

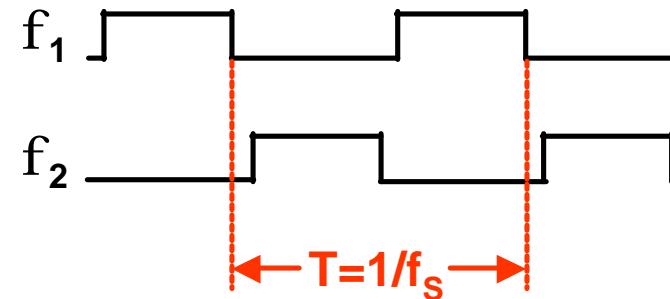
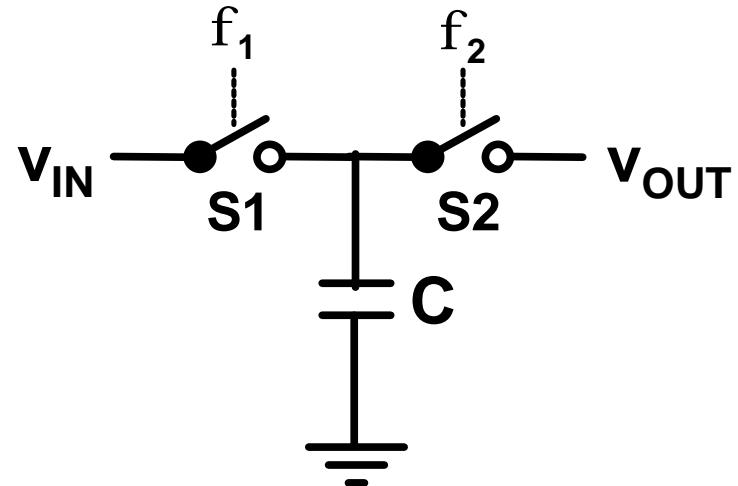
Switched-Capacitor Filters

- “Analog” sampled-data filters:
 - Continuous amplitude
 - Quantized time
- Applications:
 - Oversampled A/D and D/A converters
 - Analog front-ends (CDS, etc)
 - Standalone filters
 - E.g. National Semiconductor LMF100
 - Replaced by ADC + DSP in many cases



Switched-Capacitor Resistor

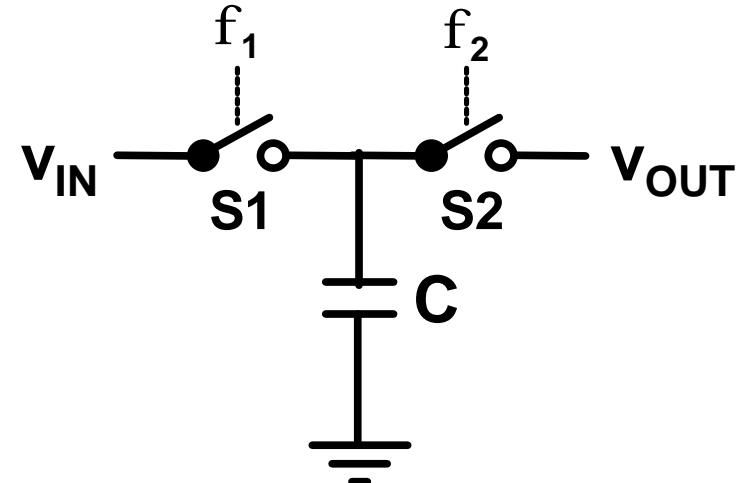
- Capacitor C is the “switched capacitor”
- Non-overlapping clocks ϕ_1 and ϕ_2 control switches S1 and S2, respectively
- v_{IN} is sampled at the falling edge of ϕ_1
 - Sampling frequency f_s
- Why is this a resistor?



Switched-Capacitor Resistors

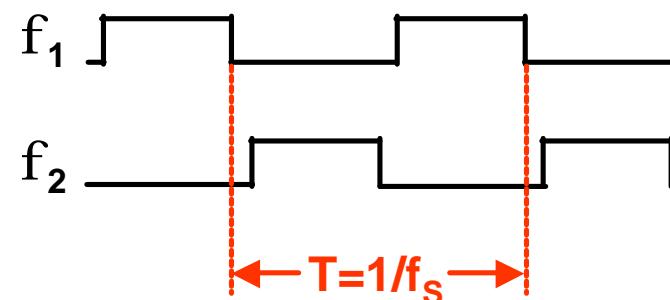
- The charge transferred from v_{IN} to v_{OUT} each sample period is:

$$Q = C(v_{IN} - v_{OUT})$$



- The average current flowing from v_{IN} to v_{OUT} is:

$$i = f_s C(v_{IN} - v_{OUT})$$



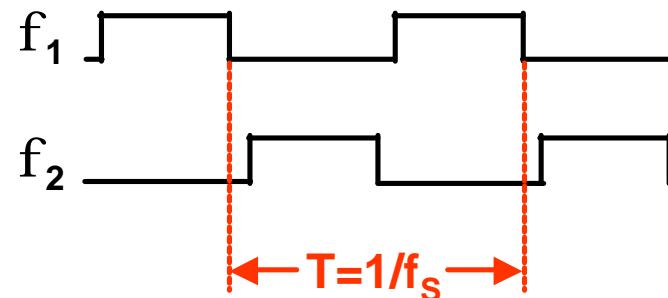
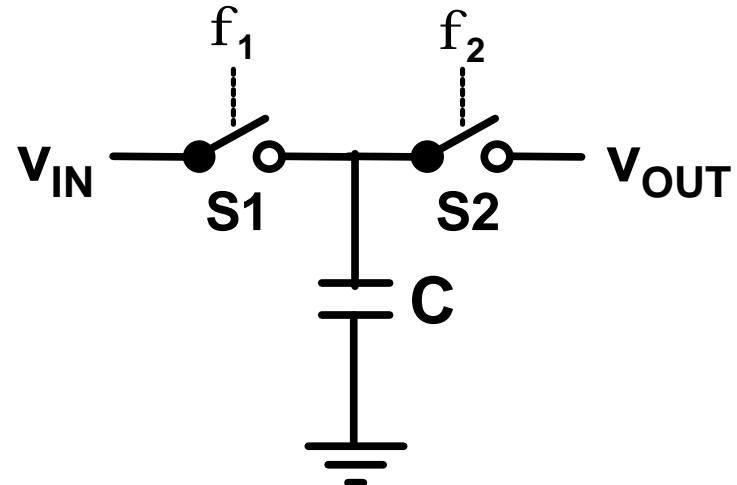
Switched-Capacitor Resistors

