

1) a) T

b) F

c) F

2) a) See next

b) 1.27 MHz

c) SNR 54.94 dB

SNDR 52.86 dB

ENOB 9.47 Bits

THD 0.196%

SFDR 54.57 dB

d) 3rd harmonic power

e)

3) a) See next for plot

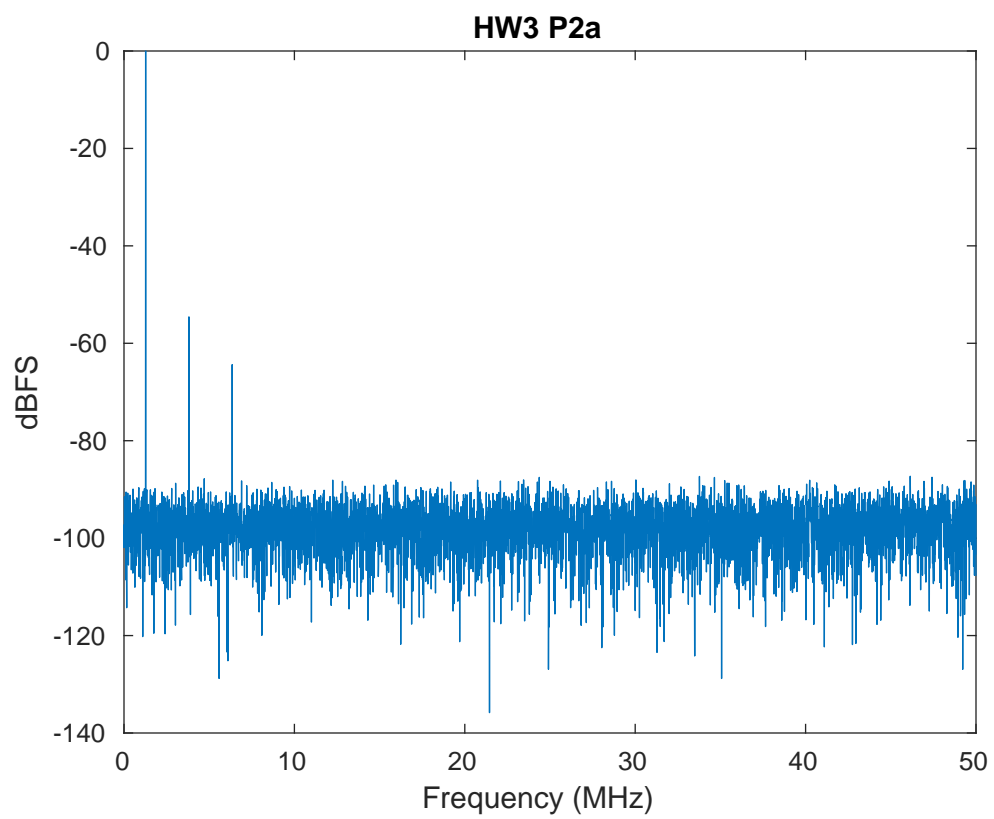
V <sub>in</sub>	Device	Resistance
0	MN	47.26
0	MN11MP	47.26
VDD	MN11MP	166.68
VDD	MP	166.68
VDD	MN11MP	166.68

