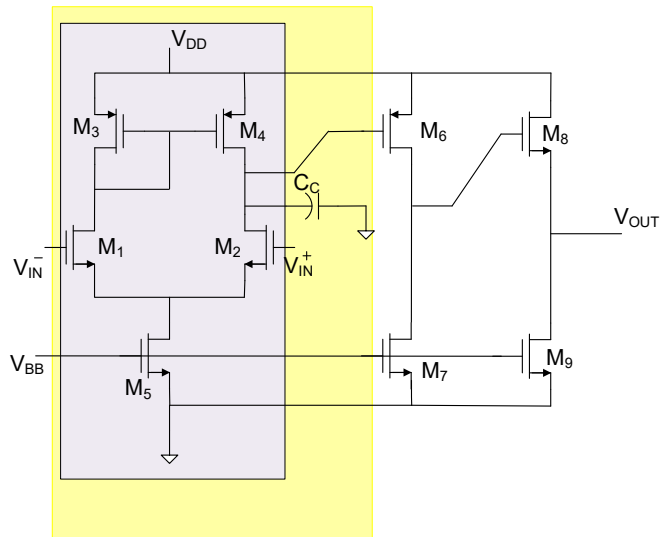
For Convenience, Consider Externally Compensated 7T Op Amp



- Behaves as Transresistance Integrator !
- Though the first commercial OTA was introduced in late 1960's the use of OTAs to design filters received almost no attention for almost 15 years
- Concept developed for a two-stage externally compensated op amp, basic properties exist for most high output impedance op amps

# Towards Active R Filters

For Convenience, Consider Externally Compensated 9T Op Amp



$$A_2 = -\frac{g_{m6}}{g_{o6} + g_{o7}}$$

If M3:M4 gain = 1

$$V_{OUT} = \frac{A_2 g_{m1}}{s C_C} V_{IN}$$

$$I_O = -\frac{A_2 g_{m1}}{s C_C} V_{IN}$$

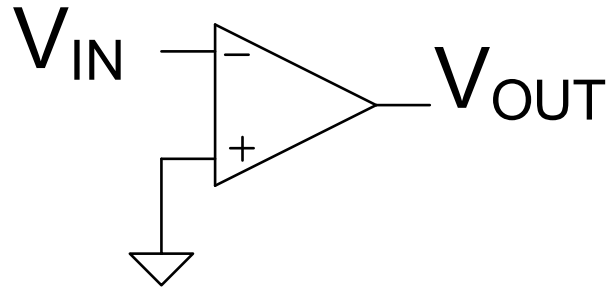
$$\frac{V_{OUT}}{V_{IN}} = -\frac{I_O}{s}$$

$$I_O = -\frac{A_2 g_{m1}}{s C_C} V_{IN}$$

- Behaves as Voltage Integrator !
- Though concept developed for a two-stage externally compensated op amp, basic properties exist for most low output impedance op amps

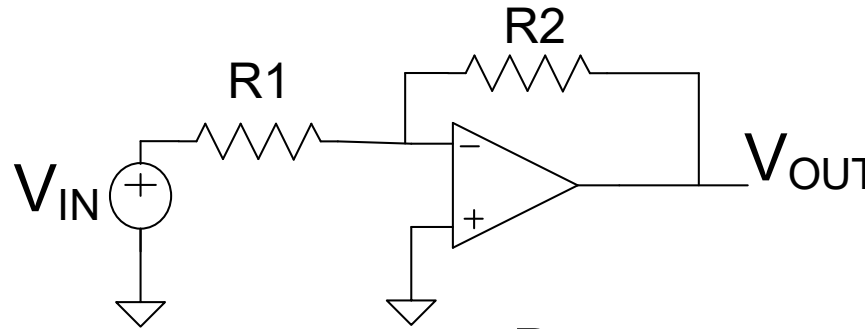
# Towards Active R Filters

Op Amp or Integrator?



$$\frac{V_{OUT}}{V_{IN}} \approx -\frac{GB}{s}$$

# Amplifier or LP Filter or Lossy Integrator?



$$\frac{V_{OUT}}{V_{IN}} \approx \frac{-\frac{R_2}{R_1}}{1 + \tau s \left( 1 + \frac{R_2}{R_1} \right)}$$

- In about 1974 Michael Soderstrand introduced this concept for building high-frequency filters and termed these “Active-R” filters
- Concept of incorporating op amp pole in determining filter response reported a bit earlier
- The compensation capacitor in the op amp serves as the energy storage element in the filter
- Can operate at very high frequencies but many problems with linearity and accuracy

[CITATION] Active R filters: Active filters using only resistors and amplifiers

MA Soderstrand - 8th Asilomar Conf. Circuits, 1974

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[A bandpass filter using the operational amplifier pole](#)

KR Rao, S Srinivasan - IEEE Journal of Solid-State Circuits, 1973 - ieeexplore.ieee.org

The pole of an operational amplifier and a grounded capacitor are used for obtaining a high Q bandpass function. The utilization of the pole of the operational amplifier enables the extension of its useful frequency range. The gain and the bandwidth of the operational amplifier are the primary factors determining the filter performance. The experimental results of a low-sensitivity filter circuit are presented. The circuit is suitable for integration.

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[Active R filters: review of theory and practice](#)

JR Brand, R Schaumann - IEE Journal on Electronic Circuits and Systems, 1978 - IET

... procedures for 1st-, 2nd- and higher-order **active R filters**; it is shown that the 1st- and 2nd-... **active** ^? topology. The paper discusses nonideal aspects and limitations of **active R filter** ...

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[Design of active R filters using only resistors and operational amplifiers](#)

MA SODERSTRAND - International Journal of Electronics ..., 1976 - Taylor & Francis

... In this section, we shall consider the practical applications of **active R filters** presently and in the fut~re. Emphasis will be on the difficulties cncounterecl, on how they effect implementat,...

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# Selected Recent Publications on Analog Filter Design



ISCAS 2024

# 0.5V 32nW Inverter-Based Gm-C Filter for Bio-Signal Processing

Ali Namdari, Orazio Aiello, Daniele D. Caviglia  
<sup>1</sup>DITEN, University of Genova  
ali.namdari@edu.unige.it; orazio.aiello@unige.it; daniele.caviglia@unige.it

180nm CMOS     $\omega_0$ : 470 Hz

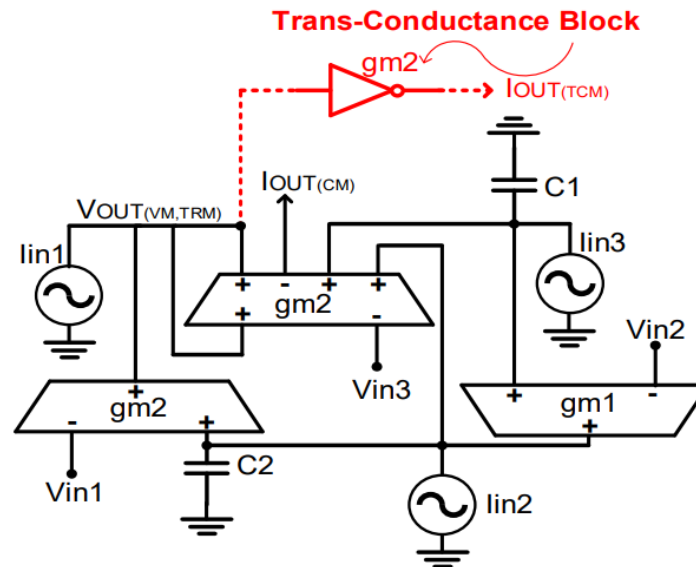


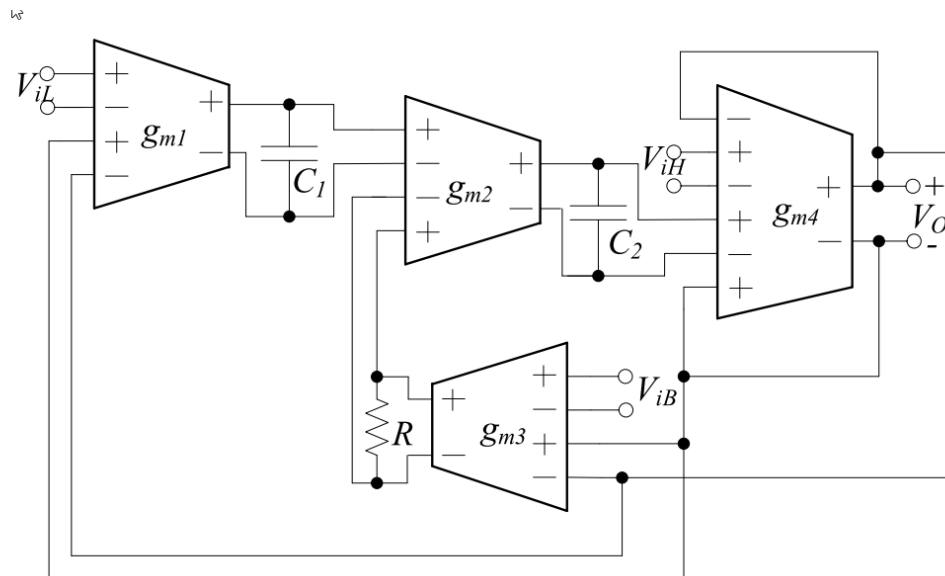
Fig.1. The proposed universal multi-mode Gm-C filter

# 0.5 V Fully Differential Universal Filter Based on Multiple Input OTAs

WINAI JAIKLA<sup>1</sup>, FABIAN KHATEB<sup>2,6</sup>, MONTREE KUMNGERN<sup>3</sup>,  
TOMASZ KULEJ<sup>4</sup>, RAJEEV KUMAR RANJAN<sup>5</sup>, (Member, IEEE),  
AND PEERAWUT SUWANJAN<sup>1</sup>

IEEE Access, Oct 2020

180nm Process,  $f_0=1\text{Hz}$



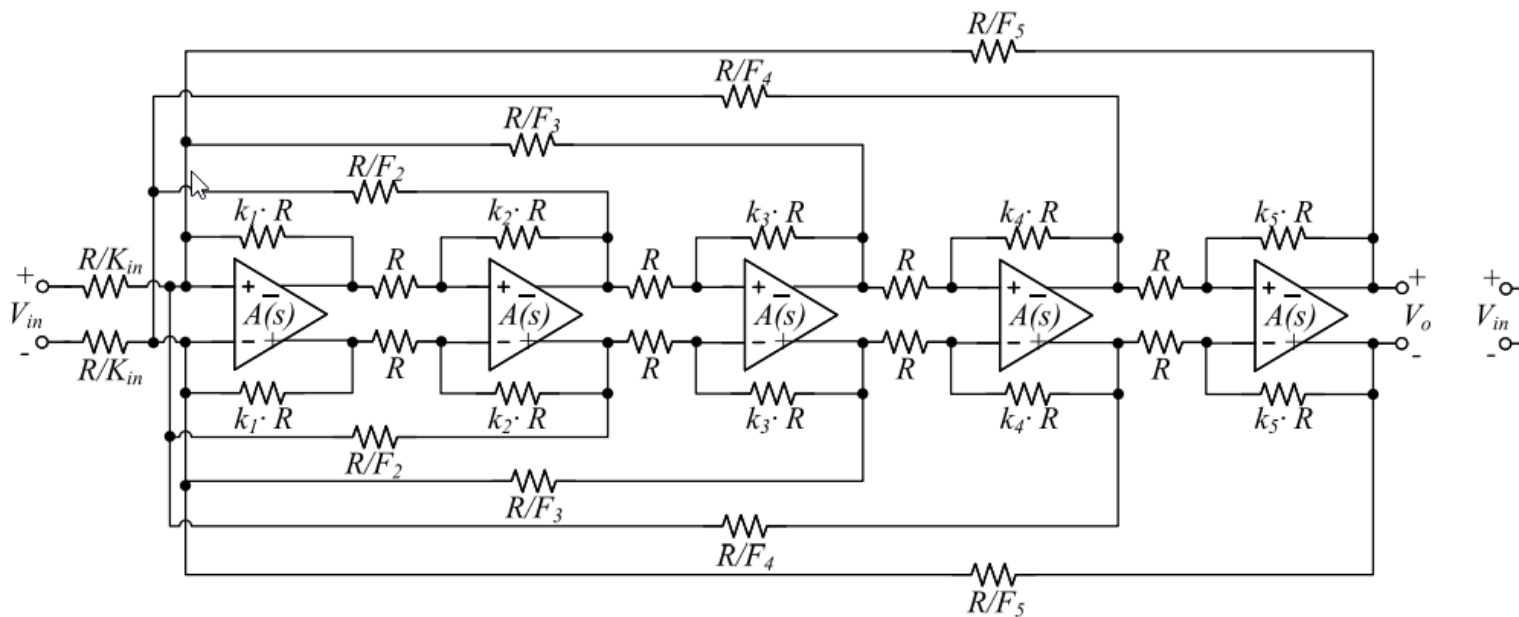
**FIGURE 3.** Proposed fully differential universal filter.

