

SAMPLING THEORY

ANALOG VOLTAGE $v_a(t)$

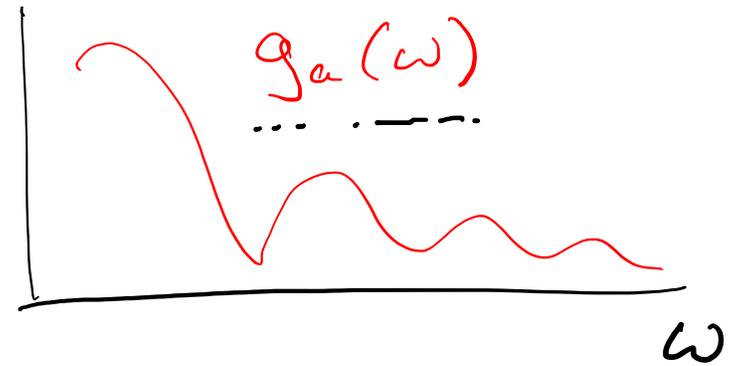
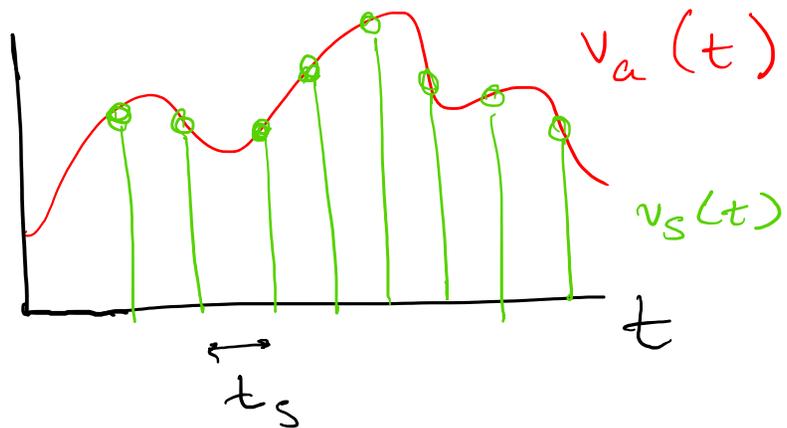
FOURIER TRANSFORM $g_a(\omega)$

$$v_a(t) = \frac{1}{\sqrt{2\pi}} \int g_a(\omega) e^{j\omega t} d\omega$$

$$g_a(\omega) = \frac{1}{\sqrt{2\pi}} \int v_a(t) e^{-j\omega t} dt$$

SAMPLE v_a @ INTERVALS t_s

WHAT IS SPECTRAL CONTEXT OF SAMPLED WAVEFORM?



$$v_s(t) = \frac{1}{\sqrt{2\pi}} \int g_s(\omega) e^{j\omega t} d\omega$$

$$g_s(\omega) = \frac{1}{\sqrt{2\pi}} \int v_s(t) e^{-j\omega t} dt$$

SAMPLES
WAVEFORM

??

INTRODUCE

$$\rightarrow \Delta(t; t_s) = \sum_{\substack{\ell = -\infty \\ \dots}}^{\infty} \delta(t - \ell t_s) t_s \quad \ell \text{ INTEGER}$$

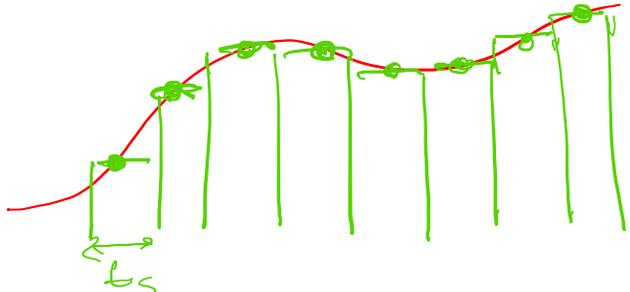


THEN

$$\rightarrow v_s(t) = v_a(t) \Delta(t; t_s) \leftarrow$$
$$= \sum_{\ell = -\infty}^{+\infty} v_a(t) \delta(t - \ell t_s) t_s$$

e.g. $\int v_s(t) dt = \int \sum v_a(t) \delta(t - q t_s) t_s dt$ $\int f(x) \delta(x - x_0) dx = f(x_0)$

$= \sum_q v_a(q t_s) t_s$ PIECEWISE INTEGRATION



Now, $\Delta(t; t_s)$ is a PERIODIC FUNCTION

→ CAN EXPRESS AS A DISCRETE FOURIER SERIES

RECALL $f(t) = \sum_{-\infty}^{\infty} c_n e^{j n \omega_0 t}$

↑ PERIODIC
↑ FUNDAMENTAL
↑ PERIOD

$\omega_0 = \frac{2\pi}{T}$

WHERE $c_n = \frac{1}{T} \int_{-T/2}^{T/2} f(t) e^{-j n \omega_0 t} dt$ [ORTHOGONALITY OF $e^{j n \omega_0 t}$]

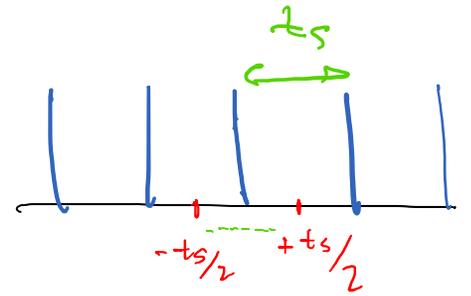
* $c_n = c_{-n}$

∴ we can write

$$\Delta(t; t_s) = \sum_{-\infty}^{\infty} \underline{a_n} e^{j n \omega_s t} \quad ; \quad \omega_s = \frac{2\pi}{t_s}$$

[SAMPLING FREQ.]

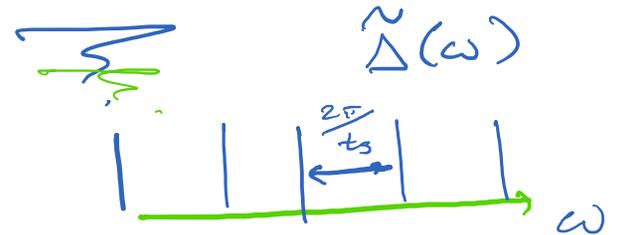
$$\rightarrow \underline{a_n} = \frac{1}{t_s} \int_{-t_s/2}^{t_s/2} \Delta(t; t_s) e^{-j n \omega_s t} dt$$



$$= \frac{1}{t_s} \int_{-t_s/2}^{t_s/2} \left[\sum \delta(t - \frac{n}{\omega_s} t_s) t_s \right] e^{-j n \omega_s t} dt$$

$$= \frac{1}{t_s} \times t_s \int_{-t_s/2}^{t_s/2} \delta(t) e^{-j n \omega_s t} dt = 1$$

$$\Delta(t; t_s) = \sum_{-\infty}^{\infty} e^{j n \omega_s t}$$



$$g_s(\omega) = \frac{1}{\sqrt{2\pi}} \int_{-\infty}^{\infty} v_a(t) \left[\sum e^{j n \omega_s t} \right] e^{-j \omega t} dt$$

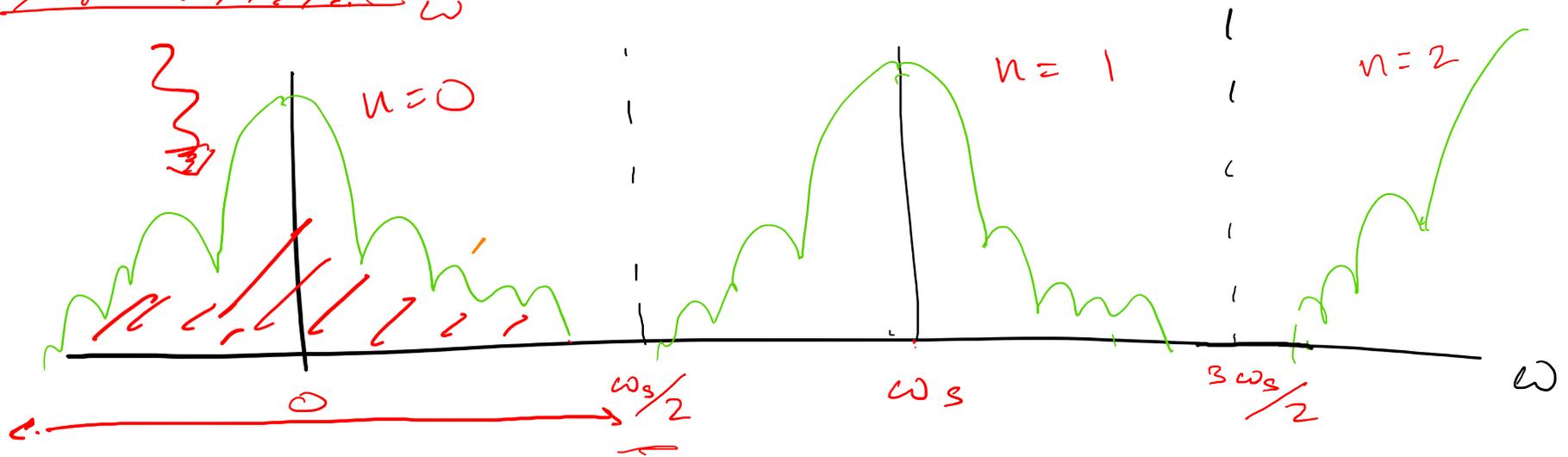
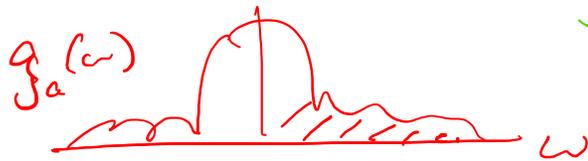
$v_a(t) \Delta(t; t_s) = v_s(t)$