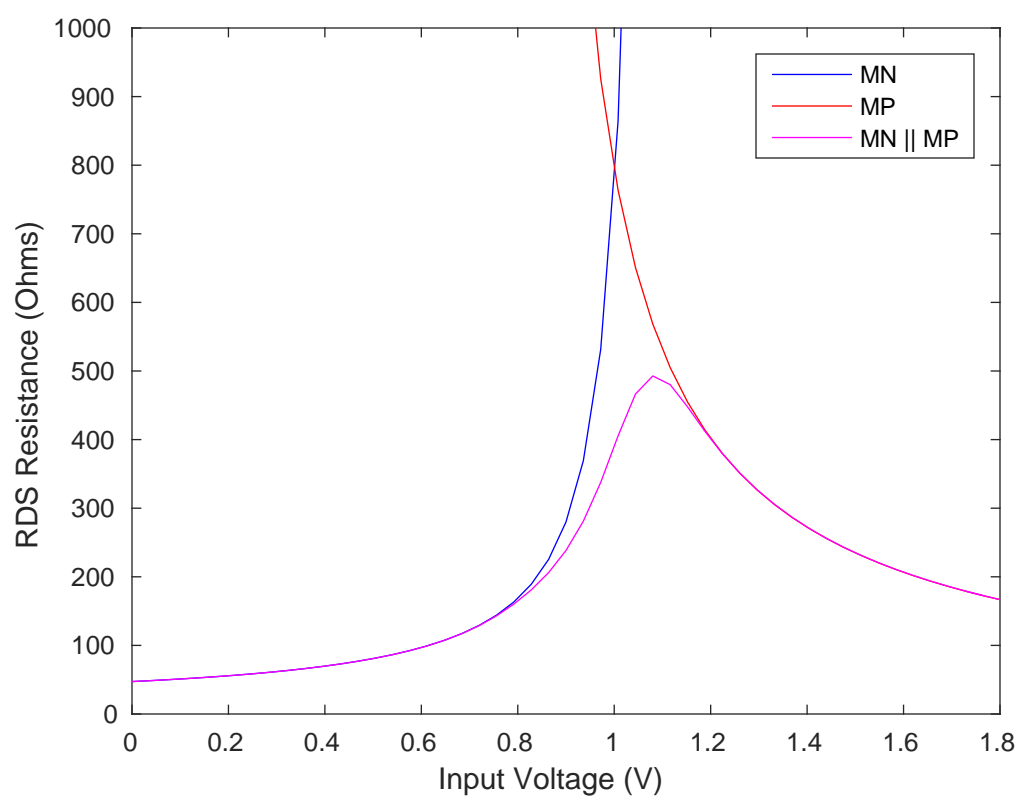
max (MN11MP) 492.74

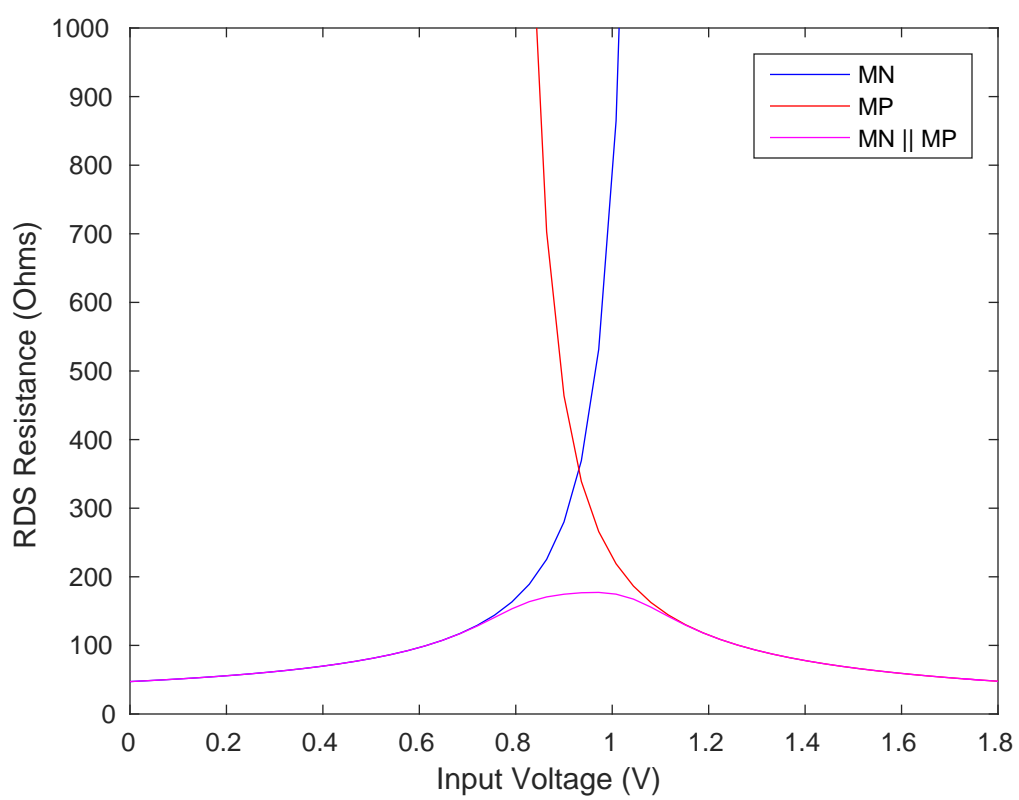
$$b) K = \frac{166.68}{47.26} = 3.52$$

c) Max R<sub>p</sub> 47.26 with W<sub>p</sub> 3.52

$$N = \ln(0.1/100) = 5.1$$







3) c)  $R_{per} = 177 \text{ w/ } 10 \mu \text{ nmos } 35.2 \mu \text{ pmos}$

$$N = -\ln(0.1/100) = 6.9$$

$$N \tau = 5 \text{ ns} \quad \tau = RC$$

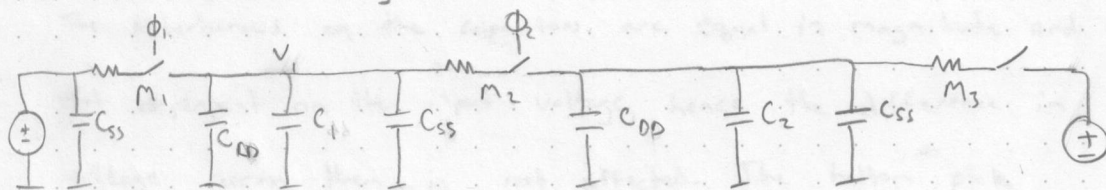
$$NRC = 5 \text{ ns} \quad R = \frac{5 \text{ ns}}{N \cdot C} = \frac{5 \text{ ns}}{6.9 \cdot 1 \text{ pF}} = 724 \Omega$$

$$\text{width}_n = 10 \mu \text{m} \cdot \frac{177}{724} = 2.44 \mu \text{m}$$

$$\text{width}_p = K \cdot \text{width}_n = 8.60 \mu \text{m}$$

4)  $W = 10 \mu \text{m} \quad L = 0.2 \mu \text{m} \quad C_{ox} = 10 \text{ fF}/\mu \text{m}^2 \quad V_t = 0.4 \quad C_{ol} = 0.1 \text{ fF}/\mu \text{m}$

Clock period is long



a)  $C_{cs} = C_{gs} + C_{sb} \quad C_{cd} = C_{gs} + C_{db} \quad C_{ol} = 10 \mu \text{m} \cdot 0.1 \frac{\text{fF}}{\mu \text{m}} = 1 \text{ fF}$

At  $t_1$  and  $t_2$   $V_g = 0$  for all nmos  $\rightarrow$  subthreshold

In subthreshold  $1 \text{ fF} \gg C_{cd} \parallel C_{cs}$

$$C_{gs} = C_{ol} \quad C_{gs} = C_{ol} \\ C_{sb} = C_{jsb} \quad C_{db} = C_{jsb}$$

$$\overline{V_n^2} = \frac{KT}{C_1} = 41.41 \mu \text{V}^2 = 64.4 \mu \text{Vrms}$$

b)  $V_i$  at  $t_1$

before  $t_1$ , while  $\phi_1$  is high,  $V_i = V_{in} = 1 \text{ V}$

when  $\phi_1 \downarrow$  the charge of  $M_1$  will be distributed into the source and into  $C_1$ ,  $\sim 50/50$

$$Q_{channel} \hat{=} -WL C_{ox} (V_{dd} - V_{in} - V_t) = -20 \mu \text{m} \cdot 0.2 \mu \text{m} \cdot 10 \text{ fF}/\mu \text{m}^2 \cdot (1.8 - 1.0 - 0.4)$$

$$Q_{ch} = -16 \text{ fC}$$

$$V_{out} = V_{in} - \frac{Q_{ch}}{2C_1} = 0.992 \text{ V} \rightarrow 8 \text{ mV offset}$$

c)  $V_{out} = V_{in} - V_{dd} \frac{C_{ov}}{C_{ol} + C} = 0.9982 \text{ V} \rightarrow 1.8 \text{ mV offset}$

5) -403m dB gain at 400MHz

b) a) See next

bi) See next for plot

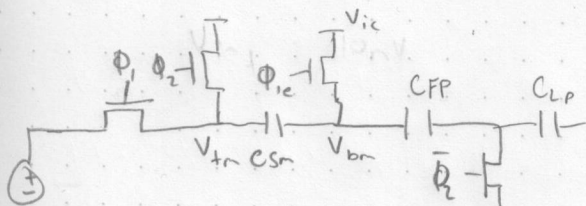
Immediately after  $\Phi_1$  rises,  $V_{cs}$  drops from 10mV to 9.386mV, an error of 0.614mV

$$V_{cs} \text{ is computed as } V_{cs} = (V_{tm} - V_{bm}) - (V_{tp} - V_{bp})$$

The disturbances on the capacitors are equal in magnitude and not dependent on the input voltage, hence the difference in voltage across them is not affected. The bottom plate sampling technique takes advantage of common mode rejection in the OTA combined with equal charge injection to null the resulting offset.

bii) See next.