TCAS I May 2021

# Synthesis of High-Order Continuously Tunable Low-Pass Active-R Filters

Adriana C. Sanabria-Borbón<sup>1</sup>, Member, IEEE, and Edgar Sánchez-Sinencio<sup>1</sup>, Life Fellow,

TSMC 180nm process,  $\omega_0$  from 1 to 50 MHz





# A 0.6-V Power-Efficient Active-RC Analog Low-Pass Filter With Cutoff Frequency Selection

Fernando Lavallo-Aviles<sup>1</sup>, *Member, IEEE*, and Edgar Sánchez-Sinencio<sup>1</sup>, *Life Fellow, IEEE*

130 nm CMOS Process

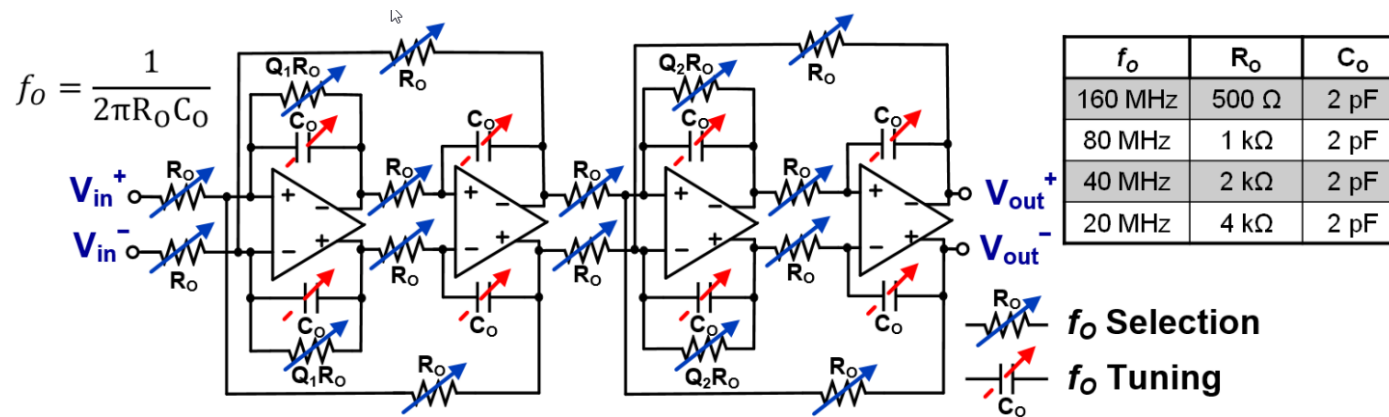


Fig. 1. FD LV fourth-order Butterworth filter implementation.

JSC July 2020

# Analysis and Design of a 260-MHz RF Bandwidth +22-dBm OOB-IIP3 Mixer-First Receiver With Third-Order Current-Mode Filtering TIA

Giacomo Pini<sup>1</sup>, Student Member, IEEE, Danilo Manstretta<sup>1</sup>, Member, IEEE,  
and Rinaldo Castello<sup>1</sup>, Life Fellow, IEEE

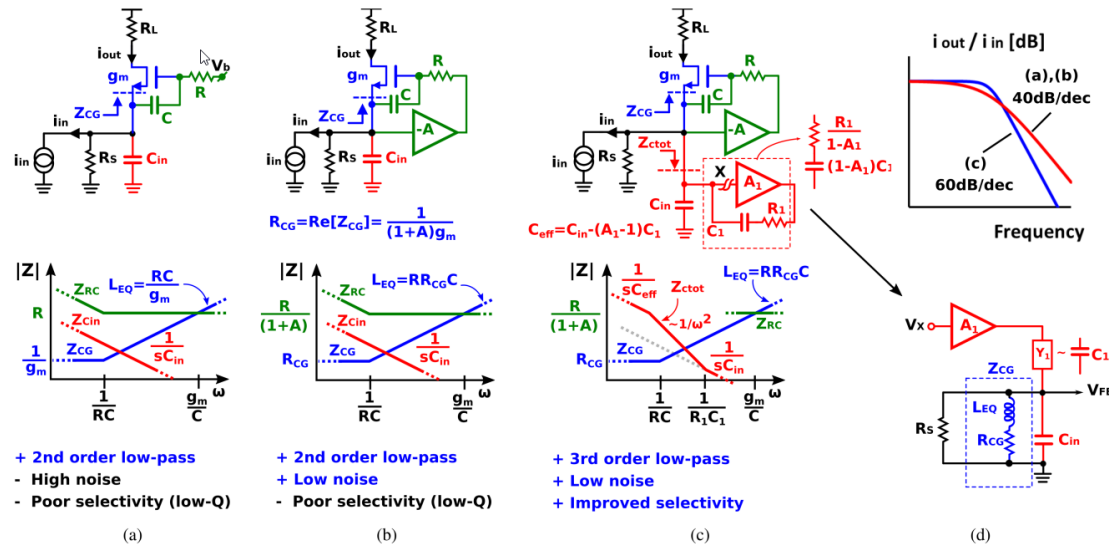


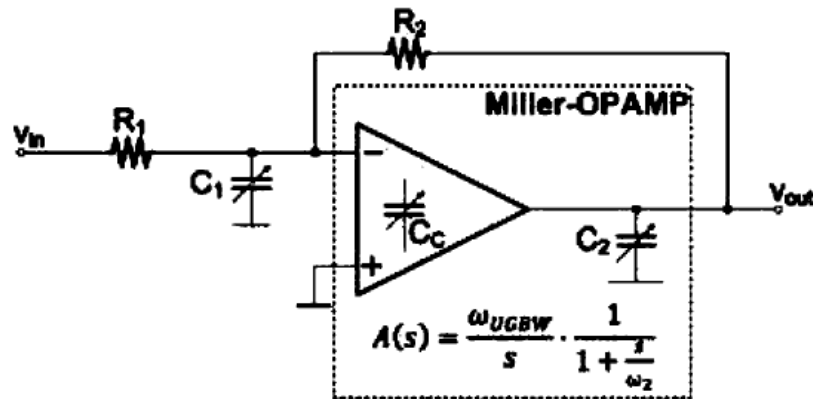
Fig. 1. Schematic representation of the CG-based TIAs and the corresponding impedance magnitude plots. (a) Filtering CG, (b) regulated cascode, (c) regulated cascode with frequency-dependent negative capacitance, and (d) simplified schematic for loop gain calculation and  $i_{out}/i_{in}$  frequency response of (a)–(c).

# A 0.9V 3rd-Order Single-OPAMP Analog Filter in 28nm CMOS-bulk

Marcello De Matteis<sup>1,2</sup>, Andrea Donno<sup>3,4</sup>, Stefano Marinaci<sup>4</sup>, Stefano D'Amico<sup>3,4</sup>, Andrea Baschiroto<sup>1,2</sup>

From IEEE Int. Workshop on Advances in Sensors and Interfaces, June 2017

“The scheme take advantage of the efficient Active-gm-RC filter [3], which exploits the Opamp unity gain bandwidth (COUGBW) to synthesize the transfer function.”



g. 1 – Single ended architecture of the proposed analog filter

Tab. I – Targeted filter transfer function parameters

Parameter	This Design
$\omega_{23}$ – real pole frequency	$2 \cdot \pi \cdot 350\text{MHz}$
$\omega_0$ – complex poles frequency	$2 \cdot \pi \cdot 160\text{MHz}$
$Q_0$ – complex pole quality factor	0.9
$f_{3dB}$ - cut-off frequency	$2 \cdot \pi \cdot 132\text{MHz}$
$G$ – low pass filter dc-gain	0dB

- [3] A. Donno, S. D'Amico, M. De Matteis, A. Baschiroto “A 150MHz 3rd-order single Opamp continuous-time analog filter in 28nm CMOS technology” Proceedings of the IEEE International Conference on Electronics, Circuits, and Systems, ICECS 2015, Cairo (Egypt); 6-9 December 2015 (DOI: 10.1109/ICECS.2015.7440274).

# A 0.9V 600MHz 4<sup>th</sup>-Order Analog Filter with Feed-Forward Compensated OPAMP in CMOS 28nm

F. Ciciotti, M. De Matteis, and A. Baschirotto

PRIME Conference, June 2017

“The transfer function is obtained with the cascade of two Active-RC Rauch biquadratic cells. Each cell is based on a novel OPAMP optimized for very high frequency operation achieving a Unity Gain Bandwidth (UGBW) > 7GHz.”