$$= \frac{1}{\sqrt{2\pi}} \sum_n \int_{-\infty}^{\infty} v_a(t) e^{-j(\omega - n\omega_s)t} dt$$

$$\rightarrow g_s(\omega) = \sum_{n=-\infty}^{\infty} g_a(\omega - n\omega_s)$$

SUM OF COPIES OF ORIGINAL FOURIER TRANSFORM $g_a(\omega)$ DISPLACED IN FREQUENCY BY INTEGER MULTIPLES OF SAMPLING FREQUENCY ω_s .

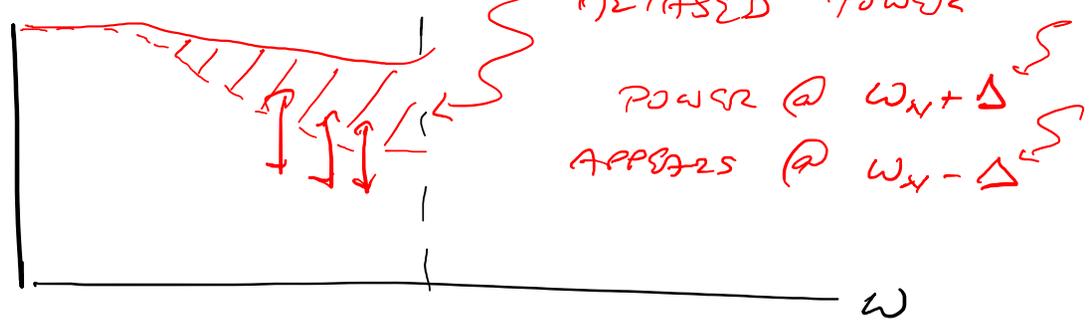
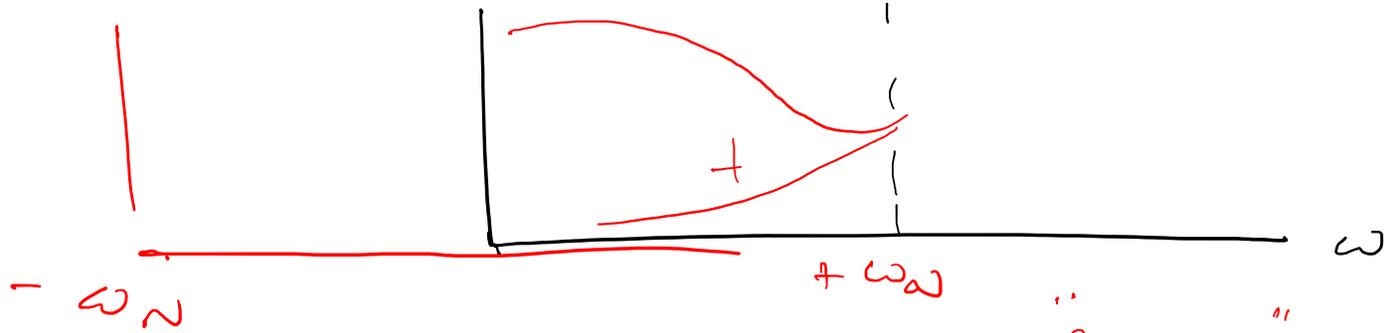
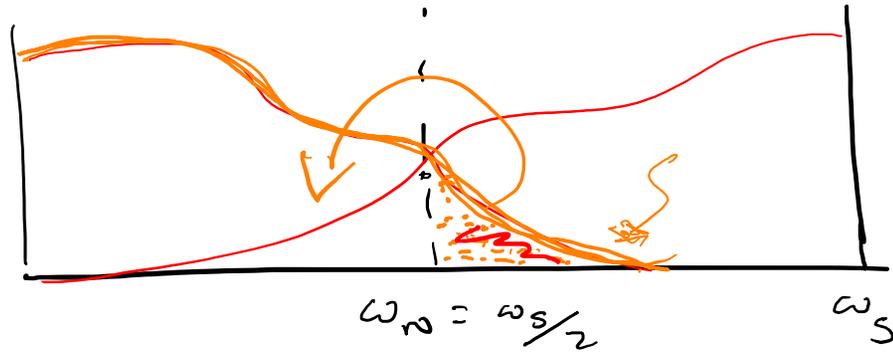
i.e.
$$g_s(\omega) = \dots g_a(\omega - 2\omega_s) + g_a(\omega - \omega_s) + g_a(\omega) + g_a(\omega + \omega_s) + g_a(\omega + 2\omega_s) + \dots$$



$\omega_s/2 \rightarrow$ NYQUIST FREQUENCY

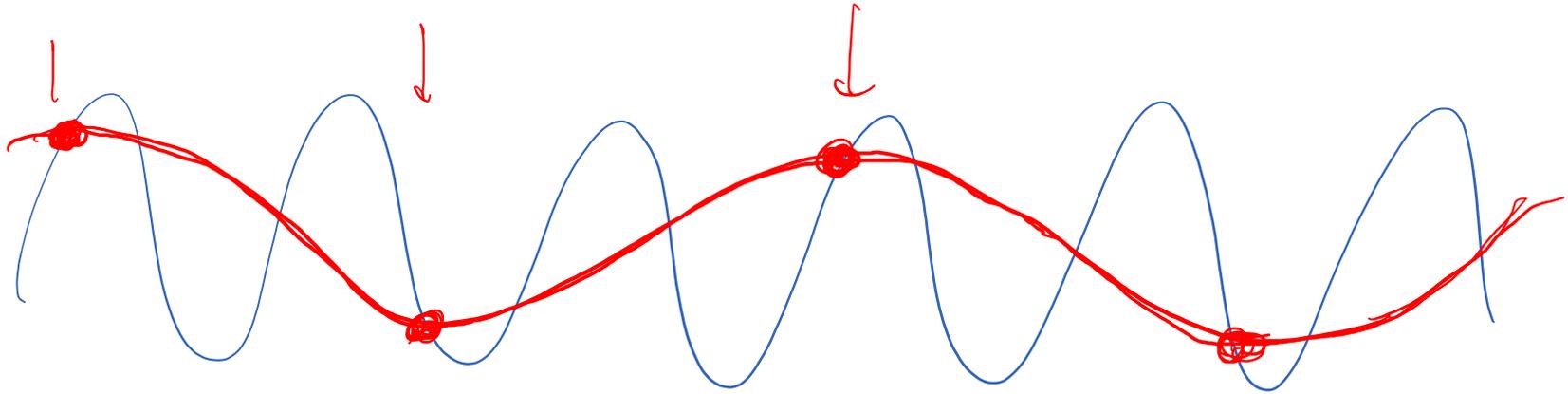
IF NO SPECTRAL CONTENT
 @ $\omega > \omega_s/2$,
 SAMPLING FAITHFULLY
 REPRODUCES "TRUE" SPECTRUM.