biii) Charge injection error headache when  $\Phi_{ie}$  opens



$$\Phi_{ie} \text{ closed} \Rightarrow V_{bm} = V_{ic} = 0.6V$$

$$\Phi_1 \text{ closed} \Rightarrow V_{tm} = V_{ic} + \frac{V_{id}}{2} = 0.605V$$

$$\Phi_{ie} \text{ mosfet channel} = 30\mu / 0.2\mu Q_{ch} = -WL C_{ox} (V_{dd} - V_{i1} - V_{i2}) = 36fC$$

$$C_{sm} || C_{FP} = 200fF$$

$$\Delta V = \frac{Q_{ch}}{2C} = \frac{36fF}{2 \cdot 200f} = 90.0mV$$

$$113mV \text{ observed}$$



6) c)

i). Output referred differential noise

Run out of time ii

ii).  $11.7 \text{ fV}^2/\text{Hz}$

$24.460^\circ$  comes from OTA Balance  $\sim 75^\circ$  from switches

6) d)

i)  $SDR = 85.4 \text{ dB}$

ii)  $SDR = 88.6 \text{ dB}$

$\text{maxstep}$  is the maximum timestep that the simulator solver will take when computing the node states of the circuit.

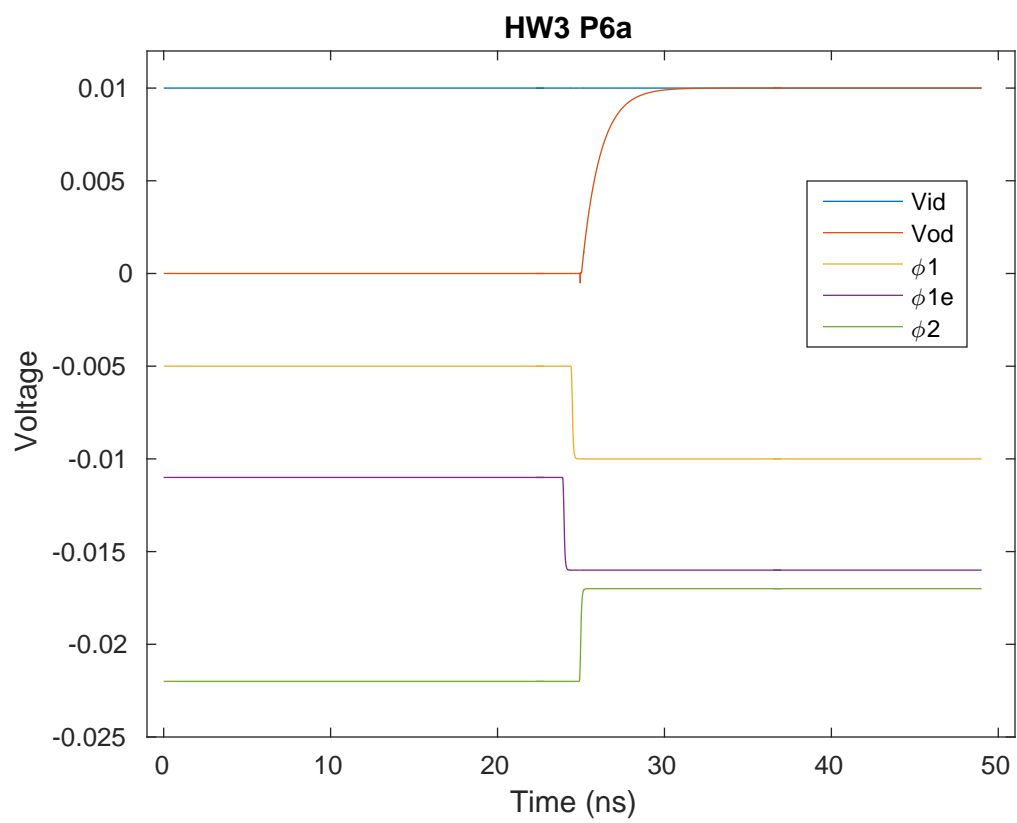
iii)  $SDR = 60.2 \text{ dB}$

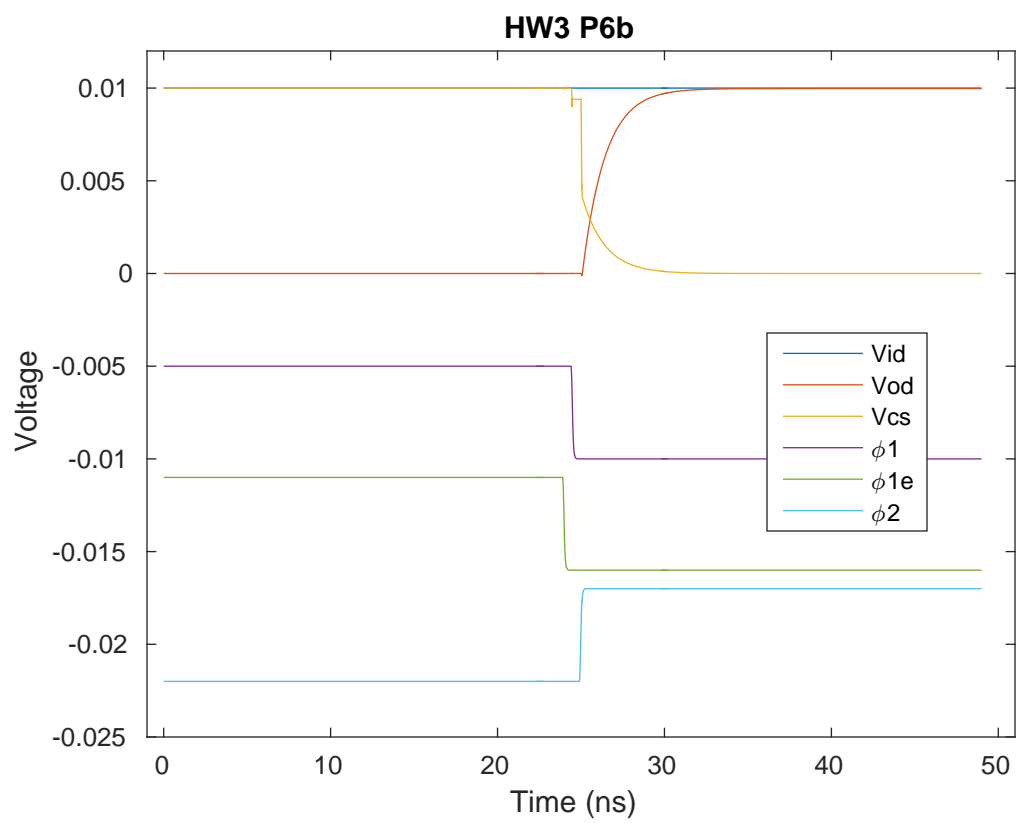
iv)  $SDR = 67.7 \text{ dB}$  with ideal switches for  $\Phi_{1e}$  and  $\Phi_2$