$$i = f_s C(v_{IN} - v_{OUT})$$

With the current through the switched capacitor resistor proportional to the voltage across it, the equivalent “switched capacitor resistance” is:

$$R_{EQ} = \frac{1}{f_s C}$$

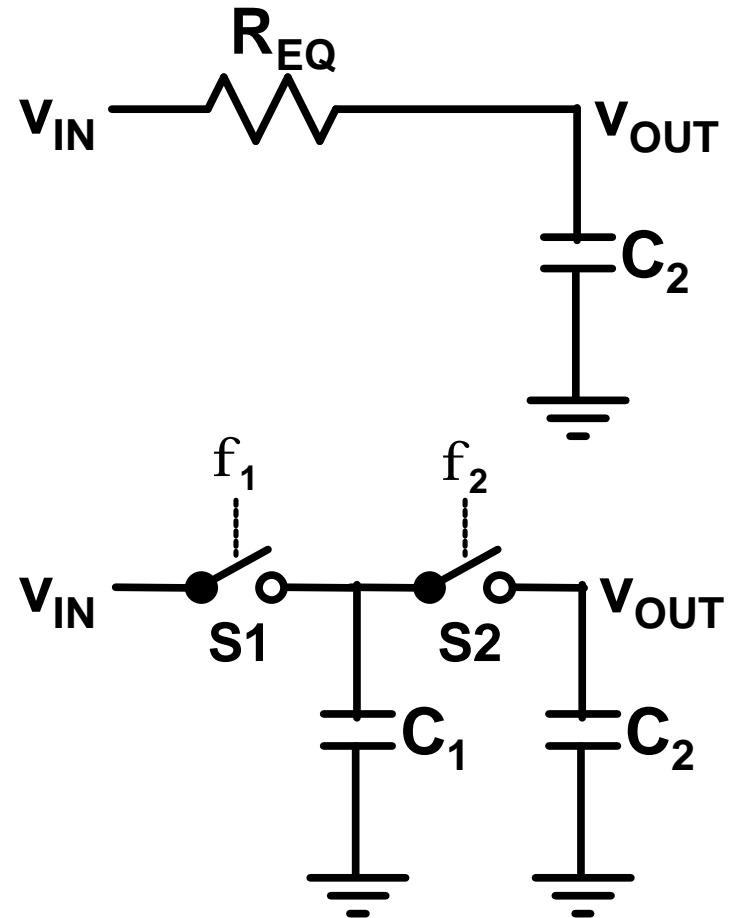
Of course this current flows in “bursts”—think of “big electrons”.



Switched-Capacitor Filter

- Let's build an "SC" filter ...
- We'll start with a simple RC LPF
- Replace the physical resistor by an equivalent SC resistor
- 3-dB bandwidth:

$$w_0 = \frac{1}{R_{EQ}C_2} = f_s \frac{C_1}{C_2}$$



Switched-Capacitor Filters

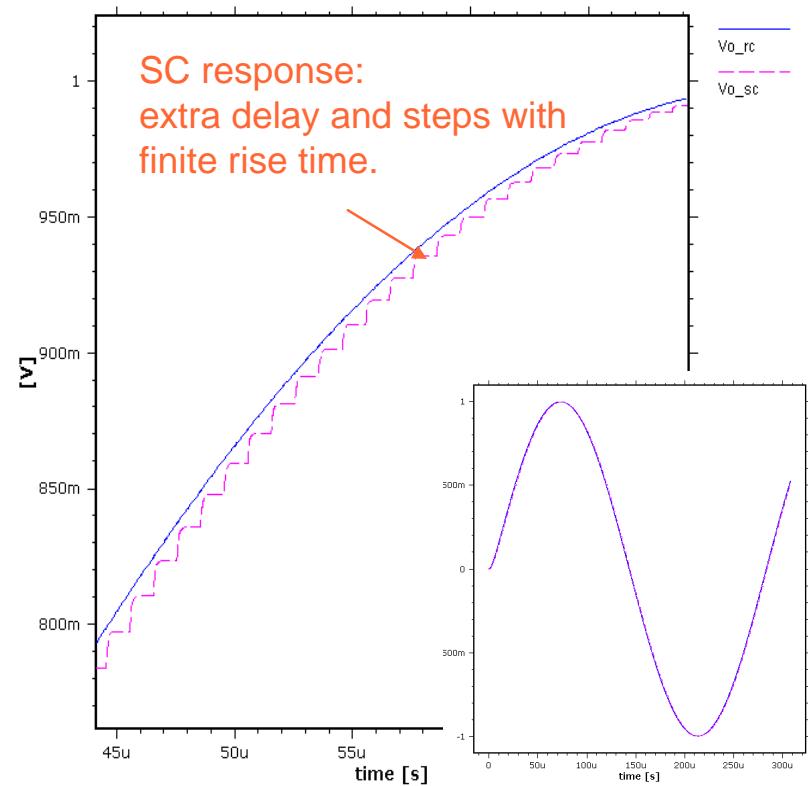
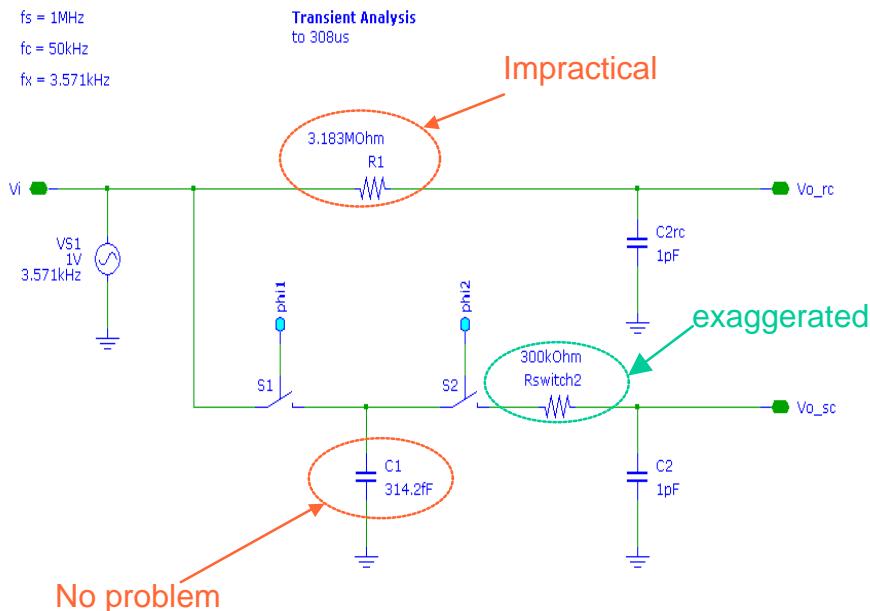
- In SCFs, all critical frequencies track the sampling frequency
 - Crystal oscillators for f_s are stable to $\sim 10\text{ppm}/^\circ\text{C}$
 - RC products used in active-RC filters can be tuned, but RCs in active-RC filters don't track together nearly as well
- Capacitor ratios in monolithic filters are perfectly stable over time and temperature
 - Capacitor ratios can't be trimmed easily
 - The trick is to achieve initial ratio accuracies of $\sim 1000\text{ppm}$ out of double-poly CMOS processes