ANOTHER EXAMPLE



"ALIASED" POWER
 POWER @ $\omega_N + \Delta$
 APPEARS @ $\omega_N - \Delta$

EXAMPLE



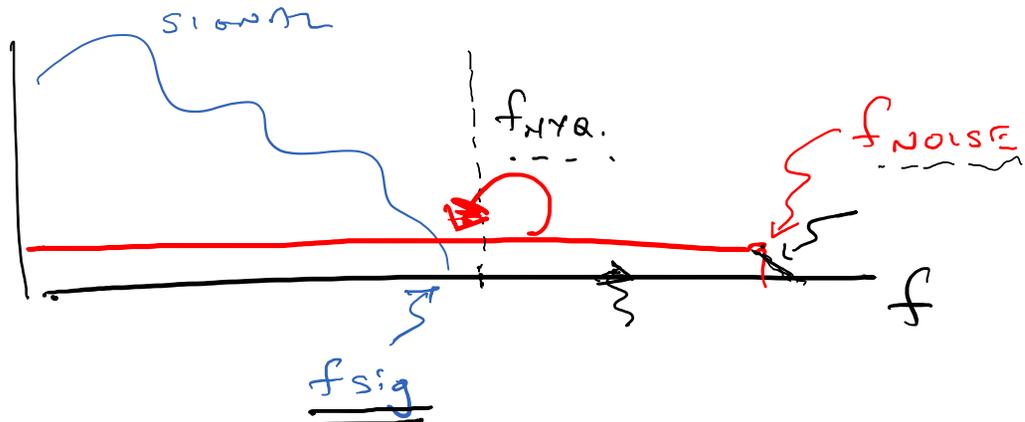
SIGNAL @ $\frac{3}{2} f_{\text{sample}}$ = $3 f_{\text{nyquist}}$ IS ALIASED

$$\text{TO } \underline{f_{\text{nyquist}}} = \underline{\frac{1}{2} f_{\text{sample}}}$$

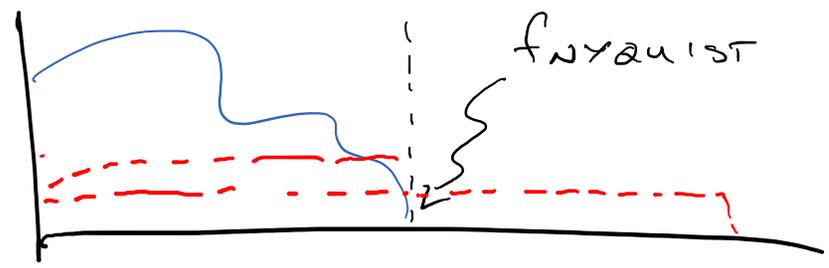
EXAMPLE 1

SIGNAL IN BW $0 \rightarrow f_{sig}$

AMPLIFIER w/ [WHITE] NOISE
IN BW $0 \rightarrow f_{noise} > f_{sig}$

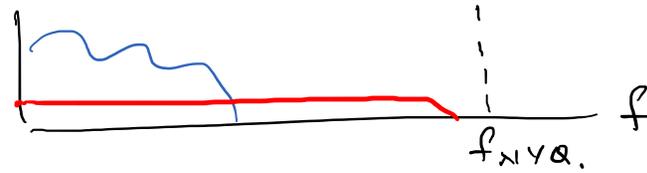


SAMPLE @ $2f_{sig}$?



SNR DEGRADATION
BY FACTOR 2

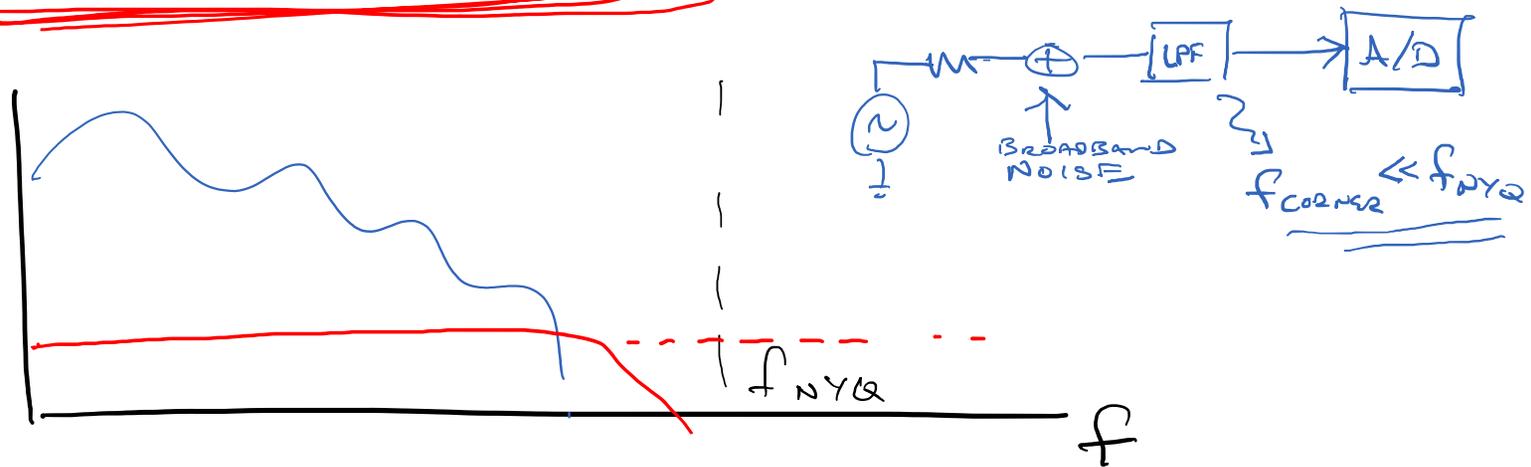
SOLUTIONS:



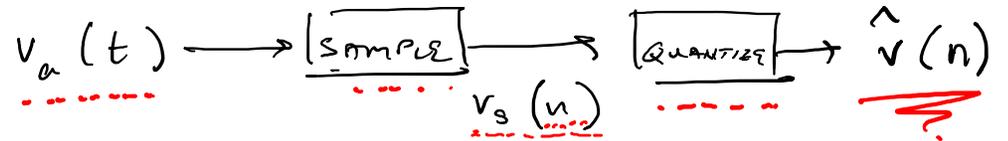
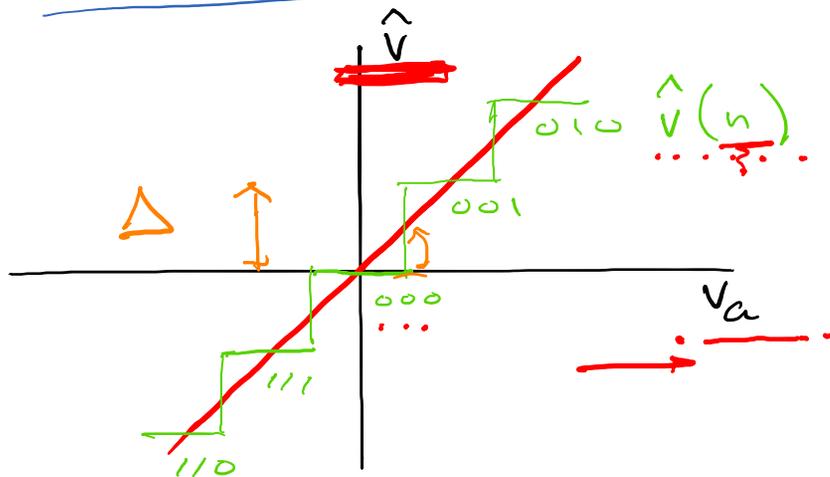
① SAMPLE @ FREQUENCY S.T. $f_{SAMPLER} > f_{NOISE}$

[NO NOISE ABOVE f_{NYQ} . TO GET FILTERED DOWN TO LOW FREQUENCY]

② ANTI-ALIASING FILTER,



FINITE SAMPLING RESOLUTION : QUANTIZATION ERROR

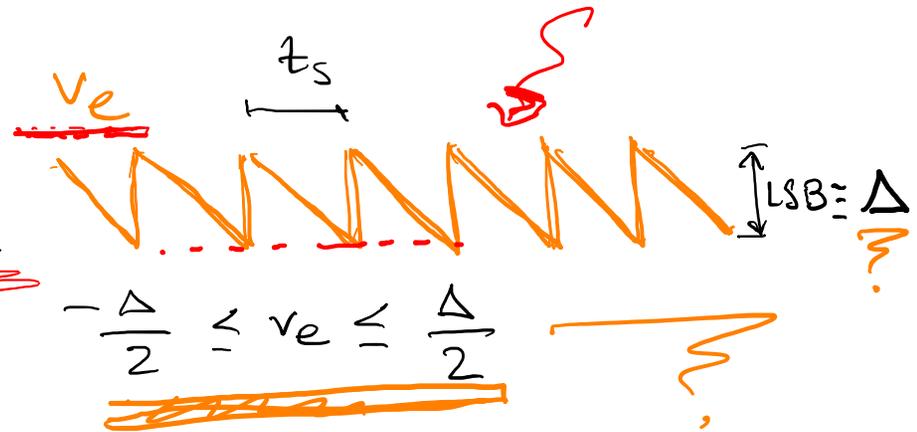
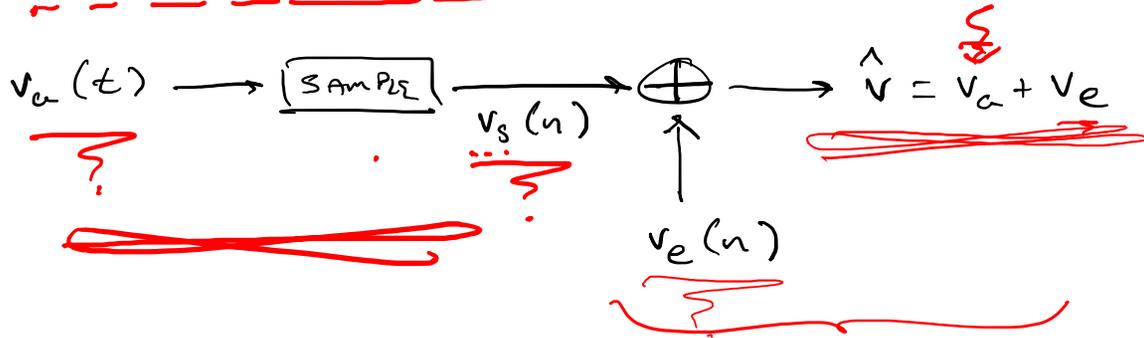