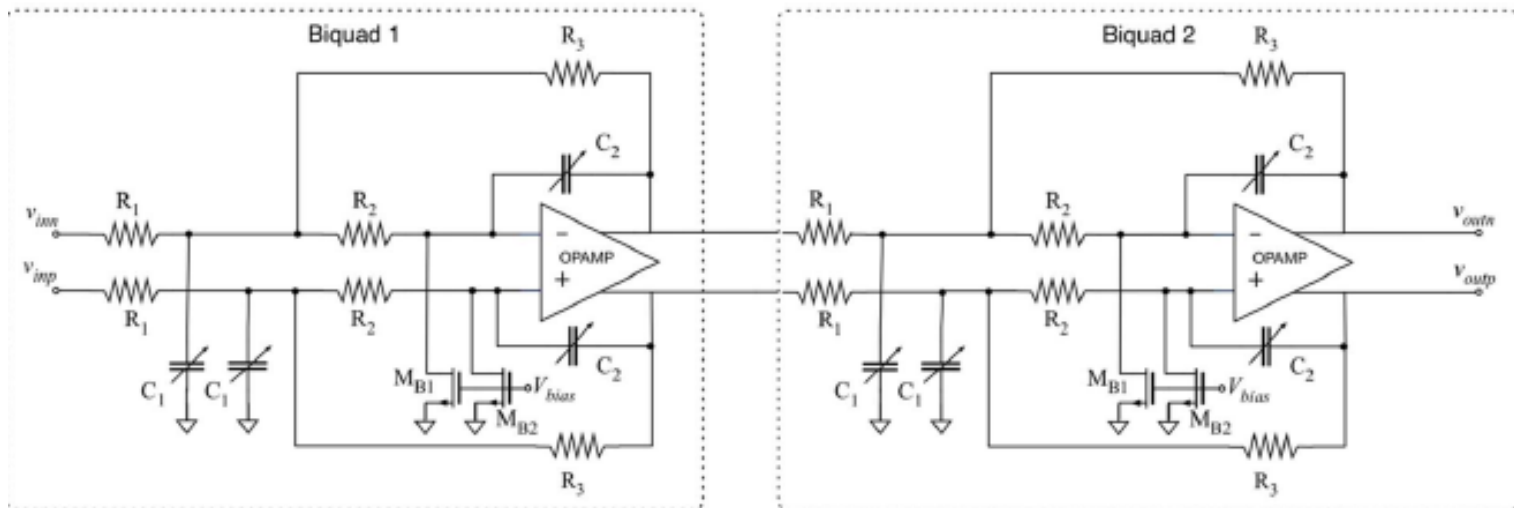


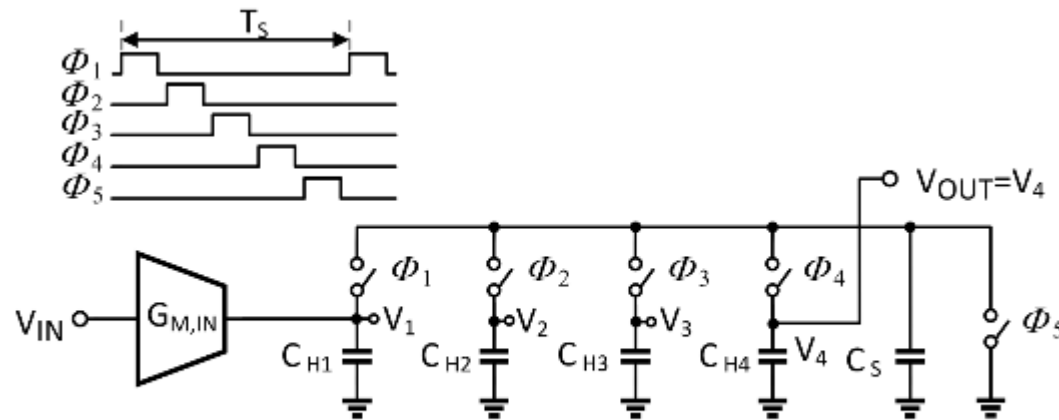
Fig. 1. Filter chain.

This is actually a bridged-T structure !



# SC Filter w/o Op Amp

M. Tohidian, I. Madadi, and R. B. Staszewski, "Analysis and design of a high-order discrete-time passive IIR low-pass filter," *IEEE J. Solid-State Circuits*, vl. 49, no. 11, pp. 2575–2587, Nov. 2014.



1. A 4<sup>th</sup>-order real-pole passive-SC LPF [2].

# SC Filter w/o Op Amp

S. Iida, "Filter circuit, integrated circuit, communication module, and communication apparatus," U.S. Patent 0 334 348 A1, Nov. 13, 2014.

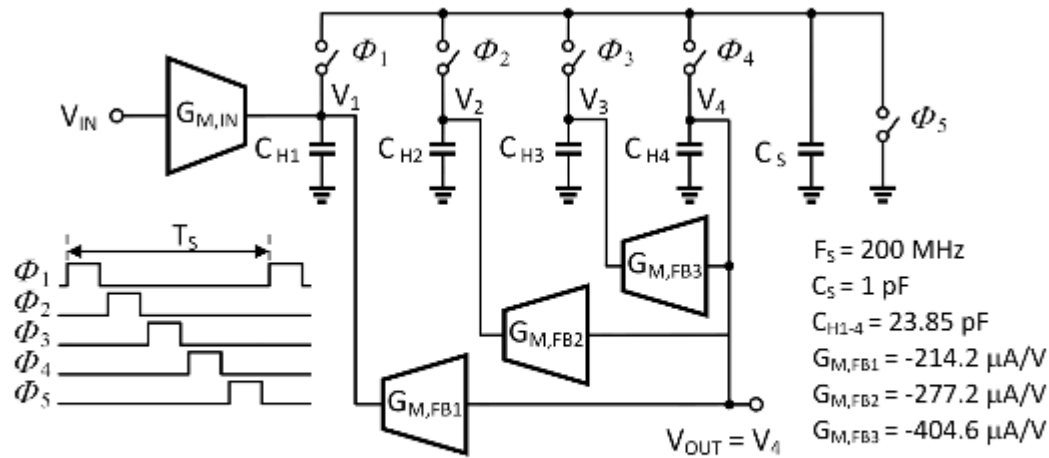


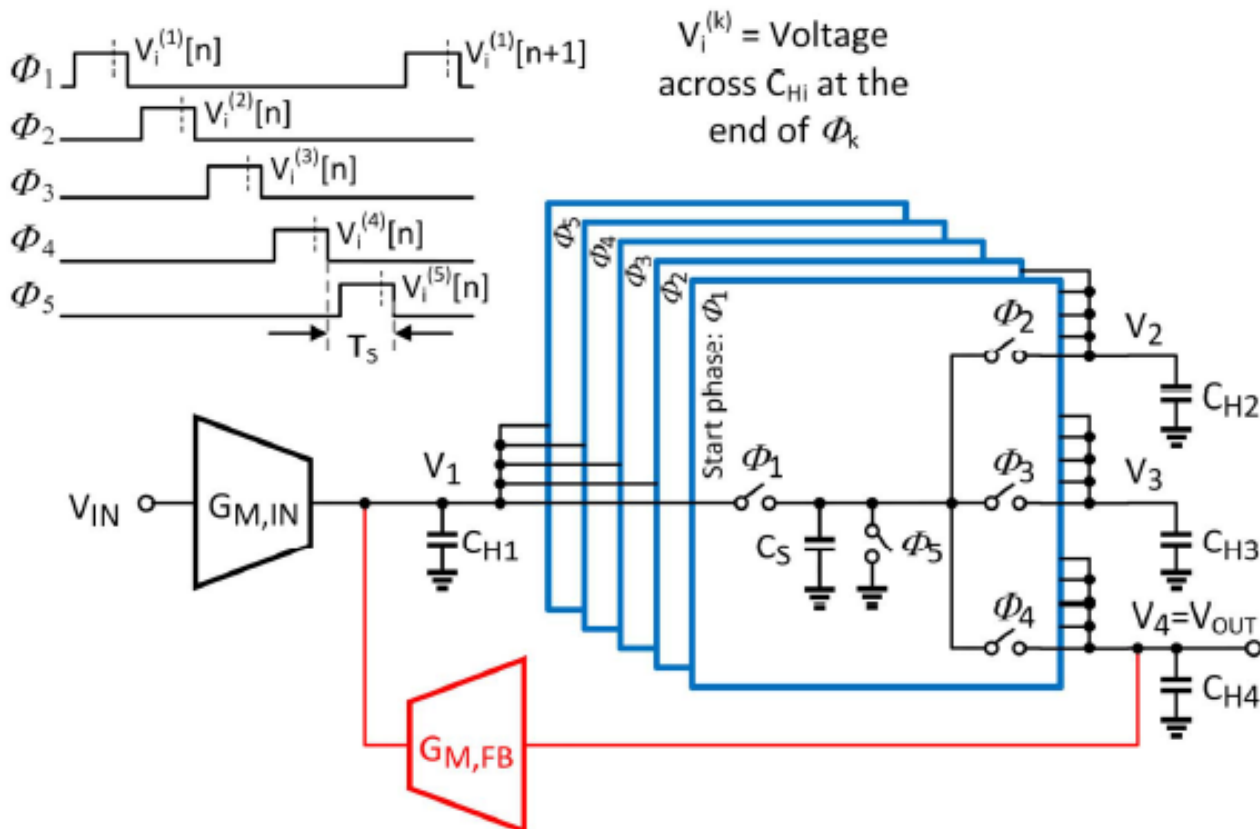
Fig. 3. A 4<sup>th</sup>-order complex-pole filter [21].

# SC Filter w/o Op Amp

## A 0.49–13.3 MHz Tunable Fourth-Order LPF with Complex Poles Achieving 28.7 dBm OIP3

Pedram Payandehnia<sup>1b</sup>, *Student Member, IEEE*, Hamidreza Maghami, *Student Member, IEEE*,  
Hossein Mirzaie<sup>1b</sup>, *Student Member, IEEE*, Manjunath Kareppagoudr, *Student Member, IEEE*,  
Siladitya Dey, *Student Member, IEEE*, Massoud Tohidian, *Member, IEEE*,  
and Gabor C. Temes, *Life Fellow, IEEE*

IEEE TRANSACTIONS ON CIRCUITS AND SYSTEMS–I: REGULAR PAPERS, VOL. 65, NO. 8, AUGUST 2018

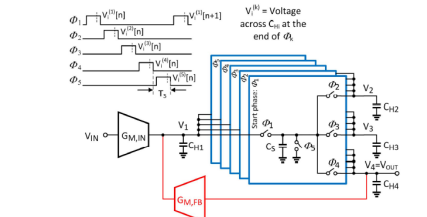


# SC Filter w/o Op Amp

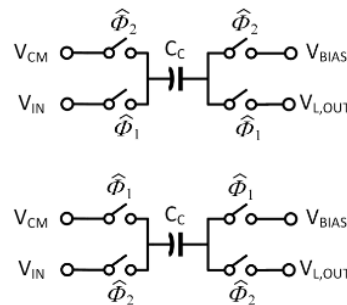
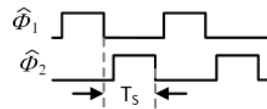
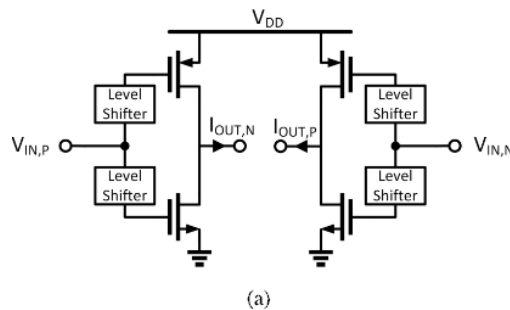
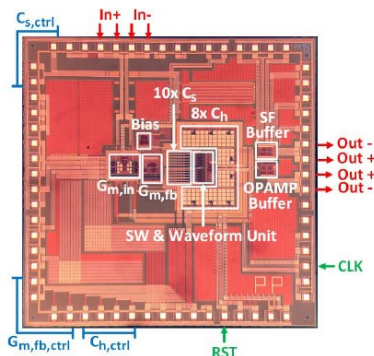
## A 0.49–13.3 MHz Tunable Fourth-Order LPF with Complex Poles Achieving 28.7 dBm OIP3

Pedram Payandehnia<sup>1b</sup>, *Student Member, IEEE*, Hamidreza Maghami, *Student Member, IEEE*,  
 Hossein Mirzaie<sup>1b</sup>, *Student Member, IEEE*, Manjunath Kareppagoudr, *Student Member, IEEE*,  
 Siladitya Dey, *Student Member, IEEE*, Massoud Tohidian, *Member, IEEE*,  
 and Gabor C. Temes, *Life Fellow, IEEE*

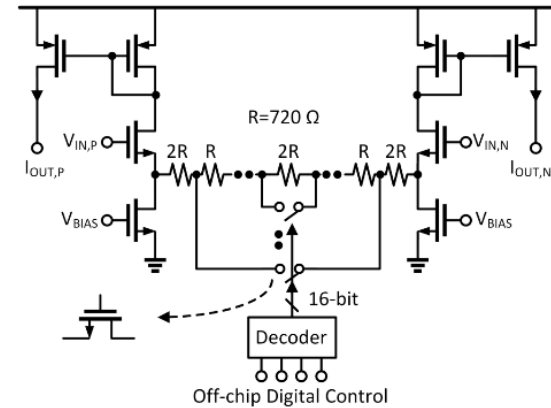
IEEE TRANSACTIONS ON CIRCUITS AND SYSTEMS—I: REGULAR PAPERS, VOL. 65, NO. 8, AUGUST 2018



Technology	180 nm
Order	4 <sup>th</sup>
VDD (V)	1.8
Power (mW)	4.3
3-dB BW (MHz)	0.49-13.3



Input OTA



Feedback OTA

Fig. 17. Chip micrograph of the proposed filter implemented in 1P4M 180 nm CMOS technology. Die size is 4 × 4 mm.