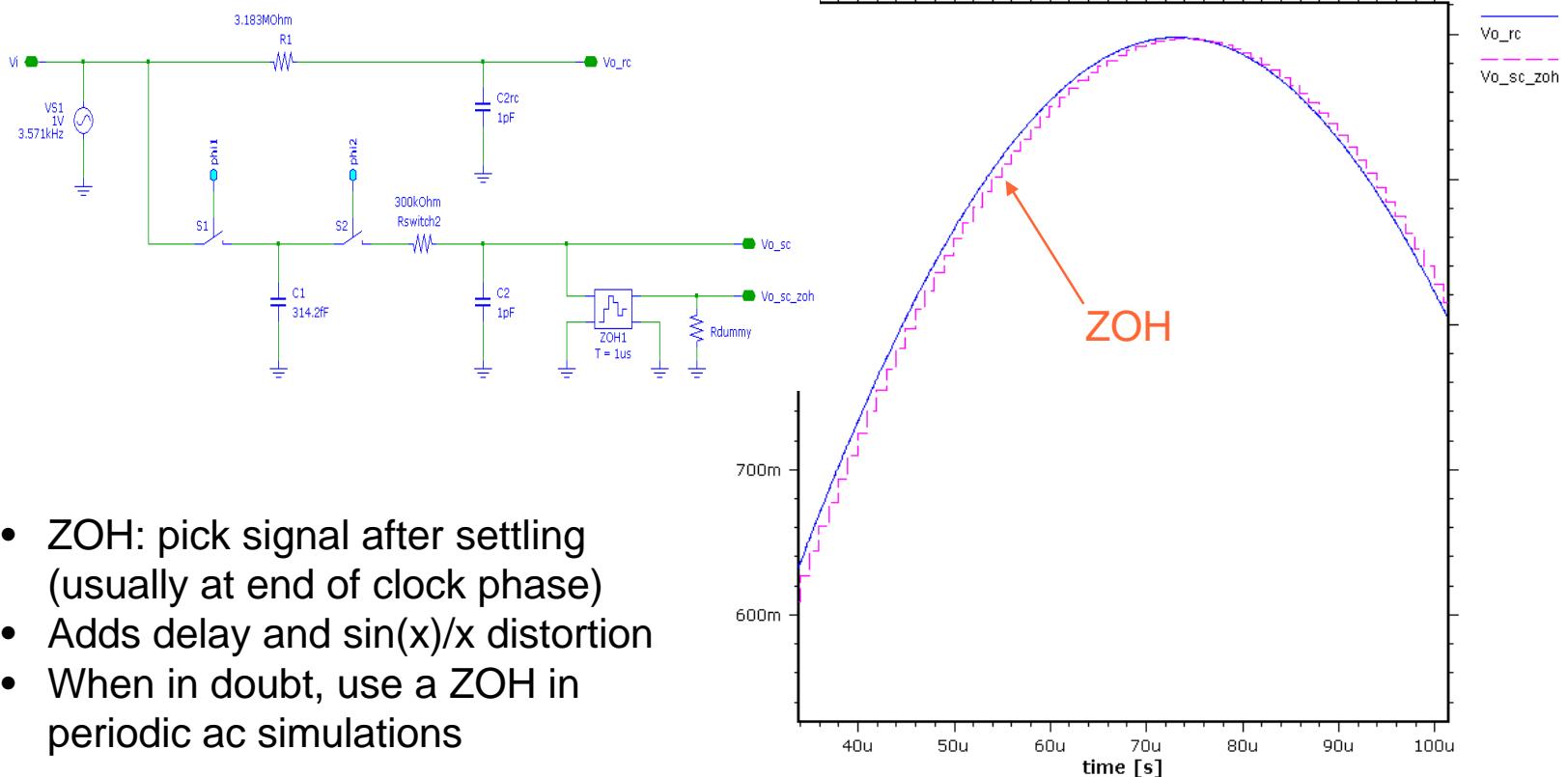
Transient Analysis

1st Order RC / SC LPF

$f_s = 1\text{MHz}$
 $f_c = 50\text{kHz}$
 $f_x = 3.571\text{kHz}$



Transient Analysis



- ZOH: pick signal after settling (usually at end of clock phase)
- Adds delay and $\sin(x)/x$ distortion
- When in doubt, use a ZOH in periodic ac simulations