$$\hat{v}(n) = Q[v_s(n)]$$

NONLINEAR TRANSFORMATION

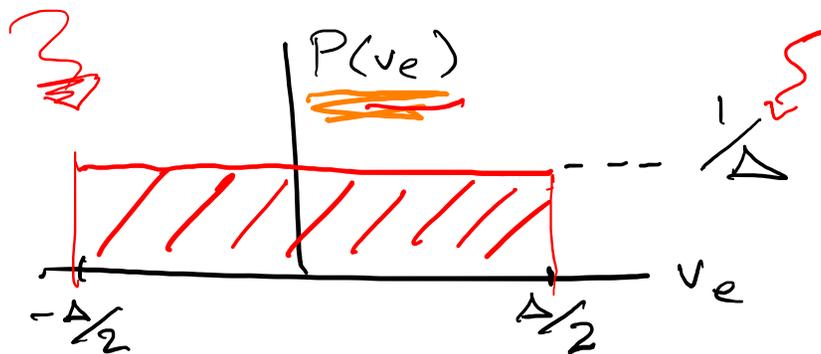


SIMPLIFIED STATISTICAL / NOISE MODEL



FOR STATISTICAL ANALYSIS OF SAMPLED SIGNALS,
WE'LL ASSUME:

1. v_e IS A SAMPLE SEQUENCE OF A STATIONARY RANDOM PROCESS
2. ERROR SEQUENCE v_e IS UNCORRELATED w/ EXACT SEQUENCE OF SIGNAL SAMPLES v_s
3. RANDOM SAMPLES OF THE ERROR SEQUENCE ARE UNCORRELATED (WHITE NOISE)
4. PROBABILITY DISTRIBUTION OF ERROR PROCESS IS UNIFORM OVER RANGE OF QUANTIZATION ERROR.



⇒ ASSUMPTIONS ARE BEST SATISFIED FOR MORE COMPLEX SIGNALS
[FOR WHICH CORRELATION BETWEEN $v_a \neq v_e$ DECREASES]

SPECTRAL DENSITY OF QUANTIZATION NOISE??

RECALL

$$\int_{-\infty}^{\infty} S_x(f) df = \lim_{T \rightarrow \infty} \left\{ \frac{1}{T} \int_{-T/2}^{T/2} x^2(t) dt \right\}$$

$= \langle x^2 \rangle$

HERE, WANT

$$S_e(f)$$

CALCULATE

$$\langle v_e^2 \rangle = \int_{-\Delta/2}^{\Delta/2} P(v_e) v_e^2 dv_e = \frac{1}{\Delta} \int_{-\Delta/2}^{\Delta/2} v_e^2 dv_e$$

$$\langle v_e^2 \rangle = \frac{1}{3\Delta} \left[\frac{\Delta^3}{8} - \left(-\frac{\Delta^3}{8} \right) \right] = \frac{1}{12} \Delta^2$$

LSB

$$\int_{-f_{\text{Nyq}}}^{f_{\text{Nyq}}} S_e(f) df = \frac{1}{12} \Delta^2$$

$$\rightarrow S_e(f) \cdot 2f_{\text{Nyq}} = \frac{1}{12} \Delta^2$$

$$\rightarrow S_e(f) = \frac{\Delta^2}{12} T_s$$

$$f_{\text{Nyq}} = \frac{1}{2T_s}$$

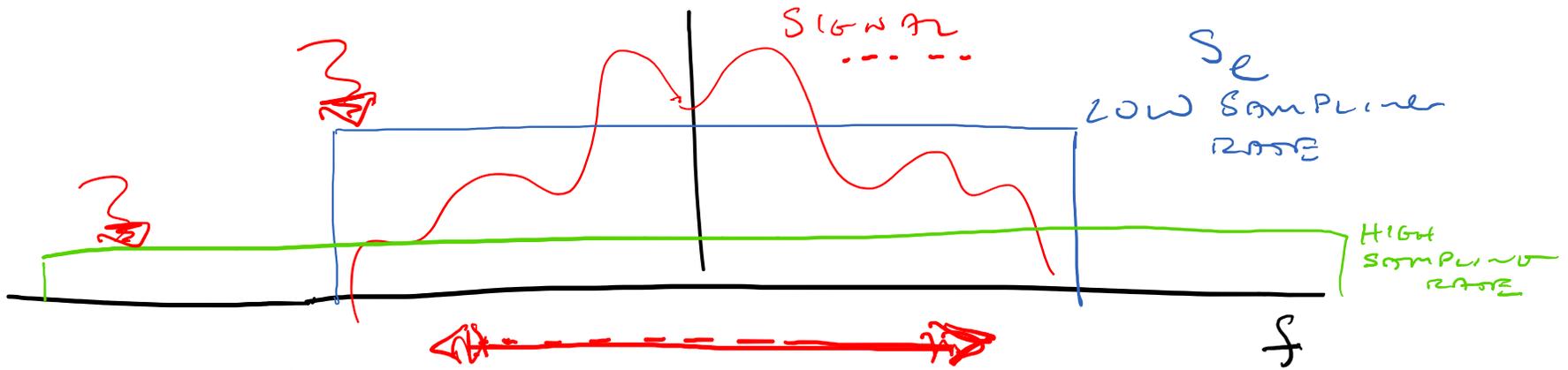
→ DECREASE QUANTIZATION NOISE BY

EITHER DECREASING Δ

OR DECREASING T_s

$$S_e(f) = \frac{\Delta^2}{12} t_s$$

POWER
SNR $\sim \frac{1}{t_s}$



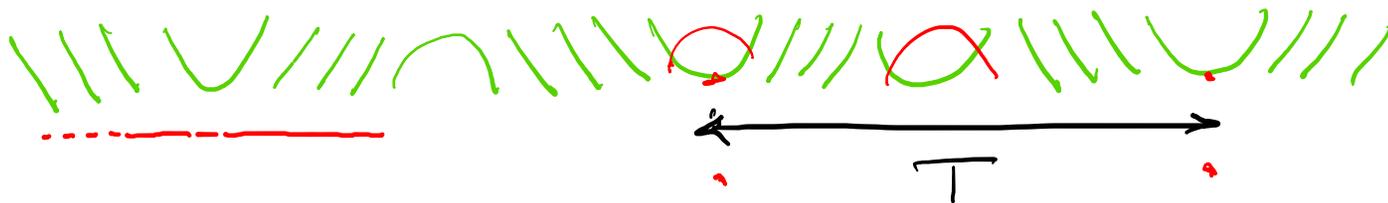
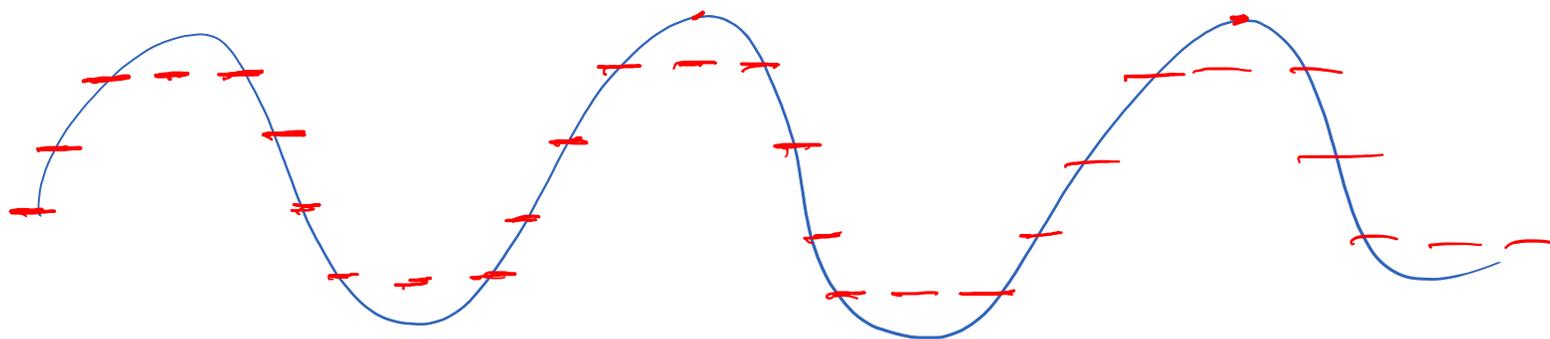
→ "OVERSAMPLING"