# A 20kHz~16MHz Programmable-Bandwidth 4<sup>th</sup> Order Active Filter using Gain-boosted Opamp with Negative Resistance in 65 nm CMOS

Jiye Lim, *Student Member, IEEE*, and Jintae Kim, *Senior Member, IEEE*

Accepted for TCAS II and pending publication Nov18

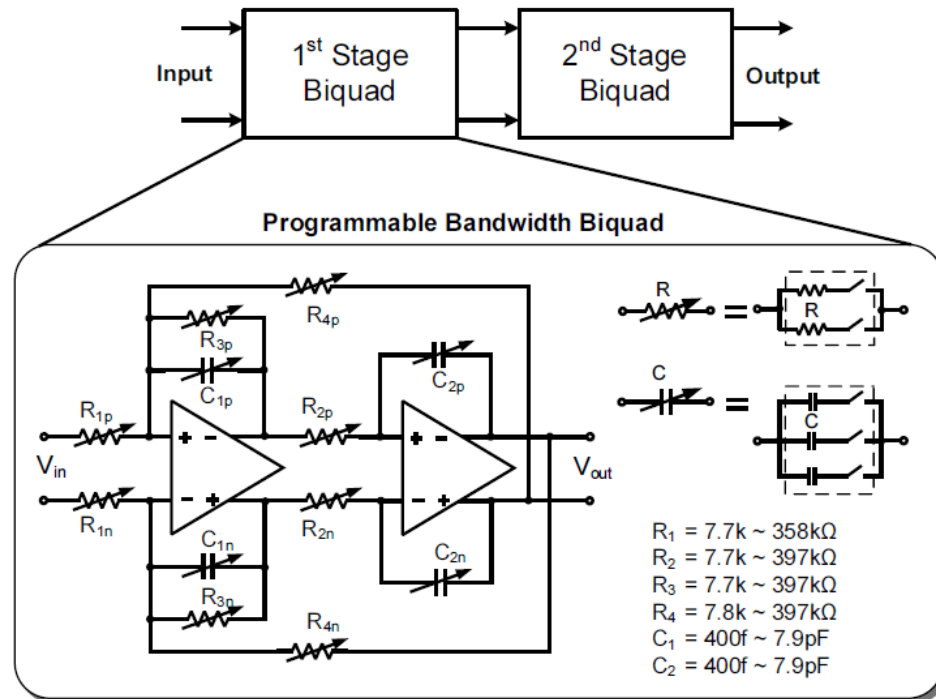


Fig. 1. A block diagram of 4th order programmable biquad filter.

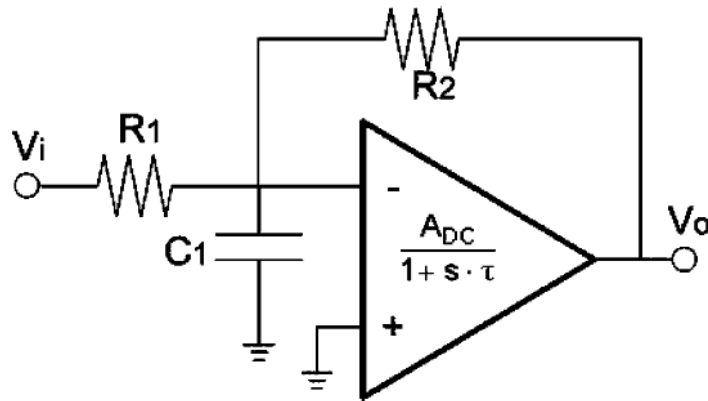
The prototype filter is fabricated in 65nm CMOS and occupies 0.098mm<sup>2</sup>. It features three programmable cutoff frequencies of 20kHz, 2MHz, and 16MHz

Power (mW)	19
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# A 4th-Order Active-G<sub>m</sub>-RC Reconfigurable (UMTS/WLAN) Filter

Stefano D'Amico, Vito Giannini, and Andrea Baschirotto

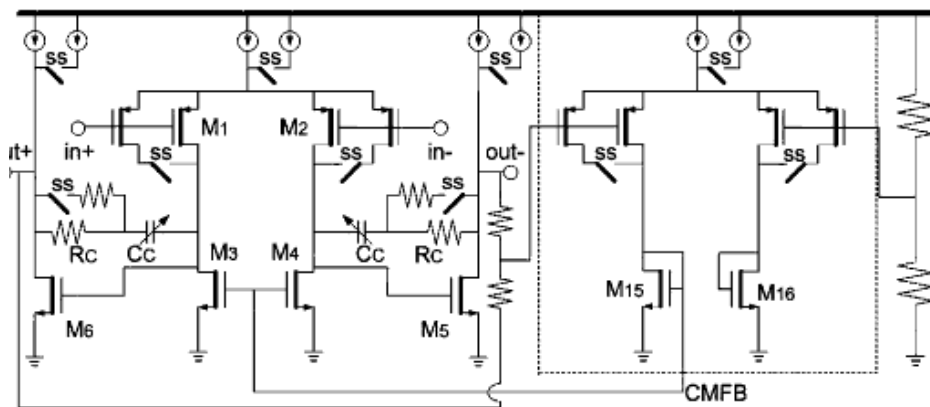
IEEE JOURNAL OF SOLID-STATE CIRCUITS, VOL. 41, NO. 7, JULY 2006



$$\left. \begin{aligned} V_{IN}G_1 + V_{OUT}G_2 &= V_X(G_1 + G_2 + sC) \\ V_{OUT} &= V_X \frac{-A_0}{1 + \tau s} \end{aligned} \right\}$$

$$\frac{V_{OUT}}{V_{IN}} = \frac{-\frac{A_0 G_1}{\tau C}}{s^2 + s \left[ \frac{G_1 + G_2}{C} + \frac{1}{\tau} \right] + \frac{G_1 + G_2(1 + A_0)}{\tau C}}$$

Active-G<sub>m</sub>-RC biquadratic cell.



Differential Amp

Realizes 4<sup>th</sup>-order filter

C1 and CC tunable,  
R1 and R2 switchable

Operates in 2MHz and  
20MHz ranges

# A 28.8-MHz 23-dBm-IIP3 3.2-mW Sallen-Key Fourth-Order Filter With Out-of-Band Zeros Cancellation

Marcello De Matteis, Federica Resta, Alessandra Pipino, Stefano D'Amico, and Andrea Baschirotto

TCAS II Dec 16

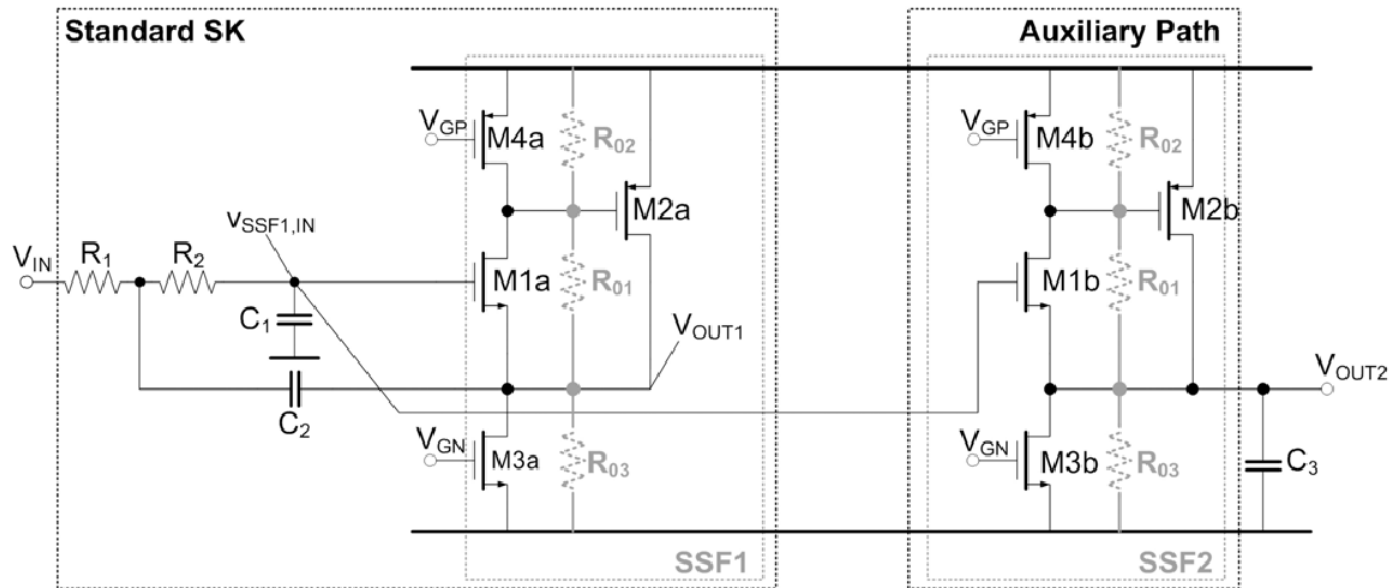


Fig. 1. SK single-ended generic scheme, with auxiliary path.