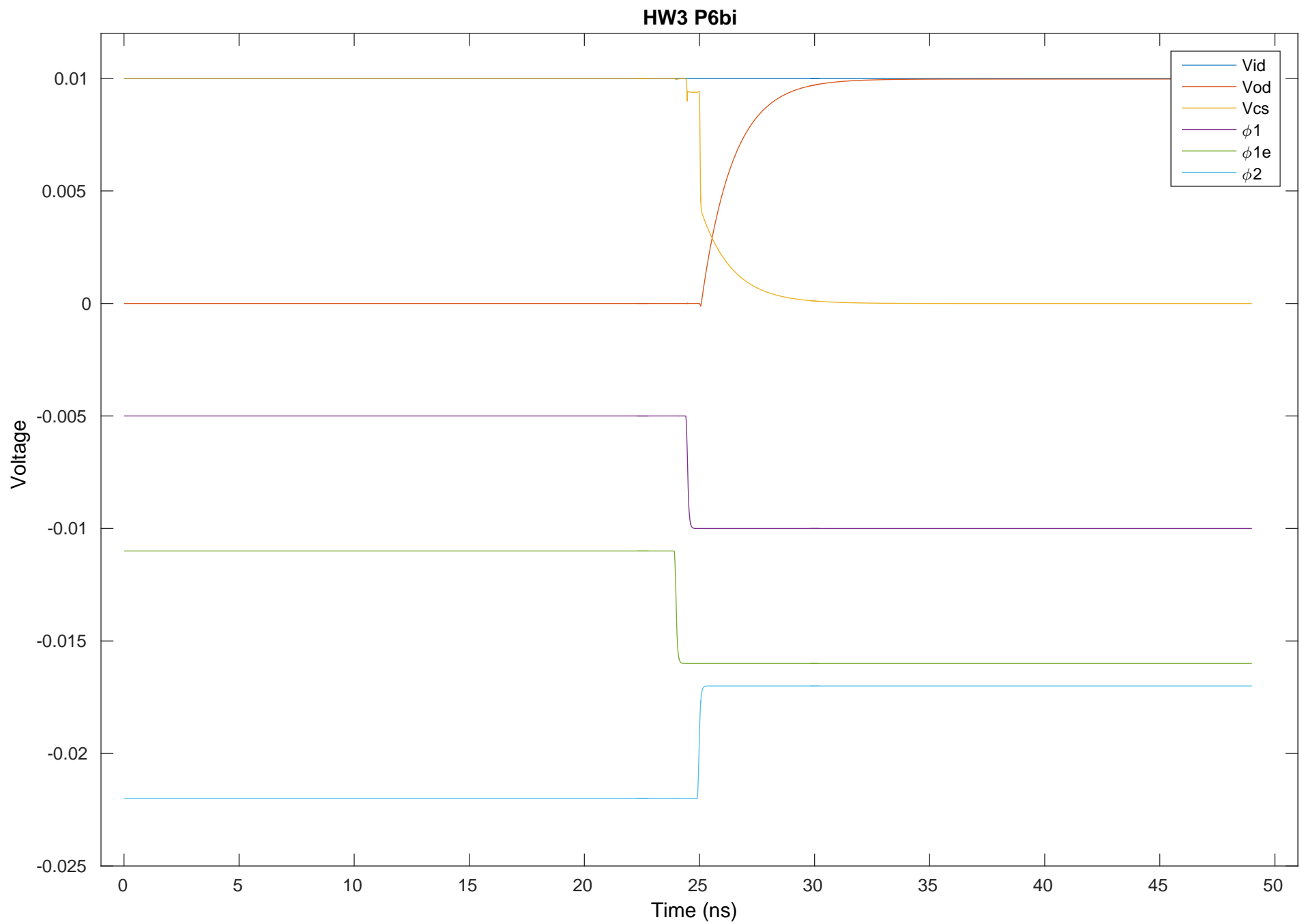
The effect is strong but it does not dominate.

v)  $SDR = -1.11 \text{ dB}$  - this doesn't make sense I think I messed something up.