e.g. AD7760 Σ - Δ ADC
2.5 MSPS, 24 bit

→ SAMPLE @ 40 MHz, THEN USE DIGITAL FILTER TO ELIMINATE out-of-band QUANTIZATION NOISE

ANOTHER CONSEQUENCE OF FINITE A/D RESOLUTION:
HARMONIC DISTORTION

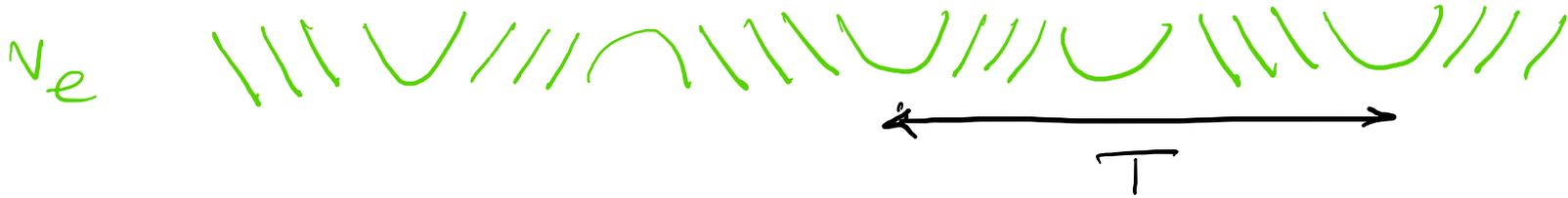
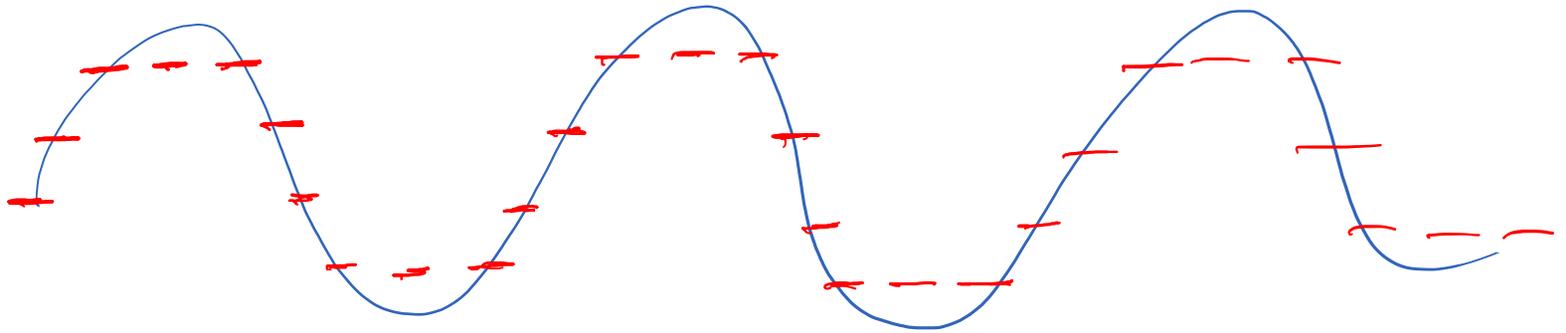
e.g., SINUSOIDAL INPUT



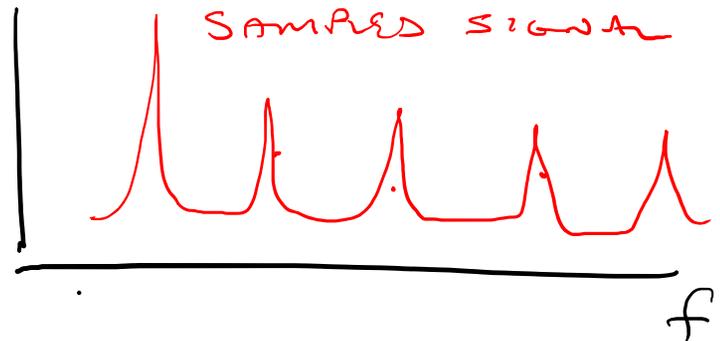
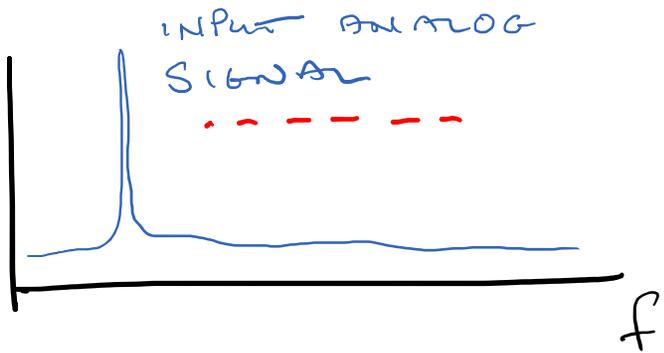
V_e
 $V_a - \hat{V}$

ERROR SIGNAL IS PERIODIC

→ SPURIOUS HARMONICS @ HIGH FREQUENCY [MULTIPLES OF INPUT FREQUENCY]

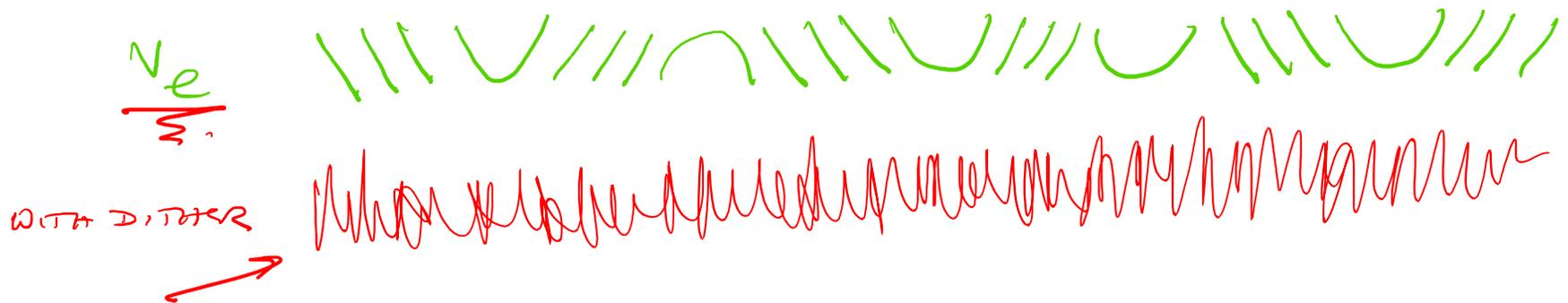
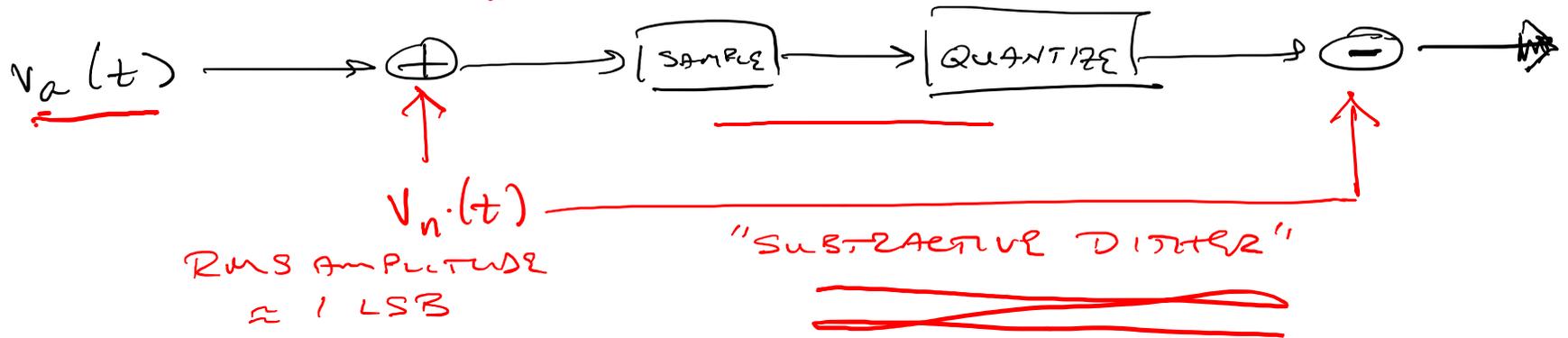