The total area occupancy is 0.12 mm<sup>2</sup>

3.2-mW power consumption

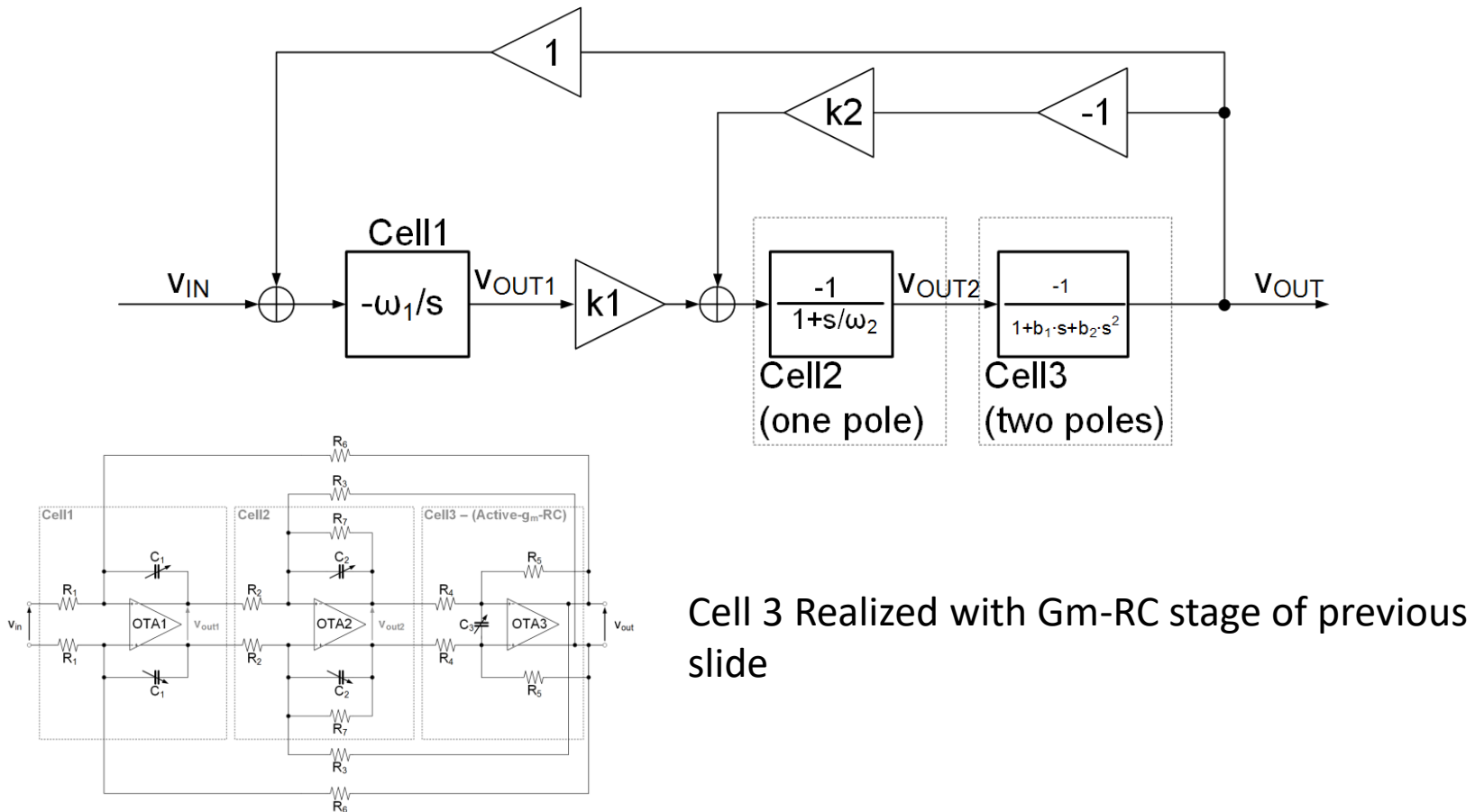
0.18 $\mu$  process

# A 63-dB DR 22.5-MHz 21.5-dBm IIP3 Fourth-Order FLFB Analog Filter

Marcello De Matteis, Alessandra Pipino, Federica Resta, Alessandro Pezzotta, Stefano D'Amico, and Andrea Baschirotto

IEEE JOURNAL OF SOLID-STATE CIRCUITS, VOL. 52, NO. 7, JULY 2017

Follow the Leader Feedback (a slight variant on the MLF approach)







Stay Safe and Stay Healthy !

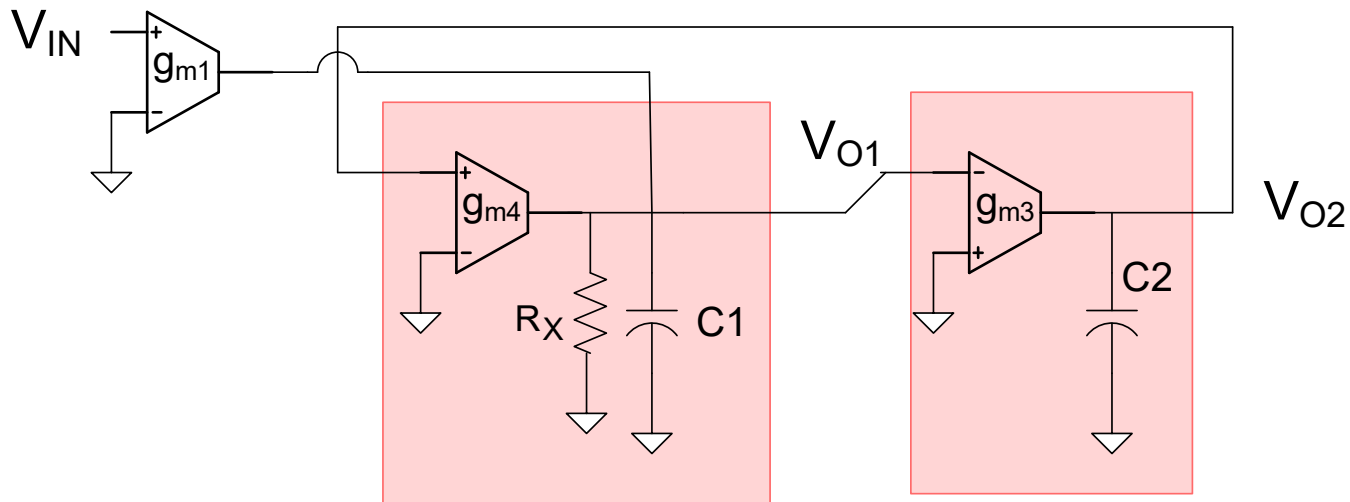
End of Lecture 42

# EE 508 Lecture 40

## Some Recent Filter Structures



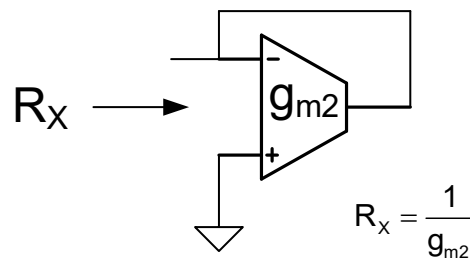
# Recall the basic two-integrator loop



$$\left. \begin{aligned} V_{01} s C_1 &= G_X V_{01} + g_{m1} V_{IN} + g_{m4} V_{02} \\ V_{02} s C_2 &= g_{m3} V_{01} \end{aligned} \right\}$$

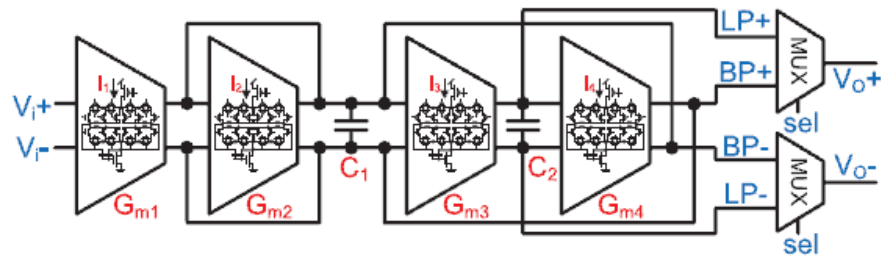
$$\frac{V_{01}}{V_{IN}} = \frac{s \frac{g_{m1}}{C_1}}{s^2 + s \frac{g_X}{C_1} + \frac{g_{m3} g_{m4}}{C_1 C_2}}$$

$$\frac{V_{02}}{V_{IN}} = \frac{\frac{g_{m3} g_{m1}}{C_1 C_2}}{s^2 + s \frac{g_X}{C_1} + \frac{g_{m3} g_{m4}}{C_1 C_2}}$$

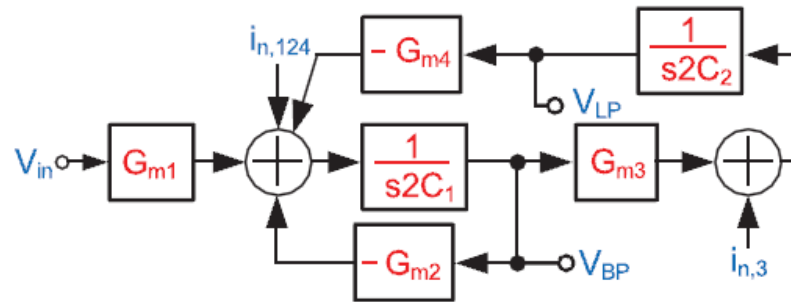


$$\frac{V_{01}}{V_{IN}} = \frac{s \frac{g_{m1}}{C_1}}{s^2 + s \frac{g_{m2}}{C_1} + \frac{g_{m3} g_{m4}}{C_1 C_2}}$$

$$\frac{V_{02}}{V_{IN}} = \frac{\frac{g_{m3} g_{m1}}{C_1 C_2}}{s^2 + s \frac{g_{m2}}{C_1} + \frac{g_{m3} g_{m4}}{C_1 C_2}}$$



(a)

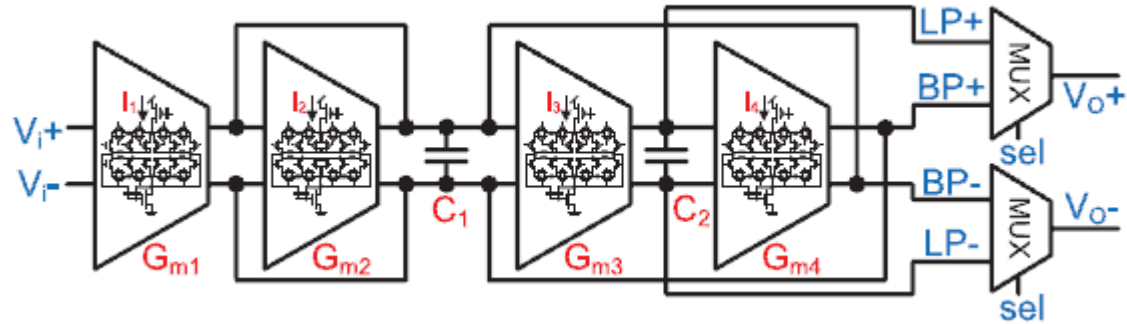


(b)

- This is a fully-differential implementation of the standard two-integrator loop
- MUX selects either LP or BP output

$$\frac{V_{01}}{V_{IN}} = \frac{s \frac{g_{m1}}{C_1}}{s^2 + s \frac{g_{m2}}{C_1} + \frac{g_{m3}g_{m4}}{C_1C_2}}$$

$$\frac{V_{02}}{V_{IN}} = \frac{\frac{g_{m3}g_{m1}}{C_1C_2}}{s^2 + s \frac{g_{m2}}{C_1} + \frac{g_{m3}g_{m4}}{C_1C_2}}$$



- This is a fully-differential implementation of the standard two-integrator loop
- MUX selects either LP or BP output

$$\frac{V_{01}}{V_{IN}} = \frac{s \frac{g_{m1}}{C_1}}{s^2 + s \frac{g_{m2}}{C_1} + \frac{g_{m3}g_{m4}}{C_1C_2}}$$

$$\frac{V_{02}}{V_{IN}} = \frac{\frac{g_{m3}g_{m1}}{C_1C_2}}{s^2 + s \frac{g_{m2}}{C_1} + \frac{g_{m3}g_{m4}}{C_1C_2}}$$