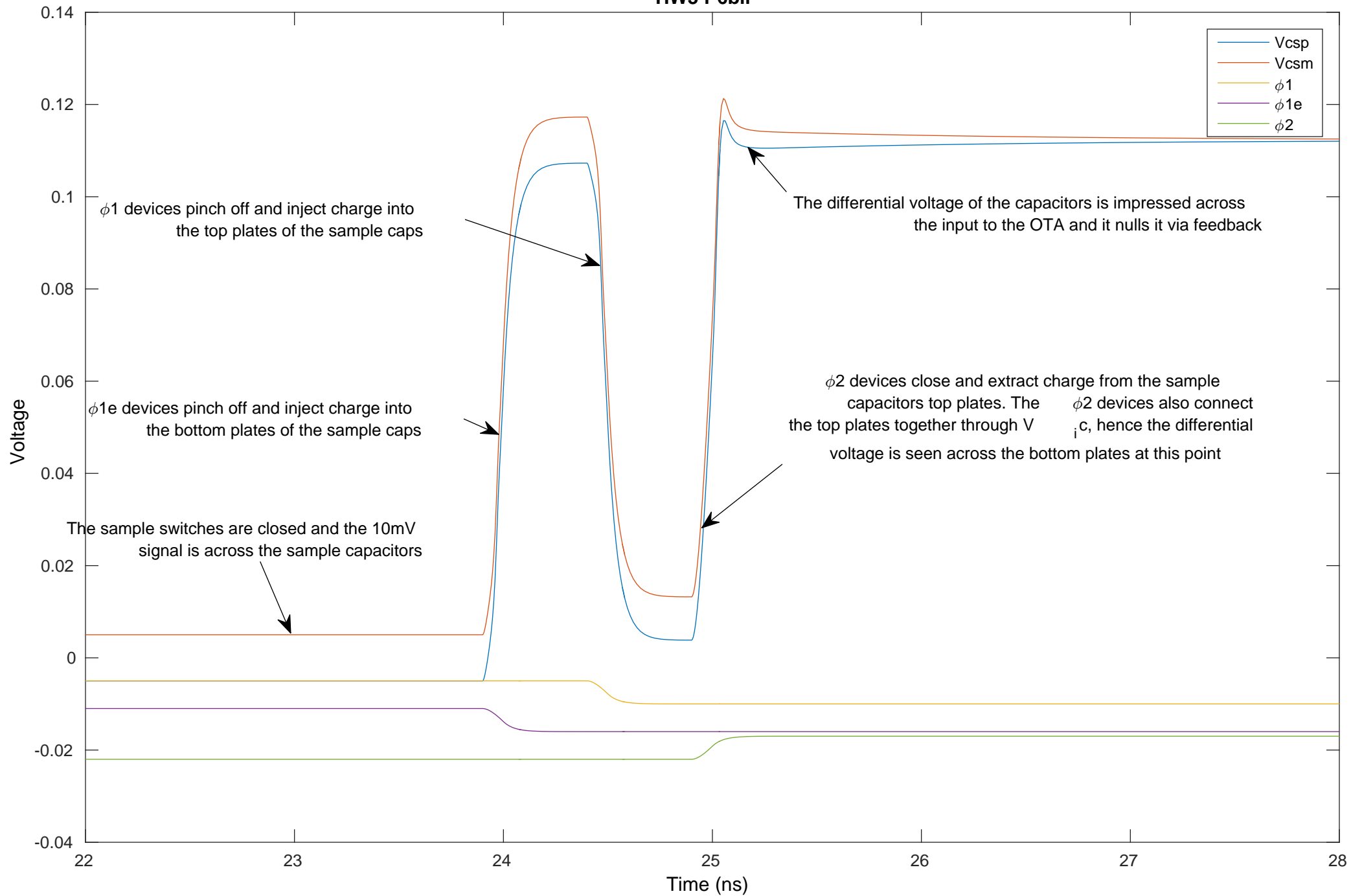




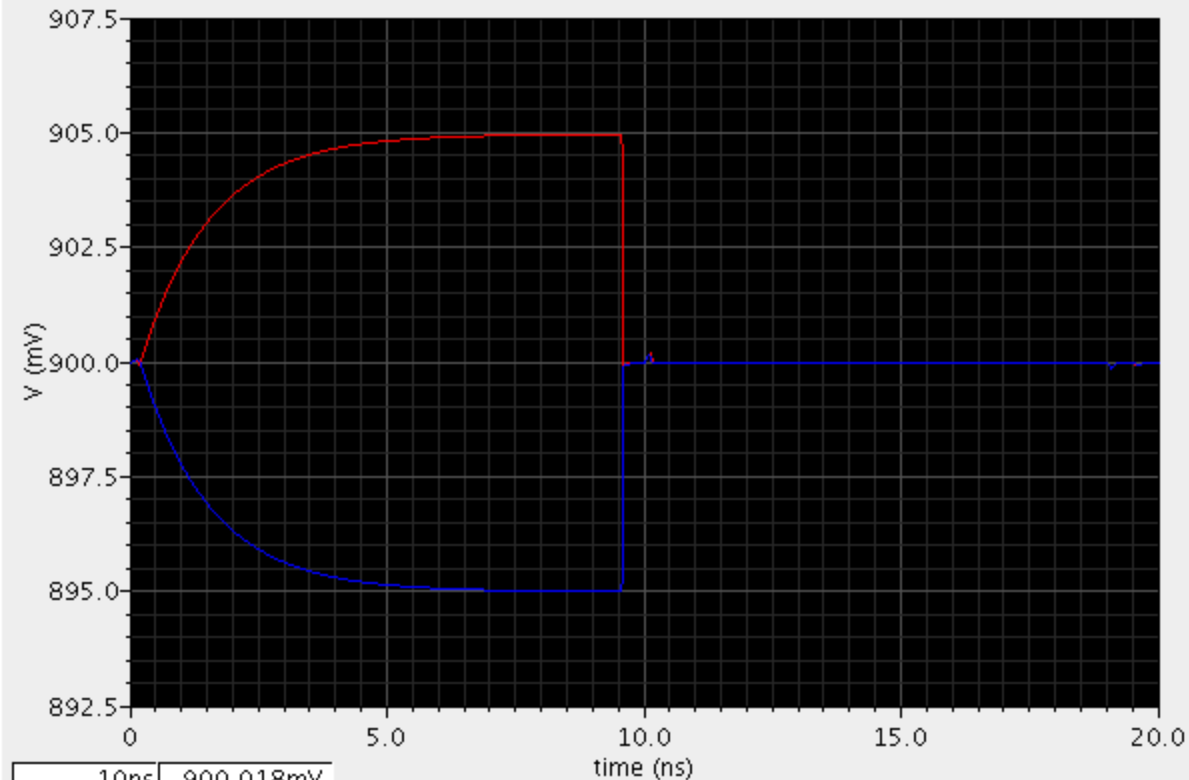




# HW3 P6bii



$v/v_{op}; pss (V)$   $v/v_{om}; pss (V)$





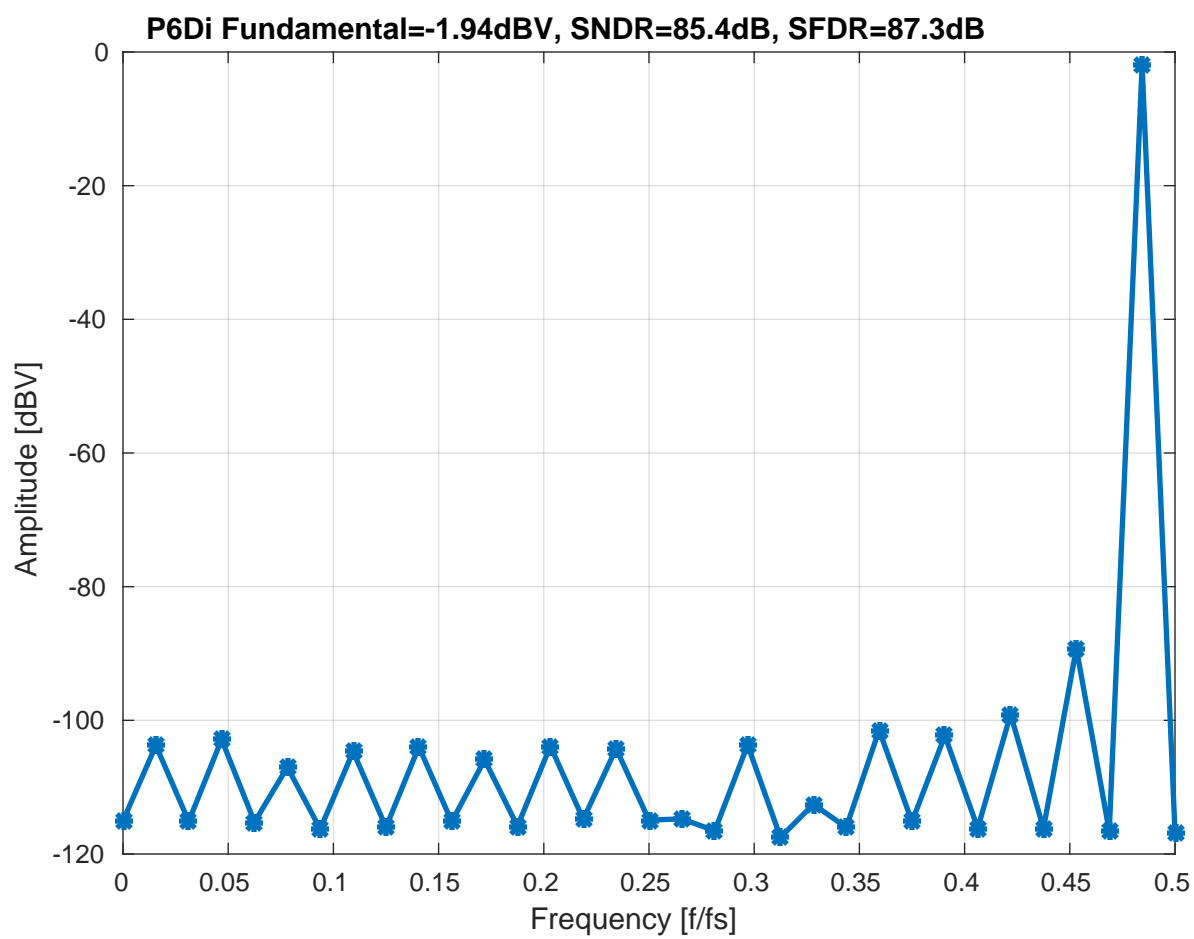
Device	Param	Noise Contribution	% Of Total
/I56/M0	id	4.21935e-15	35.86
/I54/M0	id	4.21601e-15	35.83
/I0/R2	rn	1.43879e-15	12.23
/I0/R0	rn	1.43879e-15	12.23
/I52/M0	id	1.91027e-16	1.62
/I57/M0	id	1.87769e-16	1.60
/I42/R2	rn	6.13282e-17	0.52
/I55/M0	id	6.43685e-18	0.05
/I53/M0	id	6.37452e-18	0.05