FOURIER DOMAIN'S

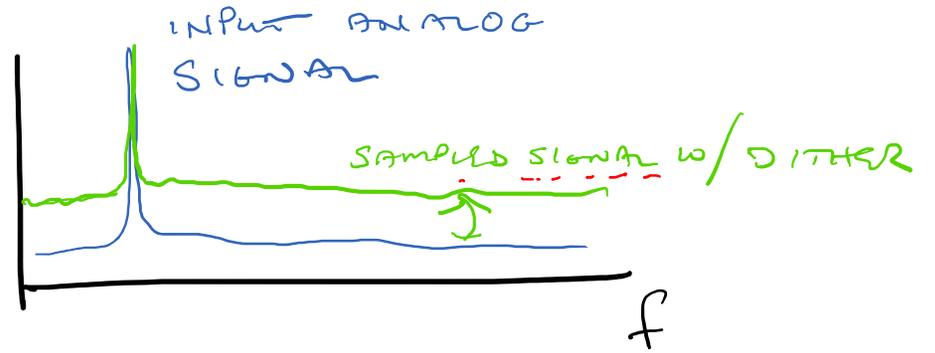
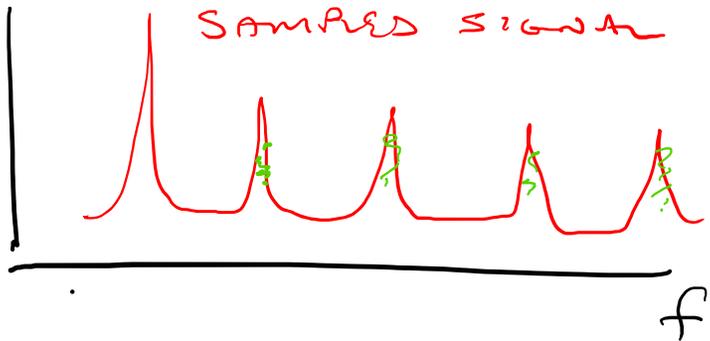


Solution?

ADD NOISE BEFORE QUANTIZATION
"DITHERING"

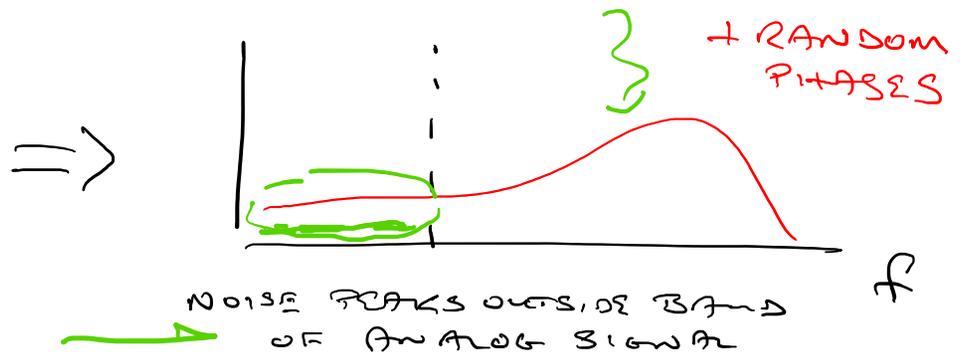


FOURIER DOMAINS



ALTERNATIVELY, SHARED NOISE

INSTEAD OF



ADC / DAC HARDWARE

→ INTERFACING TO EXPERIMENT

→ WAVEFORM GENERATION

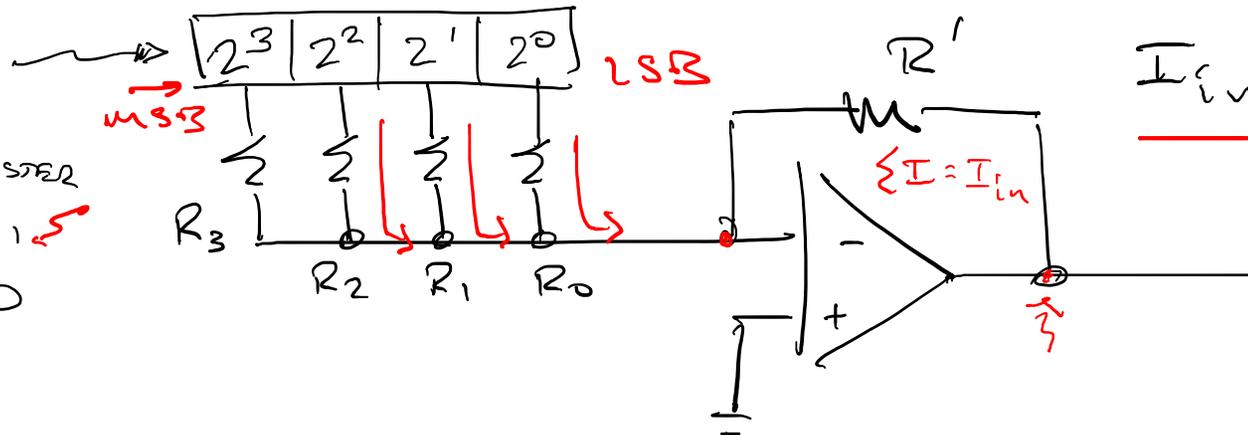
→ DATA ACQUISITION + STORAGE

→ A/D FOR NOISE-FREE TRANSMISSION OF ANALOG INFORMATION

~~ADC~~

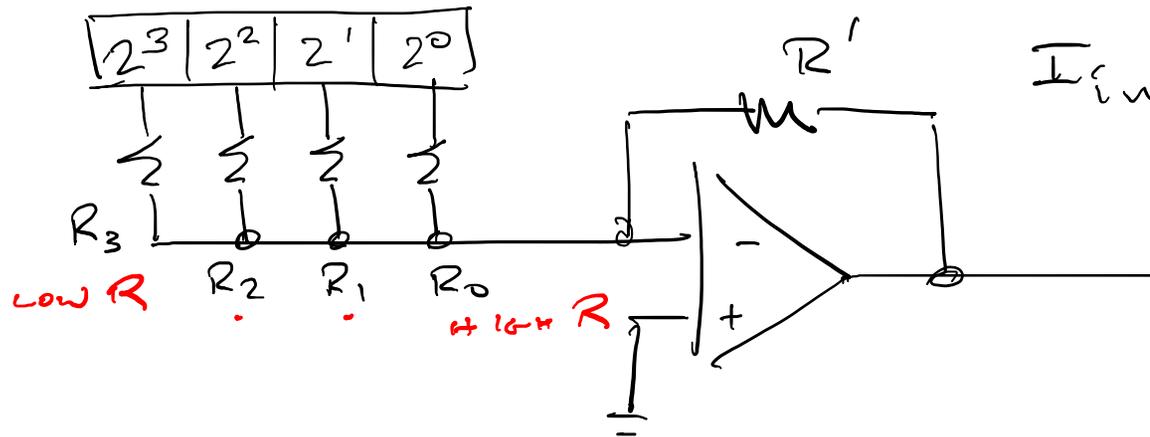
DAC → CONVERT BINARY INPUT CODE TO ANALOG OUTPUT

eg.
4 BITS
 SHIFT REGISTER
 "HIGH" = +1
 "LOW" = 0



$$I_{in} = \frac{V_3}{R_3} + \frac{V_2}{R_2} + \frac{V_1}{R_1} + \frac{V_0}{R_0}$$