OTAs operate in weak inversion

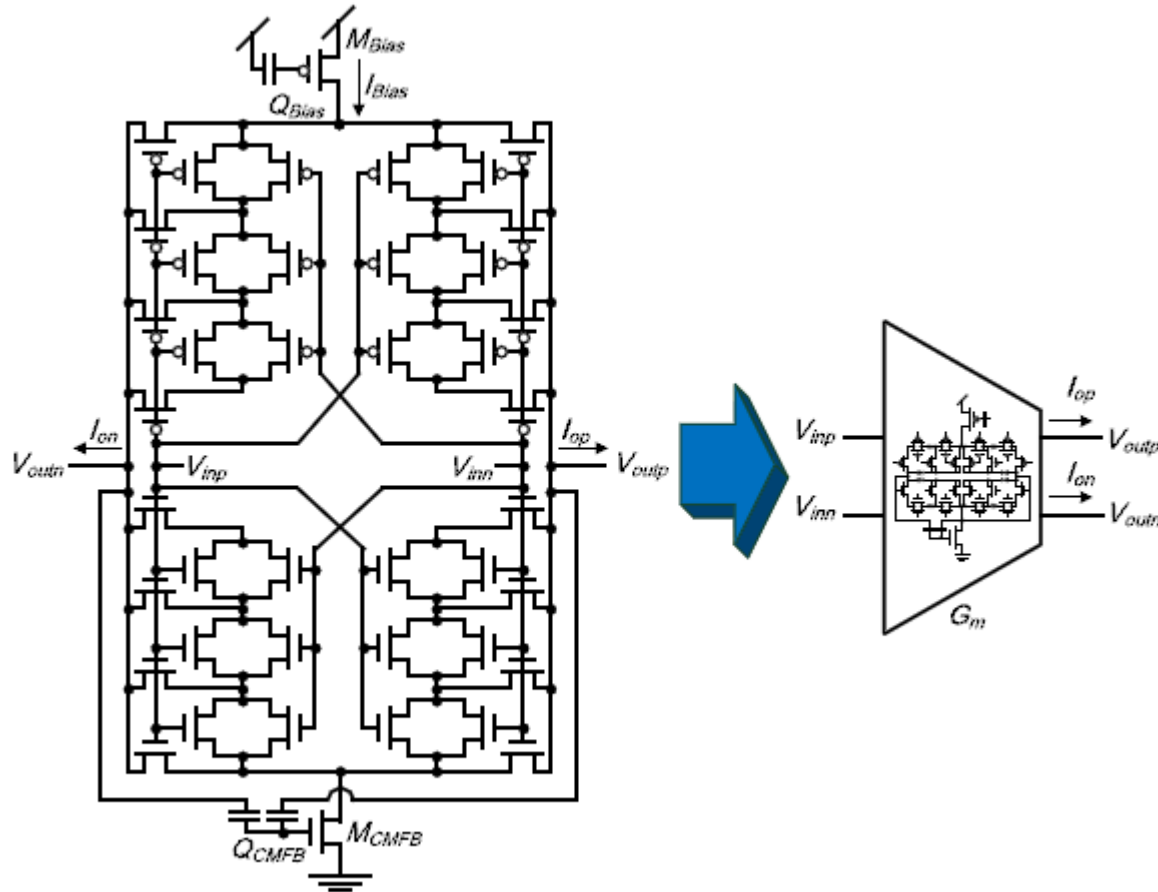
Adjust  $\omega_0$  by changing tail currents – claim in excess of 5 decades of adjustment

Target 2Hz to 20KHz though claim can go much lower (claim to 10mHz range) and higher

Bias current adjusted by changing charge on floating gate transistor

Each biquad requires 0.12mm<sup>2</sup> of die area in 350nm process

## Linearized OTA



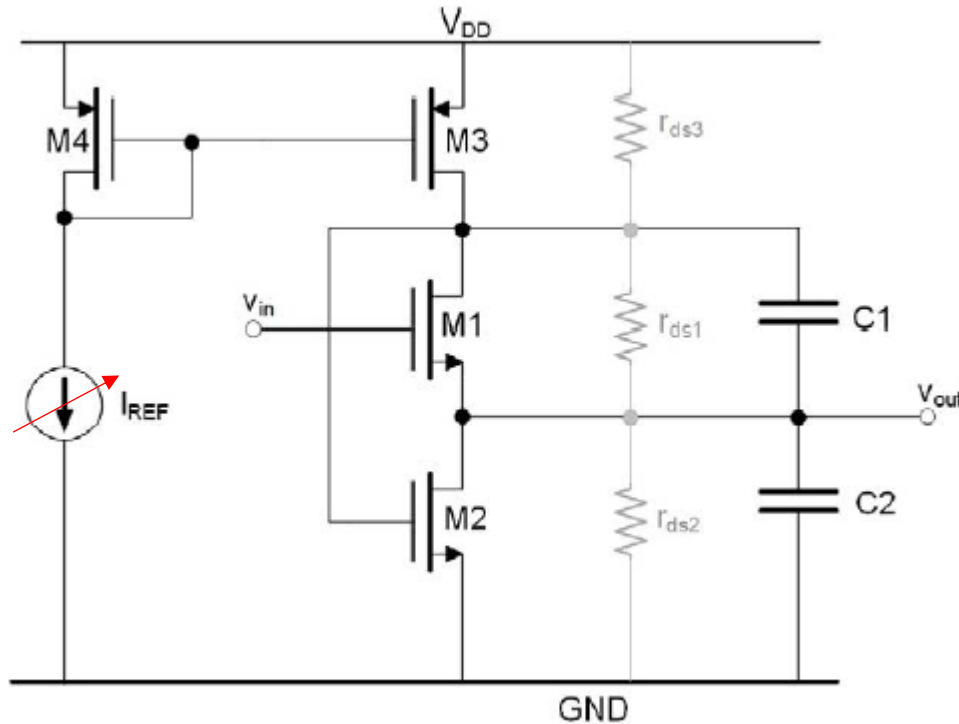
Used computer iteration to size devices in OTA  
Good linearity and low power dissipation claimed



# A 28nm-CMOS 100MHz 1mW 12dBm-IIP3 4<sup>th</sup>-order Flipped-Source-Follower Analog Filter

F. Fary<sup>1</sup>, M. De Matteis<sup>1</sup>, T. Vergine<sup>1,2</sup> and A. Baschirotto<sup>1</sup>

ESSCIRC 2018

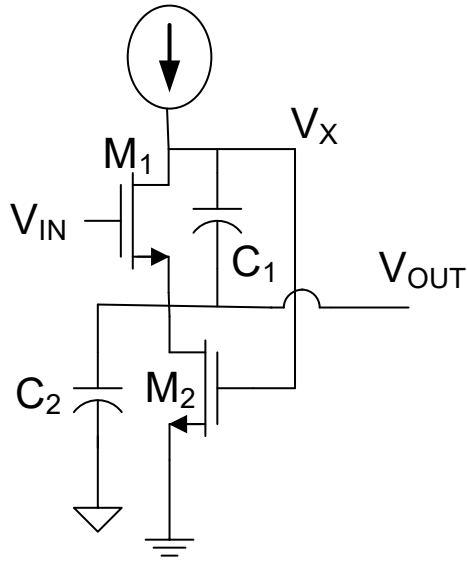


## Flipped-Source-Follower NMOS Biquadratic Cell

Table 1 – Filter Design Paramters

<i>Transfer Function</i>		<i>4<sup>th</sup>-Order Low-Pass</i>	
<b>dc-Gain</b>		0dB	
<b>Poles Frequency</b>		100 MHz	
<b>Cell A Q Factor</b>	1.306	<b>Cell B Q Factor</b>	0.5412
<b>Cell A <math>g_{m1} - g_{m2}</math></b>	1.8 mA/V	<b>Cell B <math>g_{m1} - g_{m3}</math></b>	1.8 mA/V
<b>Cell A - <math>C_{1a}</math></b>	4.8 pF	<b>Cell B - <math>C_{1b}</math></b>	1.99 pF
<b>Cell A - <math>C_{2a}</math></b>	1.75 pF	<b>Cell B - <math>C_{2b}</math></b>	3.98 pF

$A=0.026\text{mm}^2$  for 4th order BW filter in 28nm process P approx. 1mW



$$\left. \begin{aligned} V_{OUT} (sC_1 + sC_2) + g_{m2} V_{GS2} - g_{m1} V_{GS1} &= sC_1 V_{GS2} \\ V_{IN} &= V_{GS1} + V_{OUT} \\ V_{GS2} sC_1 + g_{m1} V_{GS1} &= V_{OUT} sC_1 \end{aligned} \right\}$$

$$\frac{V_{OUT}}{V_{IN}} = \frac{g_{m1} g_{m2}}{s^2 C_1 C_2 + s C_1 g_{m2} + g_{m1} g_{m2}}$$

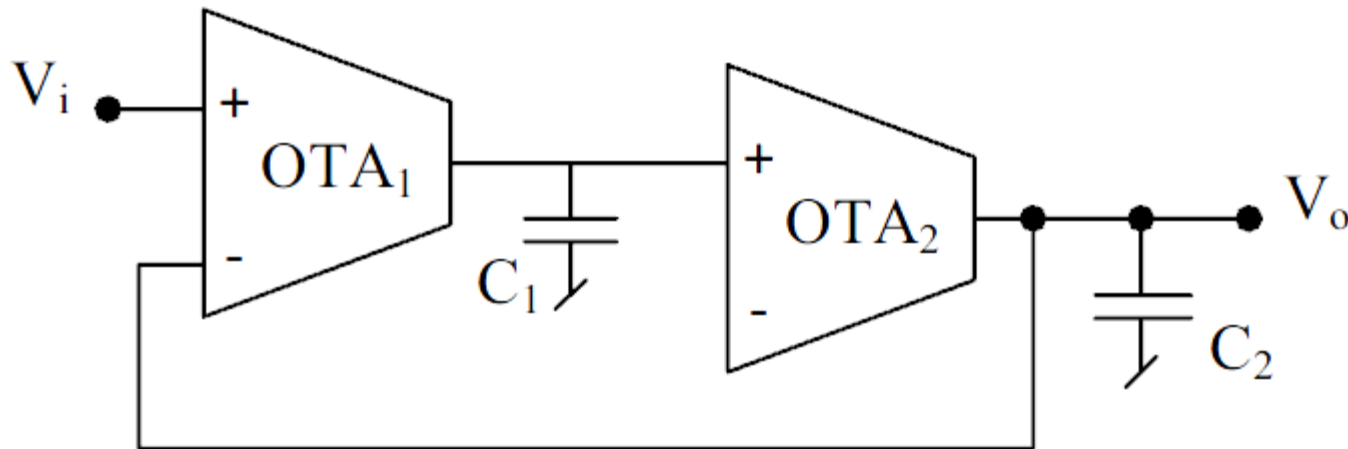
$$\omega_0 = \sqrt{\frac{g_{m1} g_{m2}}{C_1 C_2}}$$

$$Q = \sqrt{\frac{g_{m1} C_2}{g_{m2} C_1}}$$

# A New Method to Design Multi-Standard Analog Baseband Low-Pass Filter

Ersin Alaybeyoğlu<sup>1</sup>, Hakan Kuntman<sup>2</sup>

[2017 10th International Conference on Electrical and Electronics Engineering \(ELECO\)](#)