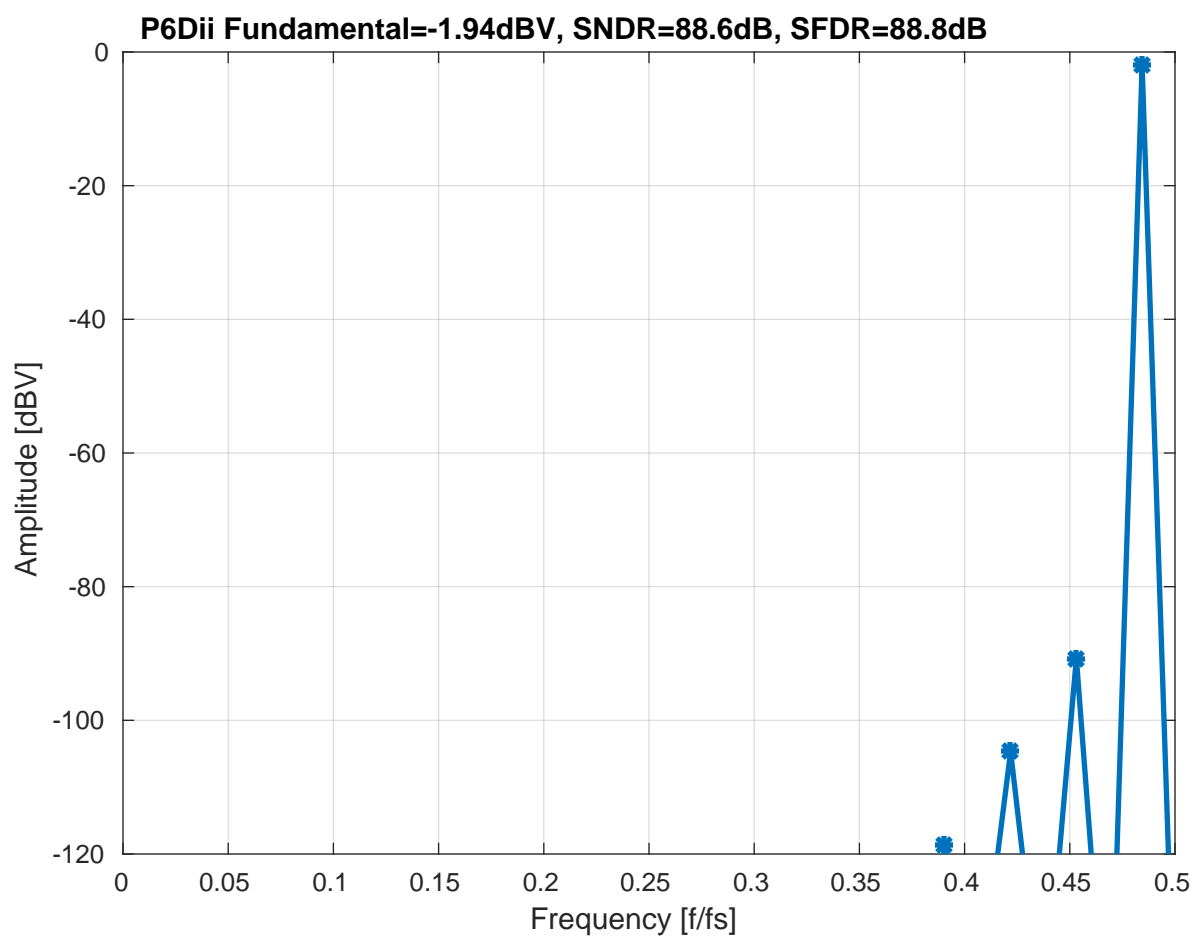
Spot Noise Summary (in  $V^2/Hz$ ) at 1K Hz Sorted By Noise Contributors