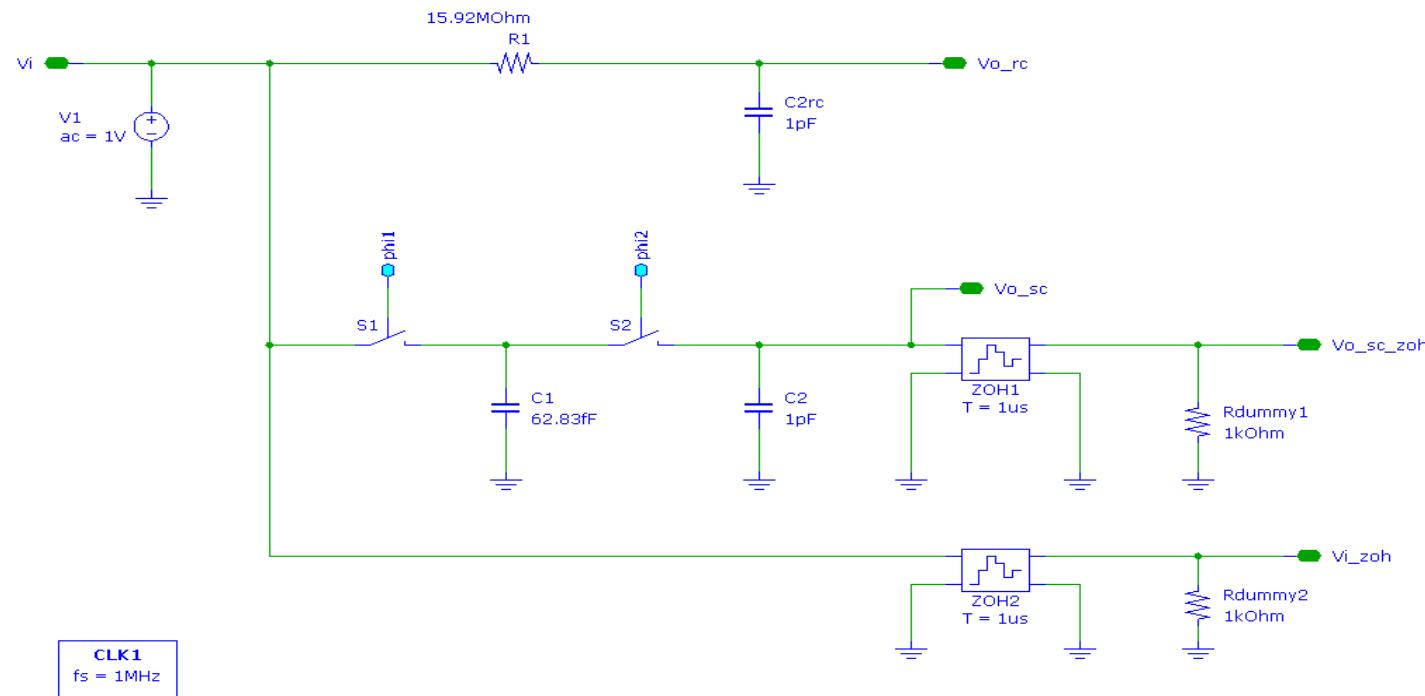
Periodic AC Analysis

1st Order RC / SC LPF

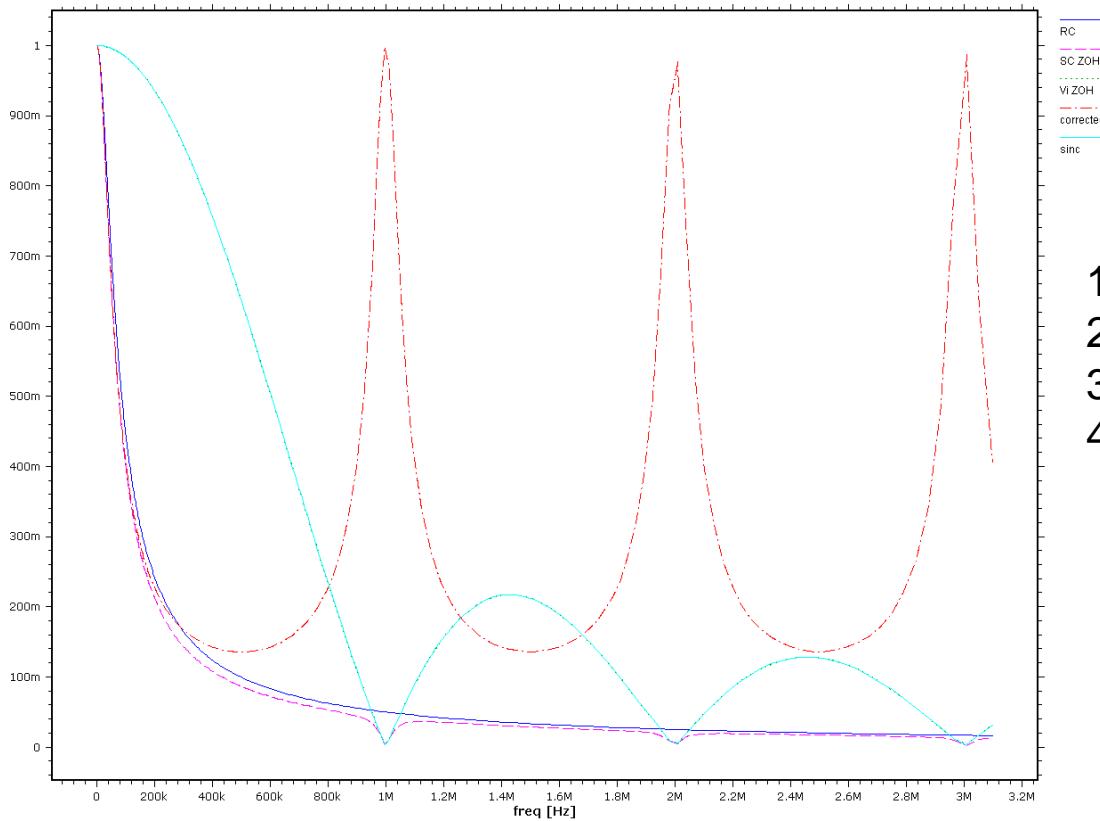
$f_s = 1\text{MHz}$
 $f_c = 50\text{kHz}$
 $f_x = 3.571\text{kHz}$

Periodic AC Analysis PAC1
log sweep from 1 to 3.1M (1001 steps)

Netlist
ahdl_include "zoh.def"



Magnitude Response



1. RC filter output
2. SC output after ZOH
3. Input after ZOH
4. Corrected output
 - (2) over (3)
 - periodic with f_s
 - Identical to RC for $f \ll f_s/2$