10MHz – 40MHz

Projected Area 0.02mm<sup>2</sup>

in 180nm proc

$$\frac{V_{LP}}{V_{in}} = \frac{g_{m1}g_{m2}}{s^2C_1C_2 + sC_1g_{m1} + g_{m1}g_{m2}}$$

$$\omega_0 = \sqrt{\frac{g_{m1}g_{m2}}{C_1C_2}}$$

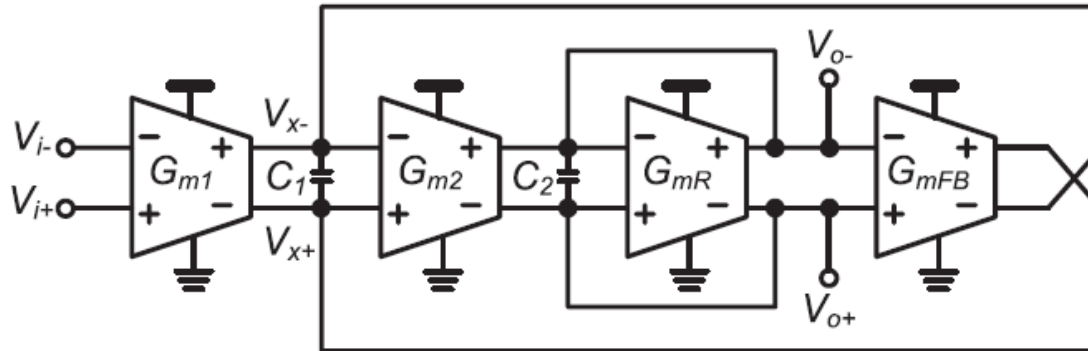
$$Q = \sqrt{\frac{C_2g_{m2}}{C_1g_{m1}}}$$

# Low-Power $G_m$ - $C$ Filter Employing Current-Reuse Differential Difference Amplifiers

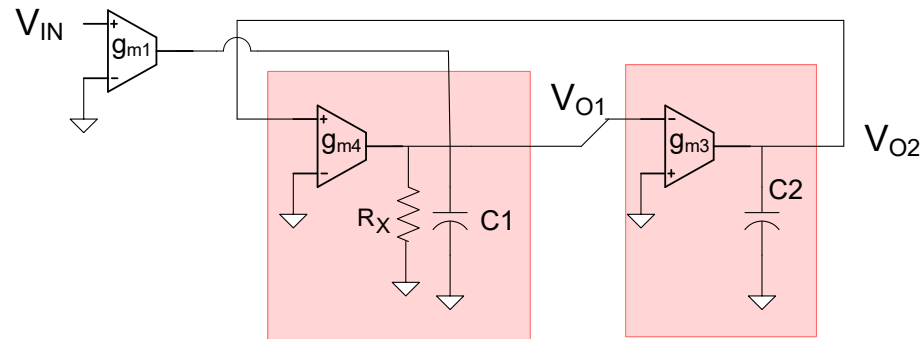
John S. Mincey, *Student Member, IEEE*, Carlos Briseno-Vidrios, *Student Member, IEEE*,  
Jose Silva-Martinez, *Fellow, IEEE*, and Christopher T. Rodenbeck, *Senior Member, IEEE*

IEEE TRANSACTIONS ON CIRCUITS AND SYSTEMS—II: EXPRESS BRIEFS, VOL. 64, NO. 6, JUNE 2017

## Typical Differential Implementation



## Typical Single-Ended Implementation



Require 4 OTAs

# Current-reuse Structures

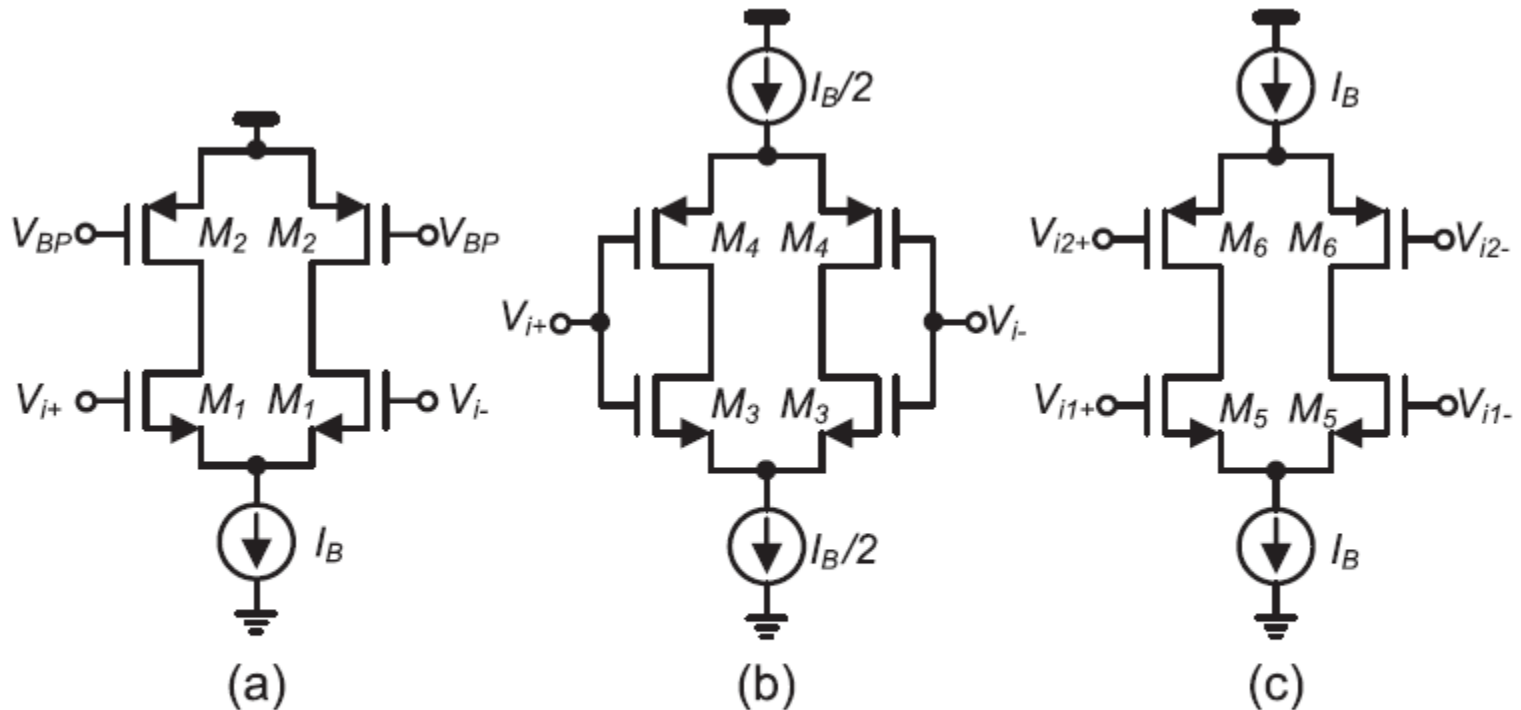


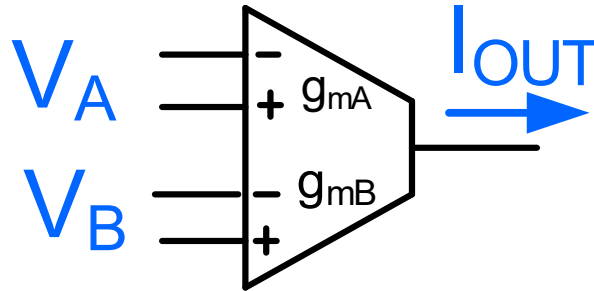
Fig. 3. (a) Conventional differential pair. (b) DDP using half the bias current. (c) Current-reuse DDA.

Dual Differential Pair: DDP

Dual Different Amplifier: DDA

Current Reuse offers potential for significant power reduction

# Current-reuse Structures

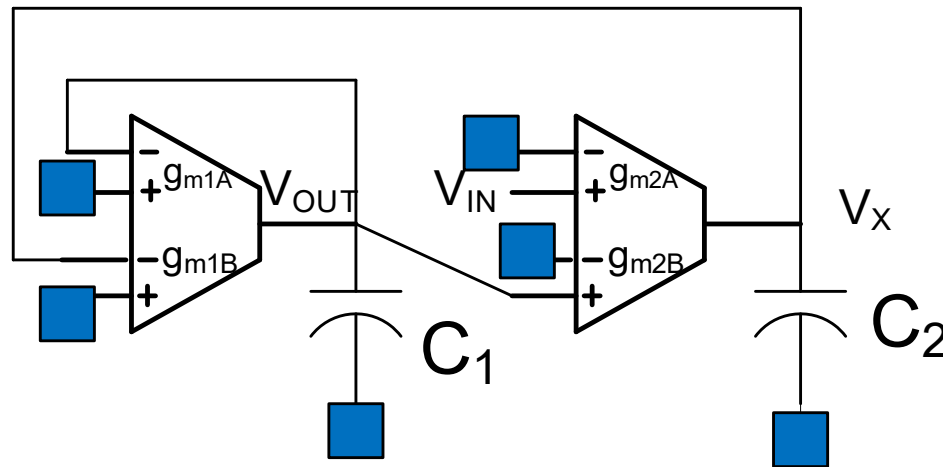


Dual input OTA

$$I_{OUT} = g_{mA} V_A + g_{mB} V_B$$

# Current-reuse Structures

Consider:



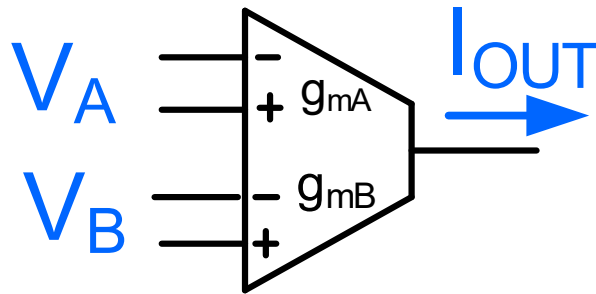
$$\left. \begin{aligned} V_{OUT} sC_1 &= -g_{m1A} V_{OUT} + g_{m1B} V_X \\ V_X sC_2 &= g_{m2B} V_{OUT} + g_{m2A} V_{IN} \end{aligned} \right\}$$

$$\frac{V_{OUT}}{V_{IN}} = - \frac{g_{m2A} g_{m1B}}{(s^2 C_1 C_2 + s C_2 g_{m1A} + g_{m1B} g_{m2B})}$$

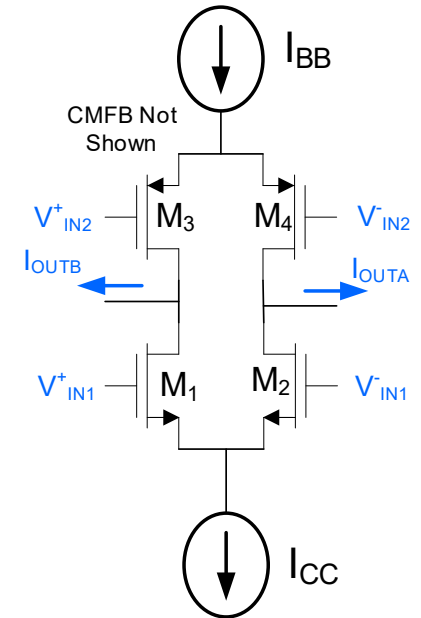
Realizes 2<sup>nd</sup>-order lowpass with just 2 OTAs