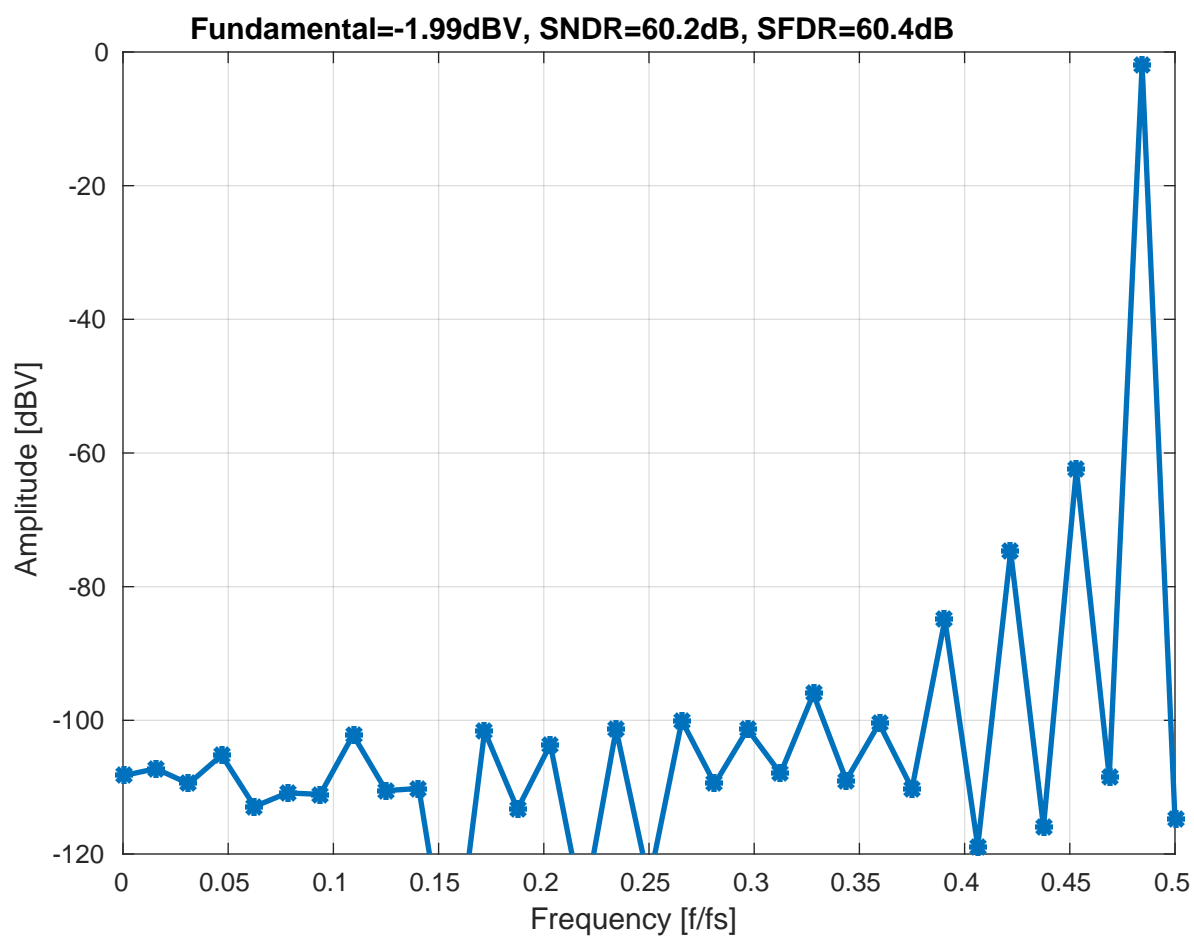
Total Summarized Noise = 1.17659e-14

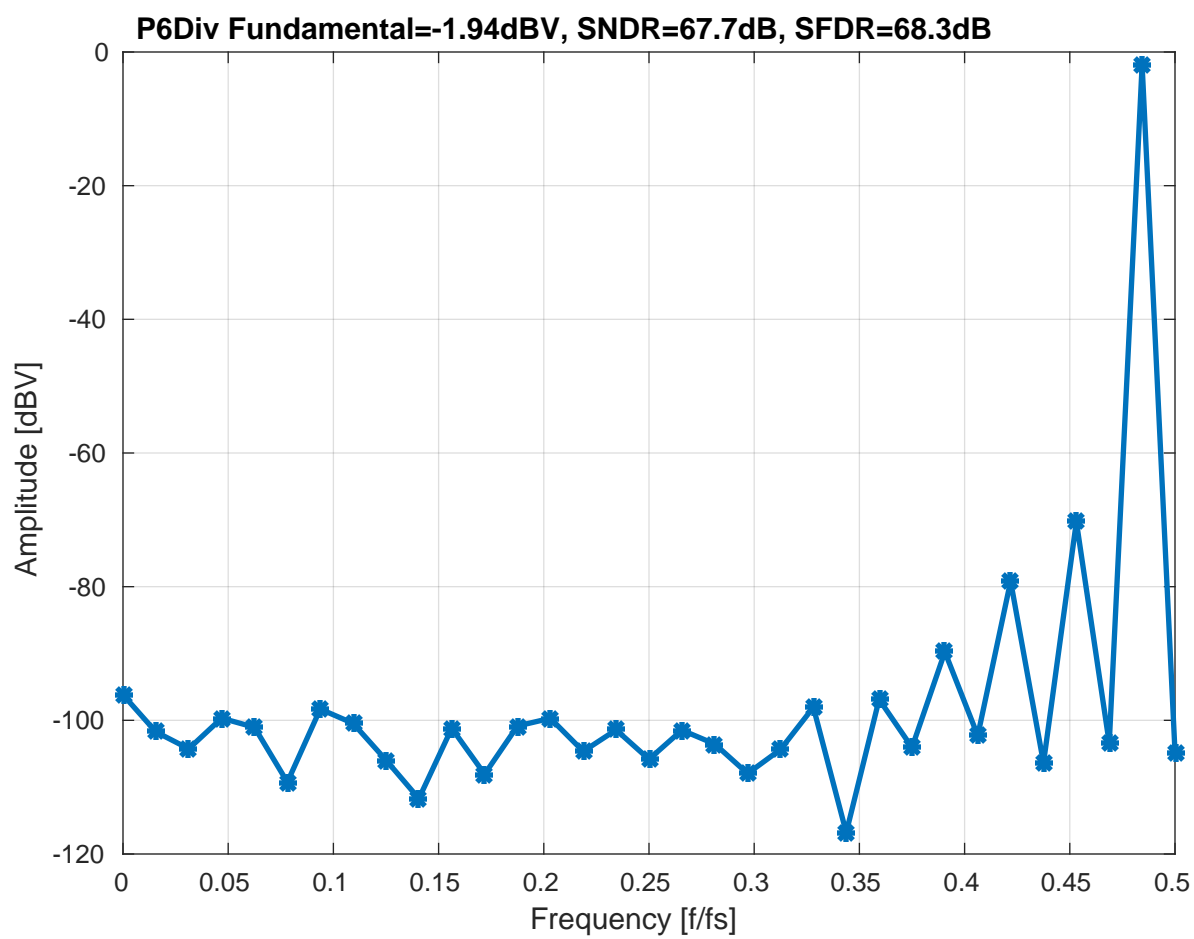
No input referred noise available

The above noise summary info is for pnoise\_td data with timeindex = 9e-09

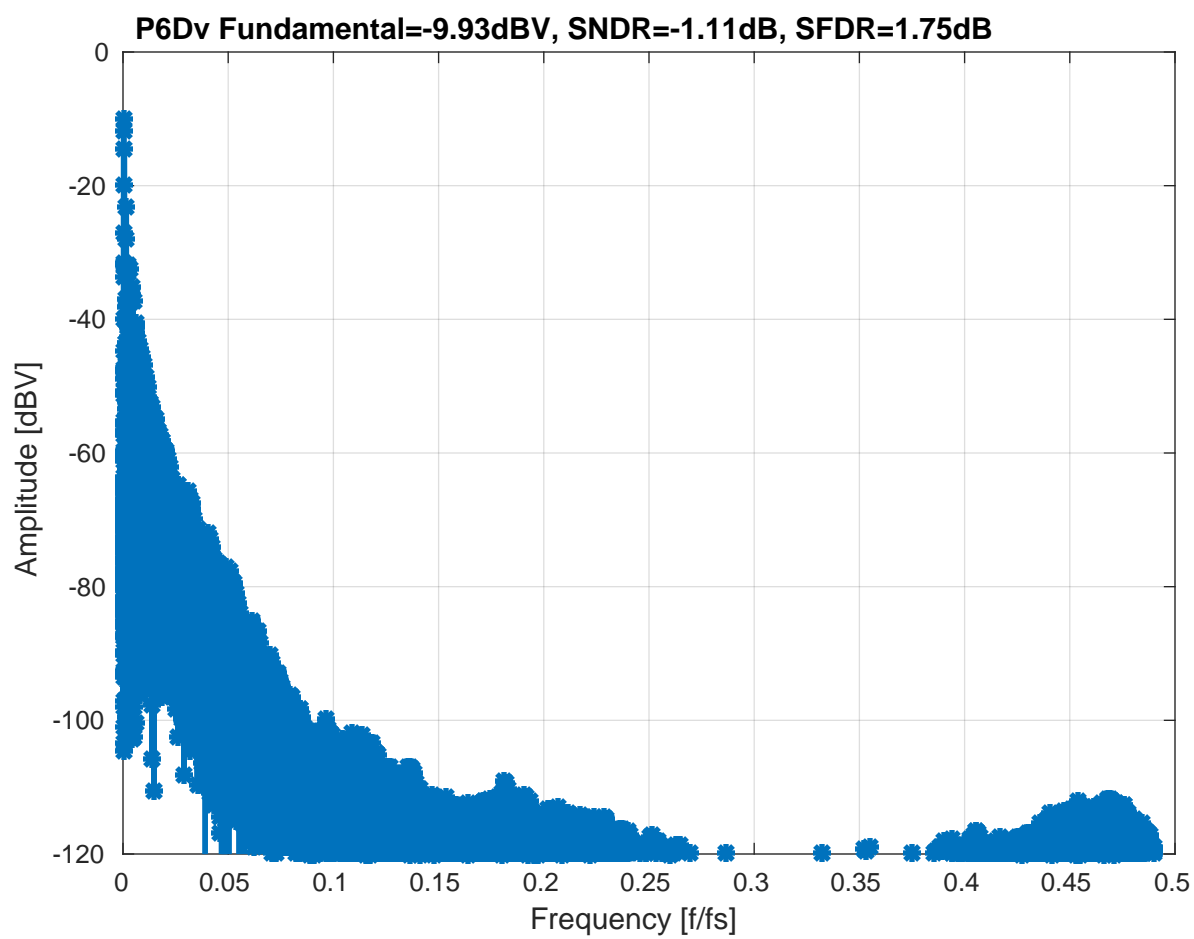












```
function enob = my_enob(x, fs)
% inputs
% x - signal (vector)
% fs - sample frequency (scalar)

    enob = (my_snr(x, fs, 6) - 1.76) / 6.02;
end
```

```
function r = my_sfdr(x, fs, n_harm)
% inputs
% x - signal (vector)
% fs - sample frequency (scalar)
% n_harm - number of harmonics to remove

% calculate fft in decibels from signal
N = length(x);

% calculate the psd with a rectangular window
[psd, f] = periodogram(x, rectwin(N), N, fs);

% find the fundamental
[~, f_idx] = max(psd);
fundamental_freq = f(f_idx);

% calculate the power of the fundamental and n harmonics
for harmonic = 1:n_harm
    h_idx = harmonic * f_idx;
    p_h(harmonic) = bandpower(psd, f, [f(h_idx-harmonic) f(h_idx+harmonic)], 'psd');
end

r = 10*log10(p_h(1) / max(p_h(2:end)));

end
```

```
function r = my_sndr(x, fs)
% inputs
% x - signal (vector)
% fs - sample frequency (scalar)

% calculate fft in decibels from signal
N = length(x);

% window function to use - matlab uses hamming by default
window = hamming(N);
>window = rectwin(N);

% calculate the psd with a rectangular window
[psd, f] = periodogram(x, window, N, fs);

% find the fundamental
[~, f_idx] = max(psd);
fundamental_freq = f(f_idx);

% remove the fundamental
psdn = psd;
for i = f_idx-1:f_idx+1
    psdn(i) = median(psd);
end