Periodic AC Analysis

- SPICE frequency analysis
 - ac linear, **time-invariant** circuits
 - pac linear, **time-variant** circuits
- SpectreRF statements

```
V1 ( Vi 0 ) vsource type=dc dc=0 mag=1 pacmag=1
PSS1 pss period=1u errpreset=conservative
PAC1 pac start=1 stop=1M lin=1001
```
- Output
 - Divide results by $\text{sinc}(f/f_s)$ to correct for ZOH distortion



Spectre Circuit File

```
rc_pac
simulator lang=spectre
ahdl_include "zoh.def"

S1 ( Vi c1 phi1 0 ) relay ropen=100G rclosed=1 vt1=-500m vt2=500m
S2 ( c1 Vo_sc phi2 0 ) relay ropen=100G rclosed=1 vt1=-500m vt2=500m
C1 ( c1 0 ) capacitor c=314.159f
C2 ( Vo_sc 0 ) capacitor c=1p
R1 ( Vi Vo_rc ) resistor r=3.1831M
C2rc ( Vo_rc 0 ) capacitor c=1p
CLK1_Vphi1 ( phi1 0 ) vsource type=pulse val0=-1 vall=1 period=1u
width=450n delay=50n rise=10n fall=10n
CLK1_Vphi2 ( phi2 0 ) vsource type=pulse val0=-1 vall=1 period=1u
width=450n delay=550n rise=10n fall=10n
V1 ( Vi 0 ) vsource type=dc dc=0 mag=1 pacmag=1
PSS1 pss period=1u errpreset=conservative
PAC1 pac start=1 stop=3.1M log=1001
ZOH1 ( Vo_sc_zoh 0 Vo_sc 0 ) zoh period=1u delay=500n aperture=1n tc=10p
ZOH2 ( Vi_zoh 0 Vi 0 ) zoh period=1u delay=0 aperture=1n tc=10p
```



ZOH Circuit File

```
// Copy from the SpectreRF Primer

module zoh (Pout, Nout, Pin, Nin) (period, delay,
aperture, tc)

node [V,I] Pin, Nin, Pout, Nout;
parameter real period=1 from (0:inf);
parameter real delay=0 from [0:inf];
parameter real aperture=1/100 from (0:inf);
parameter real tc=1/500 from (0:inf);
{
integer n; real start, stop;
node [V,I] hold;
analog {
    // determine the point when aperture begins
    n = ($time() - delay + aperture) / period +
    0.5;
    start = n*period + delay - aperture;
    $break_point(start);

    // determine the time when aperture ends
    n = ($time() - delay) / period + 0.5;
    stop = n*period + delay;
    $break_point(stop);
}

// Implement switch with effective series
// resistance of 1 Ohm
if ( ($time() > start) && ($time() <= stop))
    I(hold) <- V(hold) - V(Pin, Nin);
else
    I(hold) <- 1.0e-12 * (V(hold) - V(Pin, Nin));

// Implement capacitor with an effective
// capacitance of tc
I(hold) <- tc * dot(V(hold));

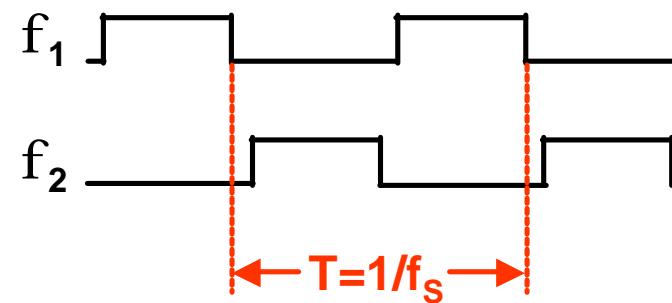
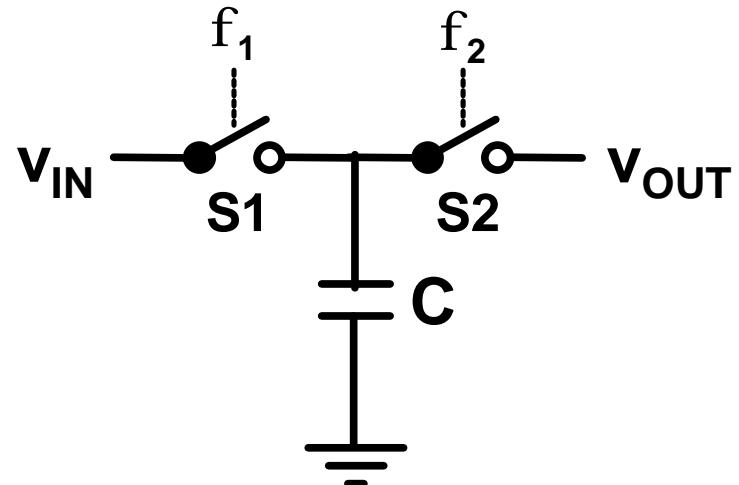
// Buffer output
V(Pout, Nout) <- V(hold);

// Control time step tightly during
// aperture and loosely otherwise
if (($time() >= start) && ($time() <= stop))
    $bound_step(tc);
else
    $bound_step(period/5);
}
```



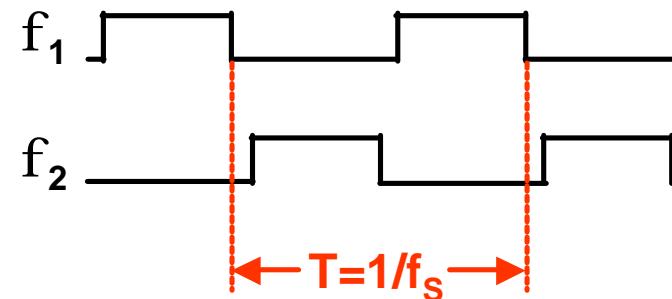
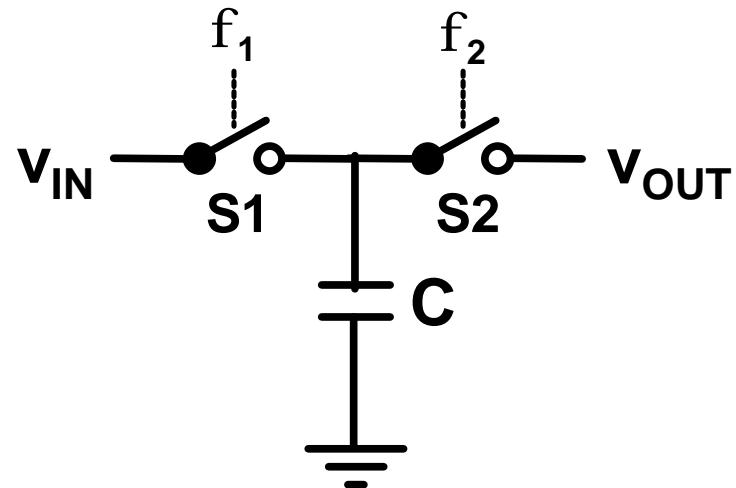
Switched-Capacitor Noise