$$V_x = 0, 1$$



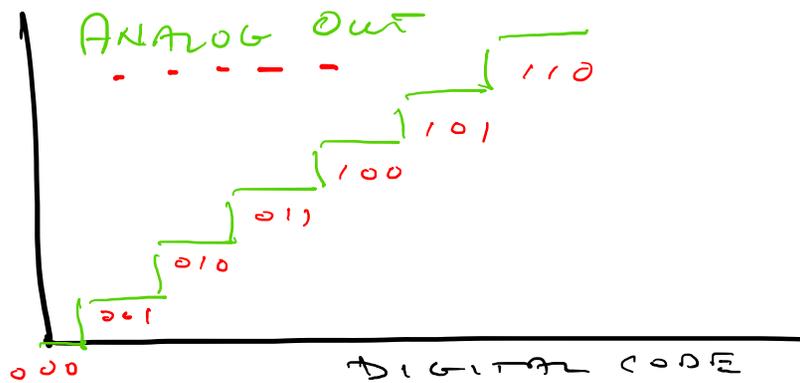
$$I_{in} = \frac{V_3}{R_3} + \frac{V_2}{R_2} + \frac{V_1}{R_1} + \frac{V_0}{R_0}$$

$$V_x = 0, 1$$

CHOOSE $R_0 = 2R, \dots, R_x = 2R_{x+1}$

→ RESISTORS INVERSELY PROPORTIONAL TO BINARY WEIGHTS

$$\underline{R_0 = R} ; R_1 = \frac{R}{2} ; R_2 = \frac{R}{4} \dots \underline{R_x = \frac{R}{2^x}}$$



$$|V_{out, max}| = (2^n - 1) \frac{R'}{R}$$

$$\text{SMALLEST INCREMENT} = \frac{|V_{out, max}|}{2^n - 1}$$

e.g. 8-BIT DAC, 10 V FS RANGE

$$\Delta V = \frac{10V}{255} \approx 39 \text{ mV}$$

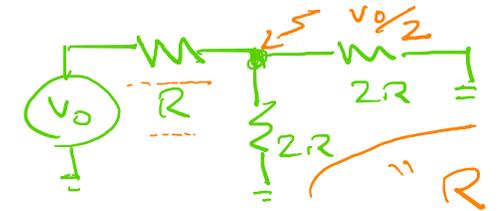
CONCLUSIONS :

- ① PRECISION LIMITED BY LSB
 - ② PRECISION ALSO DETERMINED BY PRECISION OF R_n RESISTOR [SMALLEST R_x , CORRESPONDING TO MSB]
- FOR MORE SIGNIFICANT BITS, NEED VERY HIGH PRECISION FOR SMALL RESISTOR VALUES

e.g. : 10 BITS → NEED 1:1000 PRECISION ON R FOR MSB

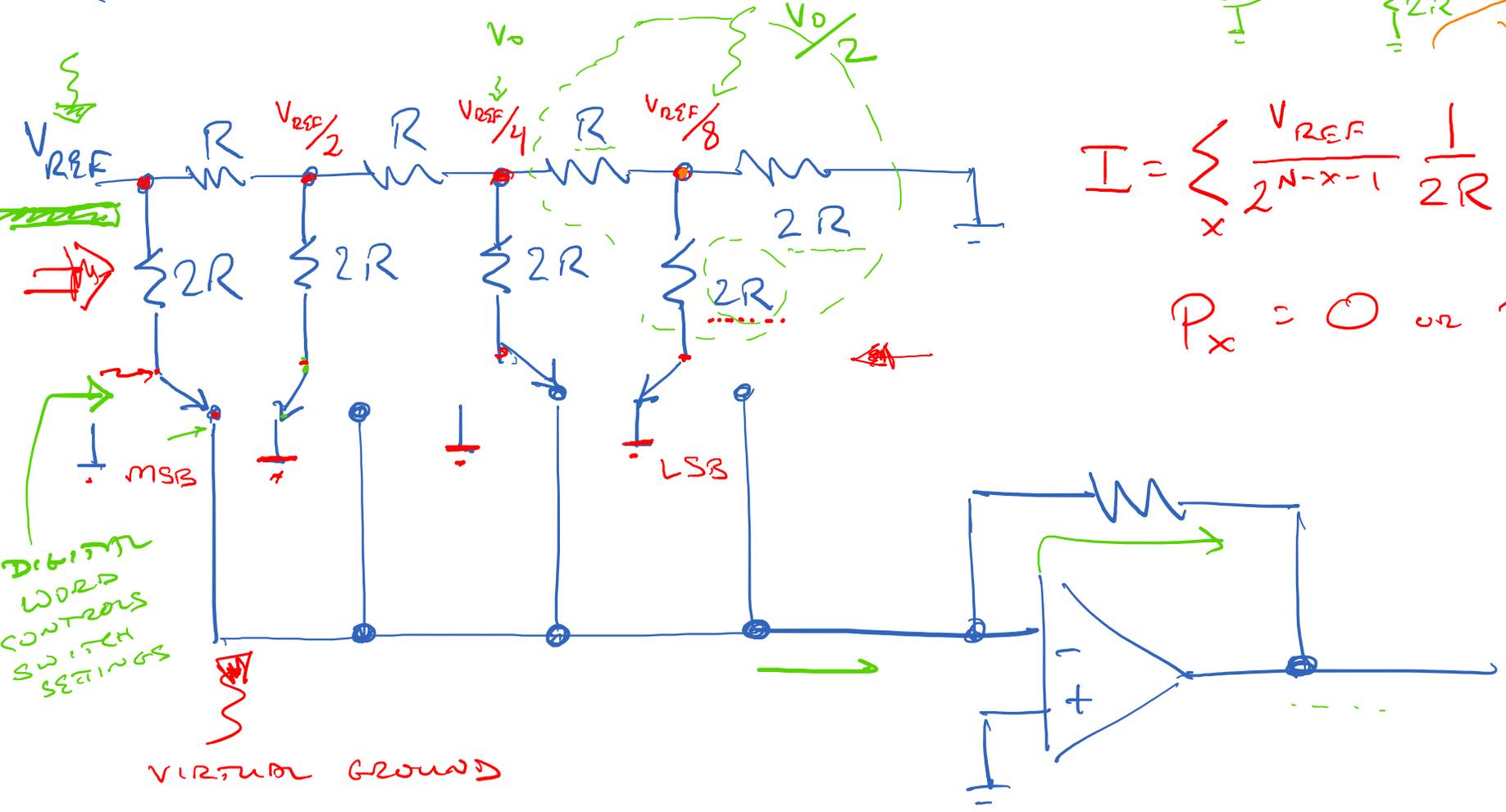
→ DIFFICULT IN PRACTICE

BETTER: R/2R LADDER



$$I = \sum_x \frac{V_{REF}}{2^{N-x-1}} \frac{1}{2R} P_x ;$$

$$P_x = 0 \text{ or } 1$$



DIGITAL WORD
CONTROLS
SWITCH
SETTINGS

VIRTUAL GROUND

AD CONVERSION

e.g. Audio $\approx 20 \text{ kHz} \rightarrow 20 \text{ kHz}$

CD: 44 kHz SAMPLING RATE
2 x 16 bit CHANNELS [\approx 1:65K RESOLUTION]

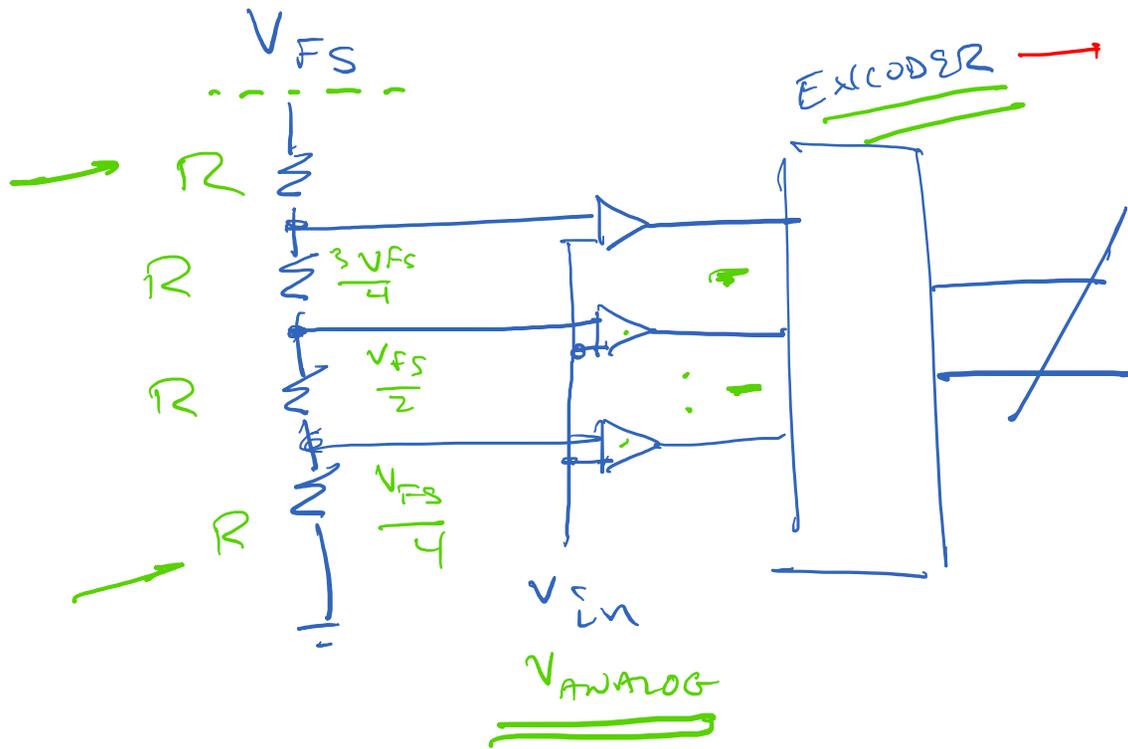
MP3: 16 kHz SAMPLING RATE
8 bit CHANNEL 1:256 RESOLUTION

TRADEOFF IN ADC: SPEED vs RESOLUTION
BANDWIDTH DYNAMIC RANGE

OTHER CONSIDERATIONS: LINEARITY, ABSOLUTE ACCURACY, ETC.