# Current-reuse Structures

Dual input OTA



$$I_{OUT} = g_{mA} V_A + g_{mB} V_B$$



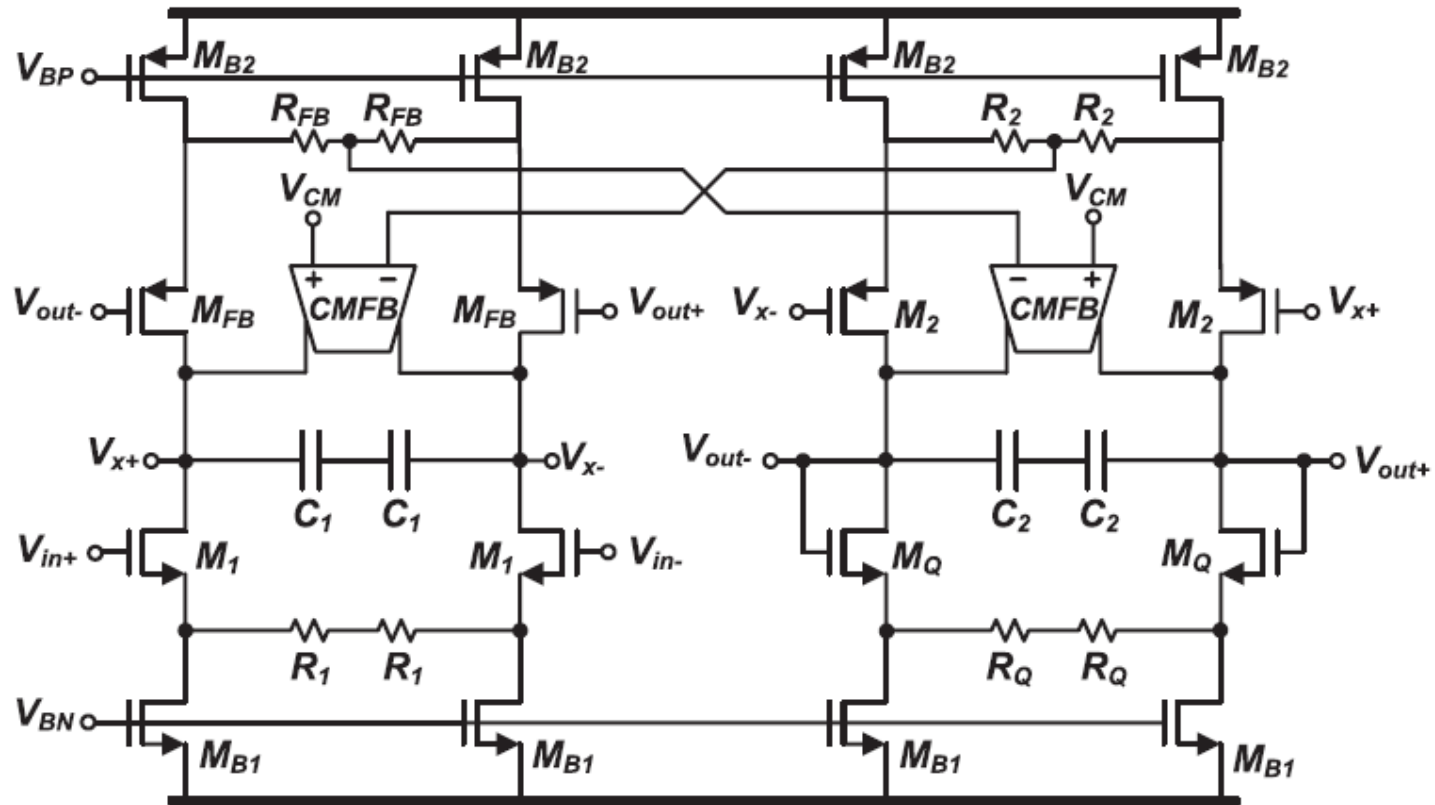
$$I_{OUTA} = g_{m2} V_{IN1}^- + g_{m4} V_{IN2}^-$$

$$I_{OUTB} = g_{m1} V_{IN1}^+ + g_{m3} V_{IN2}^+$$



# Current-reuse Structures

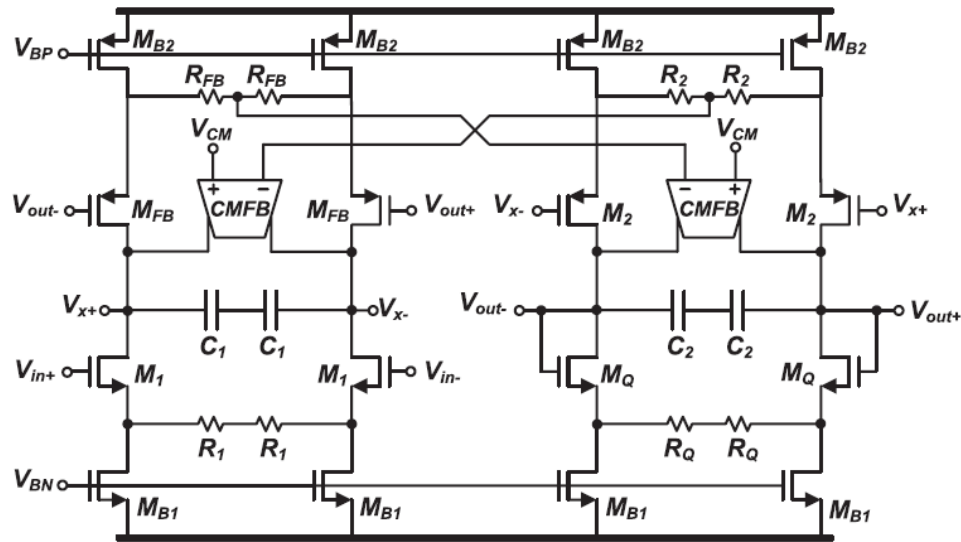
Dual input OTA



2<sup>nd</sup> Order Lowpass Biquad using Current-reuse OTA

# Current-reuse Structures

## Dual input OTA



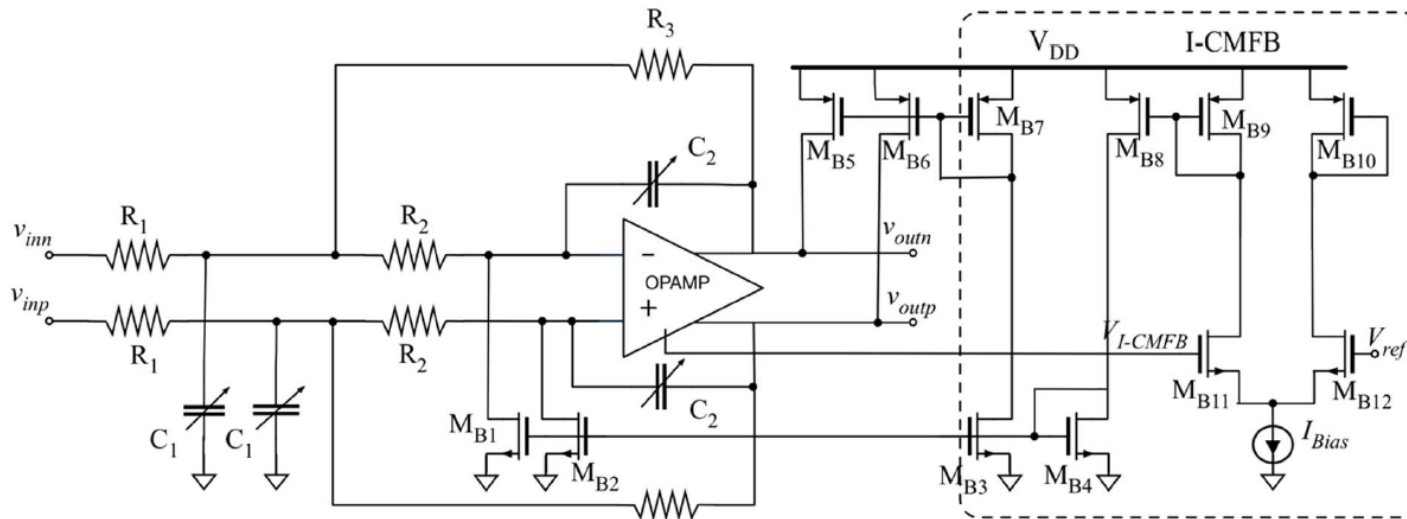
Sixth-order Butterworth  $G_m$ -C filter was fabricated

- 180-nm CMOS process
- total chip area of 0.21 mm<sup>2</sup>
- 65MHz Band Edge
- 1.3mW/pole

# A 0.9V 75MHz 2.8mW 4<sup>th</sup>-Order Analog Filter in CMOS-Bulk 28nm Technology

F. Ciciotti, M. De Matteis, and A. Baschiroto

ISCAS 2018



$$H(s) \cong -\frac{R_3}{R_1} \cdot \frac{1}{1 + sC_2 \left( R_2 + R_3 + R_2 \frac{R_3}{R_1} \right) + s^2 C_1 C_2 R_2 R_3}$$

$$\omega_0 = \sqrt{\frac{1}{R_2 R_3 C_1 C_2}} \quad Q = \sqrt{\frac{C_1}{C_2}} \frac{\sqrt{R_2 R_3}}{R_2 + R_3 + \frac{R_2 R_3}{R_1}}$$

# A 0.9V 75MHz 2.8mW 4<sup>th</sup>-Order Analog Filter in CMOS-Bulk 28nm Technology

F. Ciciotti, M. De Matteis, and A. Baschirotto

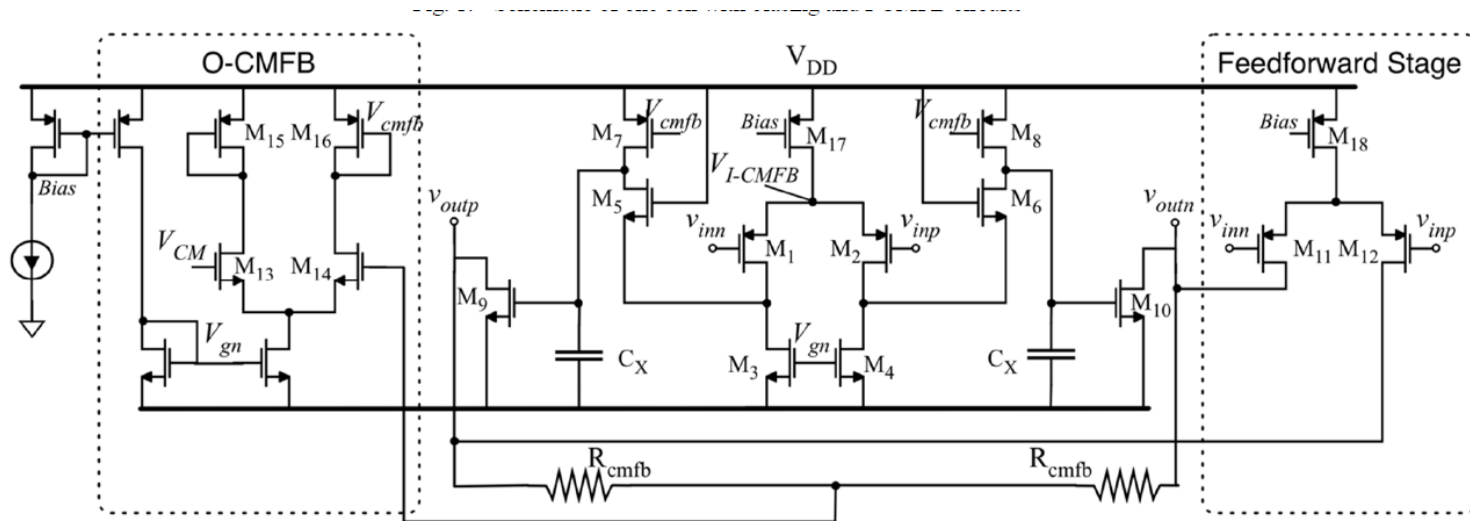


Fig. 2. Op Amp with feedforward compensation and O-CMFB circuit

CMOS 28nm process

4-bit capacitor arrays are used for frequency response programmability Filter covers the 40–105MHz range

0.7mW/pole

Area = 0.08mm<sup>2</sup>



Stay Safe and Stay Healthy !

End of Lecture 40