- The resistance of switch S_1 produces a noise voltage on C with variance kT/C
- The corresponding noise charge is $Q^2=C^2V^2=kTC$
- This charge is sampled when S_1 opens



Switched-Capacitor Noise

- The resistance of switch S2 contributes to an uncorrelated noise charge on C at the end of ϕ_2
- The mean-squared noise charge transferred from v_{IN} to v_{OUT} each sample period is $Q^2=2kTC$



Switched-Capacitor Noise

- The mean-squared noise current due to S1 and S2's kT/C noise is :

$$\overline{i^2} = (Qf_s)^2 = 2k_B T C f_s^2$$

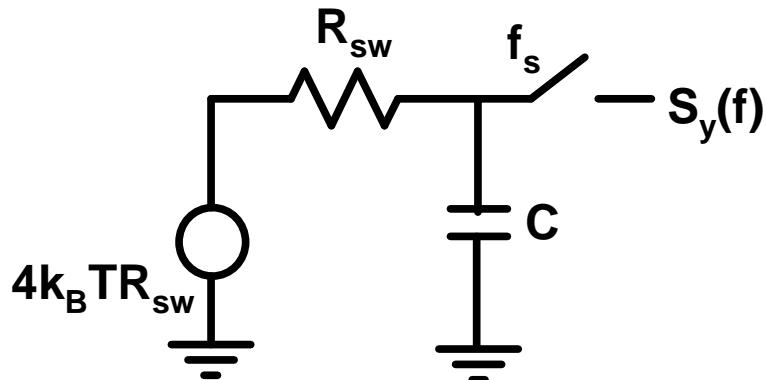
- This noise is approximately white (see next slide) and distributed between 0 and $f_s/2$ (noise spectra are single sided by convention)
The spectral density of the noise is:

$$\frac{\overline{i^2}}{\Delta f} = \frac{2k_B T C f_s^2}{f_s/2} = 4k_B T C f_s = \frac{4k_B T}{R_{EQ}} \quad \text{using} \quad R_{EQ} = \frac{1}{f_s C}$$

- The noise from an SC resistor equals the noise from a physical resistor with the same value!**



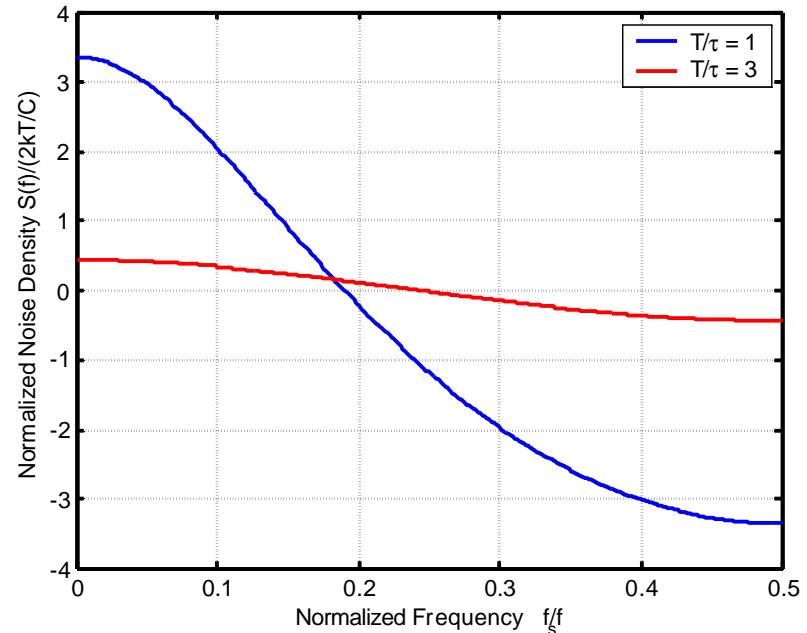
SC Resistor Noise Spectrum



$$S_y(f) = \frac{k_B T_r}{C} \frac{2}{f_s} \frac{1 - e^{-2a}}{1 + e^{-2a}(1 - \cos 2\pi f T)}$$