FAST: PARALLEL ENCODING / FLASH ADC

n BITS: DIVIDE V_{FS} INTO 2^n EQUAL INTERVALS
[REQ. $2^n - 1$ COMPARATORS]



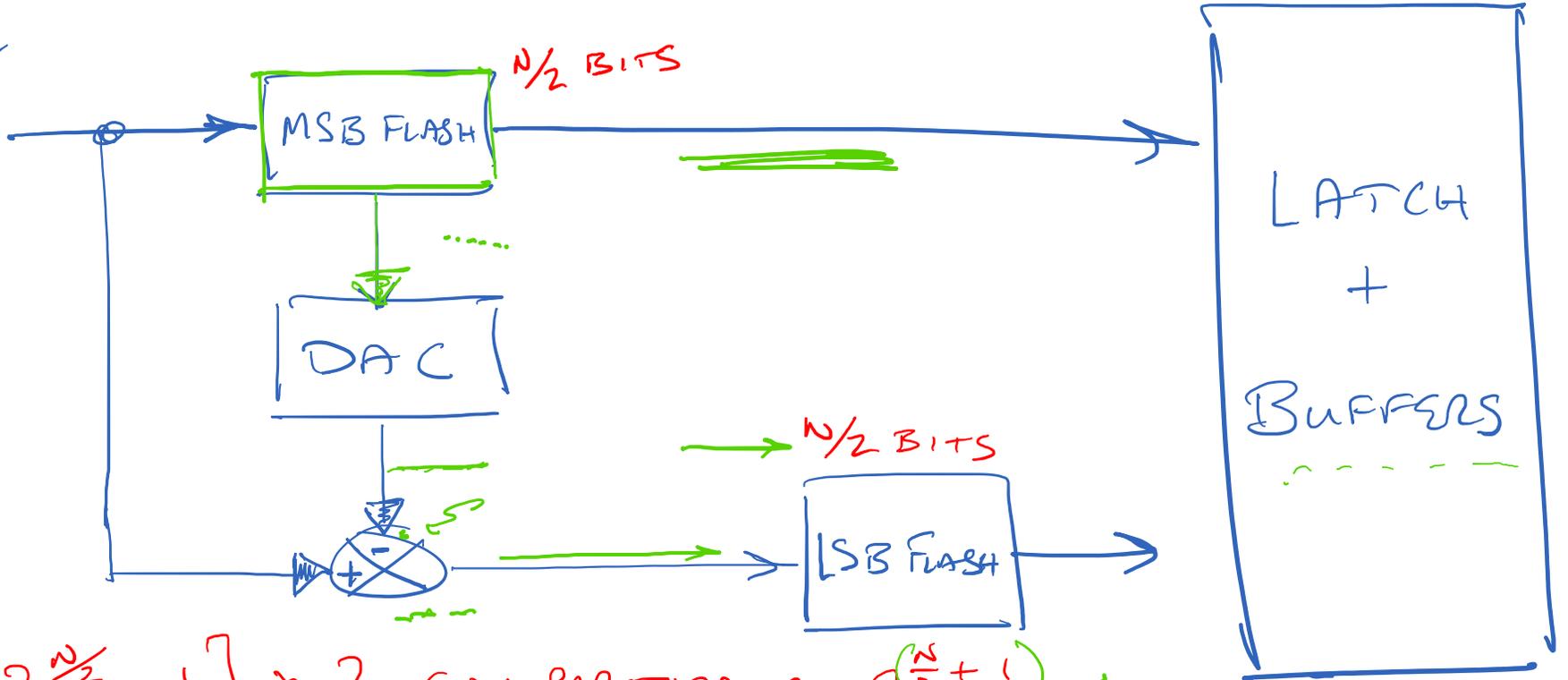
ENCODER → RETURN BINARY ADDRESS OF HIGHEST COMPARATOR THAT IS ASSERTED

2 BIT OUTPUT

~ 20 ns PROPAGATION DELAY
[COMPARATORS + ENCODER]

HIGHER RESOLUTION, SLIGHTLY SLOWER
 " 1/2 FLASH "

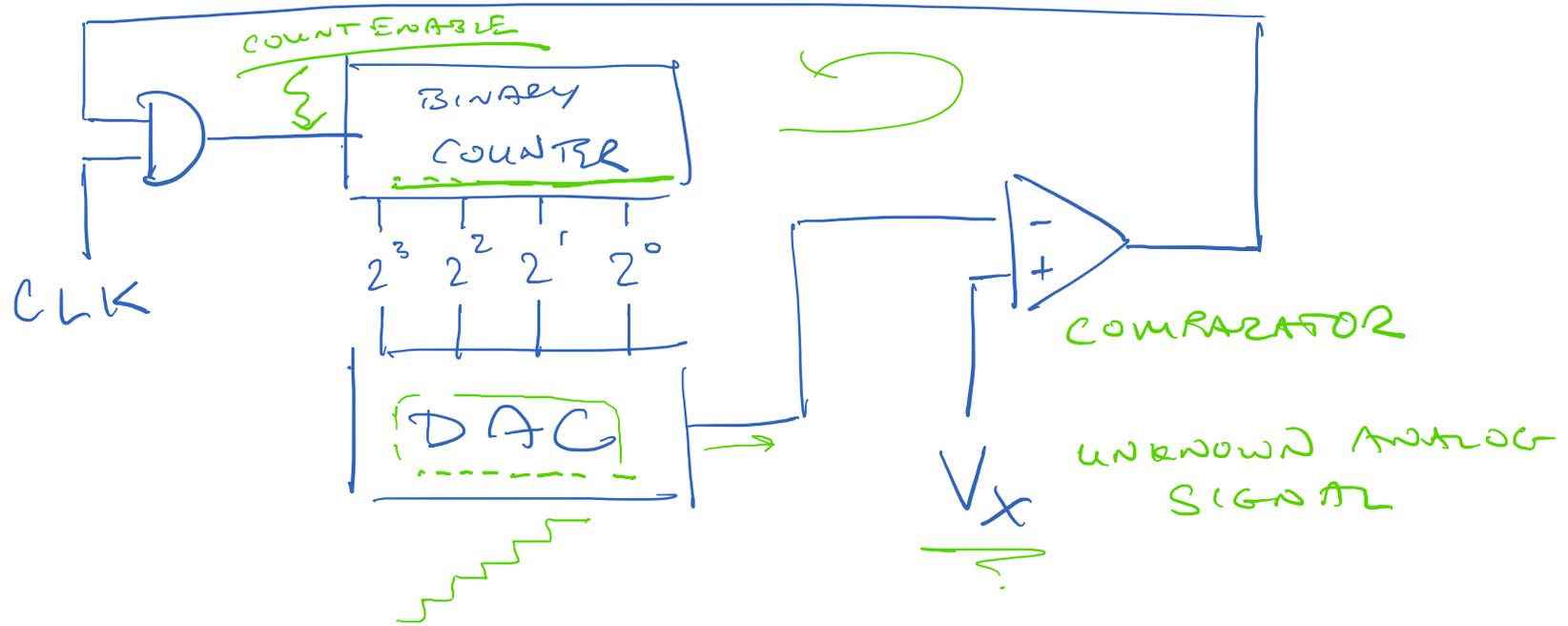
ANALOG DATA



$[2^{N/2} - 1] \times 2$ COMPARATORS $\approx 2^{(N/2 + 1)}$

$\approx 1/2$ AS FAST AS FULL FLASH

IN LAB : SIMPLE ADC BASED ON
BINARY COUNTER + DAC IN FEEDBACK LOOP



VARIANT OF SINGLE-SLOPE ADC