%psd_db = 10*log10(psd);
%psdn_db = 10*log10(psdn);

sig_pwr = bandpower(psd, f, 'psd');
noise_pwr = bandpower(psdn, f, 'psd');
r = 10*log10(sig_pwr / noise_pwr);

end
```

```
function r = my_snr(x, fs, n_harm)
% inputs
% x - signal (vector)
% fs - sample frequency (scalar)
% n_harm - number of harmonics to remove

% calculate fft in decibels from signal
N = length(x);

% calculate the psd with a rectangular window
[psd, f] = periodogram(x, rectwin(N), N, fs);

% find the fundamental
[~, f_idx] = max(psd);
fundamental_freq = f(f_idx);

% remove the fundamental and n harmonics
psdn = psd;
for harmonic = 1:n_harm
    harm_f = harmonic * fundamental_freq;
    harm_idx = harmonic * f_idx;
    for i = harm_idx-harmonic:harm_idx+harmonic
        psdn(i) = median(psd);
    end
end

%psd_db = 10*log10(psd);
%psdn_db = 10*log10(psdn);

sig_pwr = bandpower(psd, f, 'psd');
noise_pwr = bandpower(psdn, f, 'psd');
r = 10*log10(sig_pwr / noise_pwr);

end
```



```
function r = my_thd(x, fs, n_harm)
% inputs
% x - signal (vector)
% fs - sample frequency (scalar)
% n_harm - number of harmonics to remove

% calculate fft in decibels from signal
N = length(x);

% calculate the psd with a rectangular window
[psd, f] = periodogram(x, rectwin(N), N, fs);

% find the fundamental
[~, f_idx] = max(psd);
fundamental_freq = f(f_idx);

% calculate the power of the fundamental and n harmonics
for harmonic = 1:n_harm
    h_idx = harmonic * f_idx;
    p_h(harmonic) = bandpower(psd, f, [f(h_idx-harmonic) f(h_idx+harmonic)], 'psd');
end

r = 100 * sqrt(sum(p_h(2:end)) / p_h(1));
end
```