$$a = \frac{T}{R_{sw} C} \quad \text{and} \quad T = \frac{1}{f_s}$$

$$\int_0^{f_s/2} S_y(f) df = \frac{k_B T_r}{C}$$



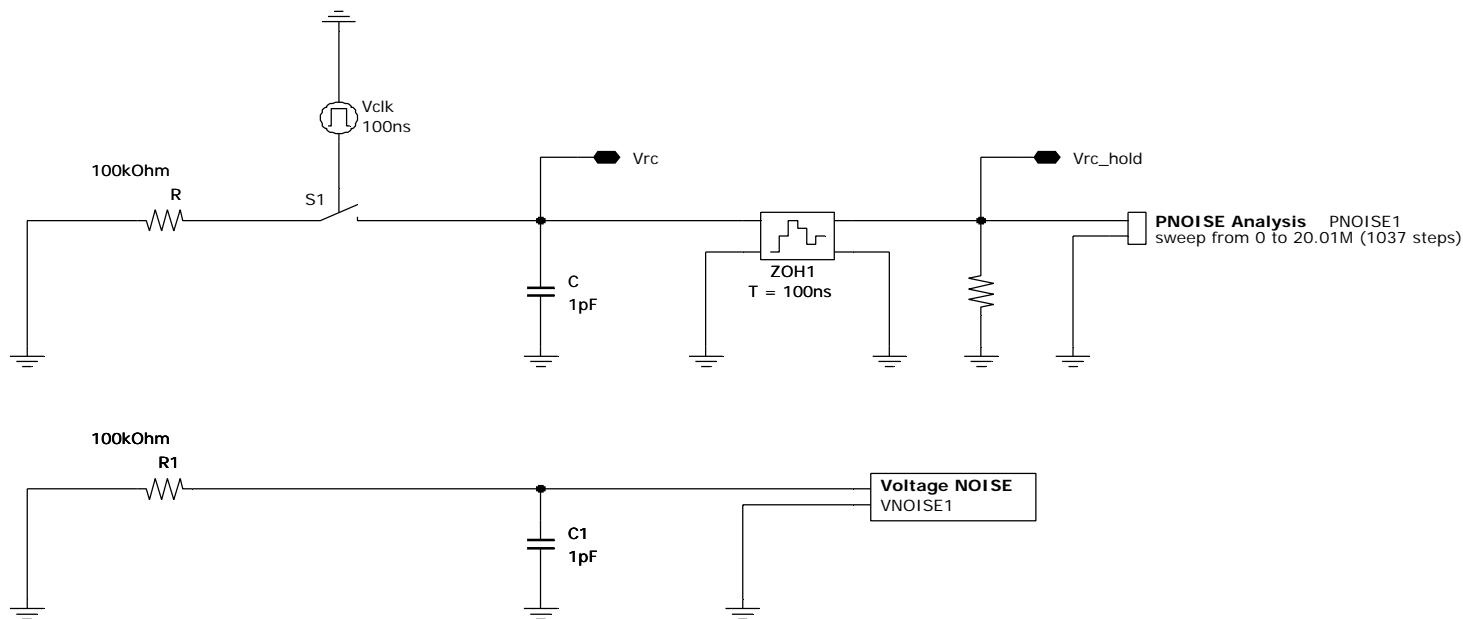
- Noise essentially white for $T/t > 3$
- Settling constraints ensure that this condition is usually met in practice
- Note: This is the noise density of an SC resistor only. The noise density from an SC filter is usually not white.

Periodic Noise Analysis

Sampling Noise from SC S/H

Netlist
ahdl_include "zoh.def"

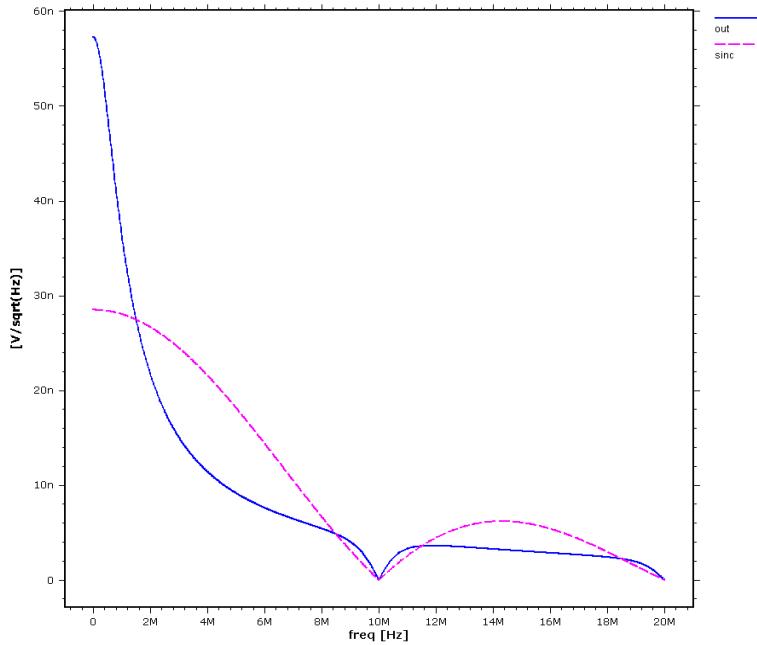
Netlist
simOptions options reltol=10u vabstol=1n labstol=1p



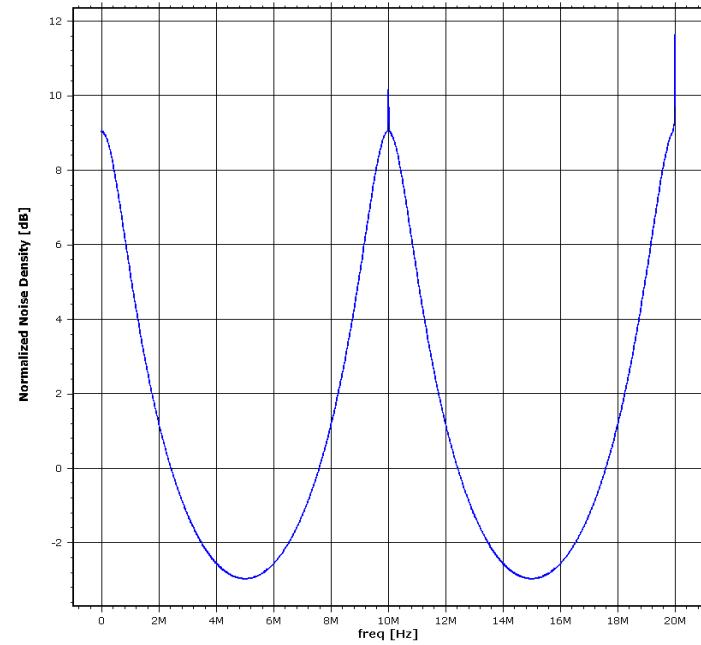
```
PSS pss period=100n maxacfreq=1.5G errpreset=conservative  
PNOISE ( Vrc_hold 0 ) pnoise start=0 stop=20M lin=500 maxsideband=10
```



Sampled Noise Spectrum



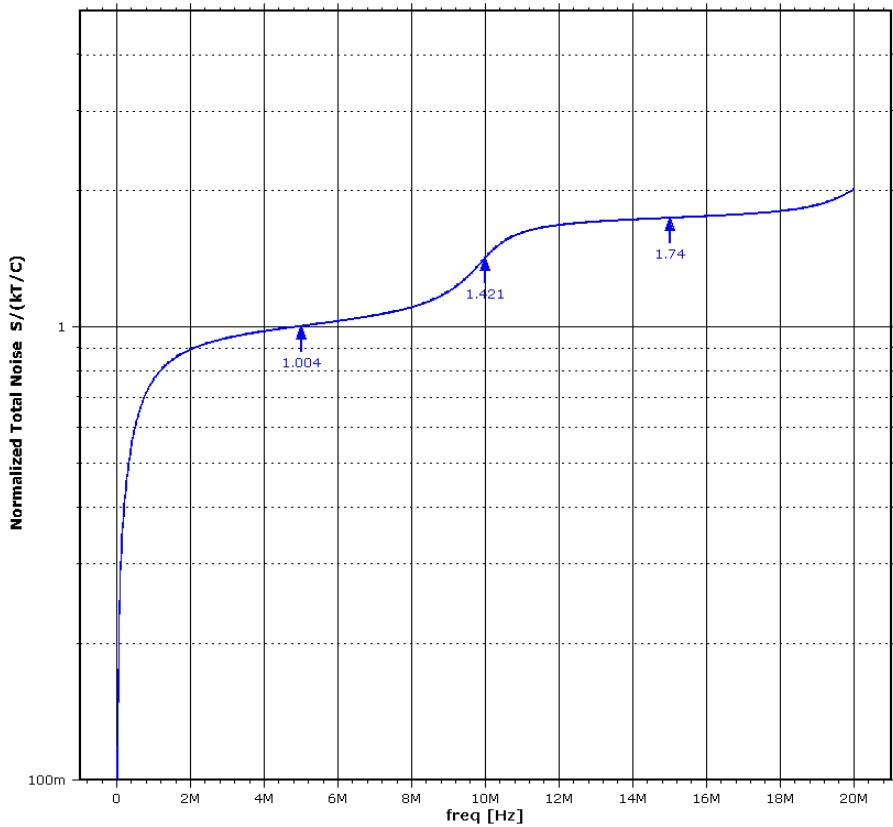
Density of sampled noise
with sinc distortion.



Normalized density of
sampled noise, corrected for
sinc distortion.



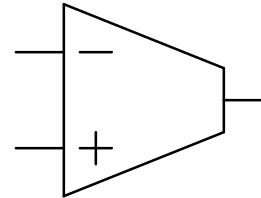
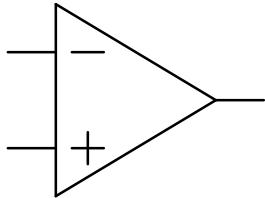
Total Noise



Sampled noise in
 $0 \dots f_s/2: 62.2\mu\text{V rms}$

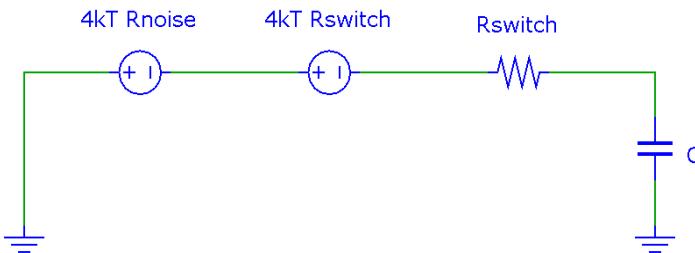
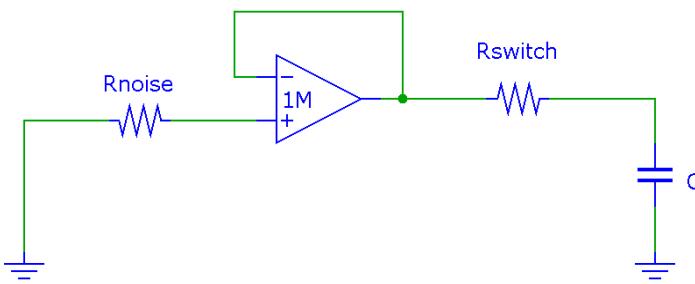
(expect $64\mu\text{V}$ for 1pF)

Opamps versus OTA



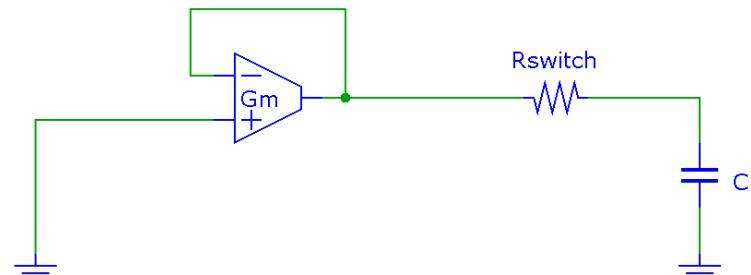
- Low impedance output
- Can drive R-loads
- Good for RC filters,
OK for SC filters
- Extra buffer adds complexity,
power dissipation
- High impedance output
- Cannot drive R-loads
- Ideal for SC filters
- Simpler than Opamp

Opamps versus OTA Noise



$$\sqrt{v_{ot}^2} = \frac{kT}{C} \left(1 + \frac{R_{noise}}{R_{switch}} \right)$$

Opamp and switch noise add



$$\sqrt{v_{ot}^2} = \frac{kT}{C}$$

OTA contributes no excess noise
(actual designs can increase noise)

Amplifier Bandwidth Requirements

SC Filter :

$$t \leq \frac{T}{10} = \frac{1}{10f_s} \quad \text{for } 16^+ \text{ Bit settling accuracy}$$

$$t = \frac{1}{w_u} = \frac{1}{2pf_u}$$

$$\rightarrow f_u \geq \frac{10}{2p} f_s \cong 2f_s$$

$$f_s = 8 \dots 100 \times f_{\text{corner}}$$

$$f_u = 16 \dots 200 \times f_{\text{corner}}$$

CT Filter :

$$f_u = 50 \dots 1000 \times f_{\text{corner}}$$

→ SC filters have comparable or slower amplifier bandwidth requirements than CT filters

SC Filter Summary

- ✓ Pole and zero frequencies proportional to
 - Sampling frequency f_s
 - Capacitor ratios
- High accuracy and stability in response
- Low time constants realizable without large R, C
- ✓ Compatible with transconductance amplifiers
 - “No” excess opamp noise
 - Reduced circuit complexity, power
- ✓ Amplifier bandwidth requirements comparable to CT filters
- o Catch: Sampled data filter → aliasing

