



*Ingegneria delle Telecomunicazioni*  
Satellite Communications

## 21. TOMTOM and beyond - GNSS Receiver Design

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## His(Her) Majesty TomTom GO, 2004

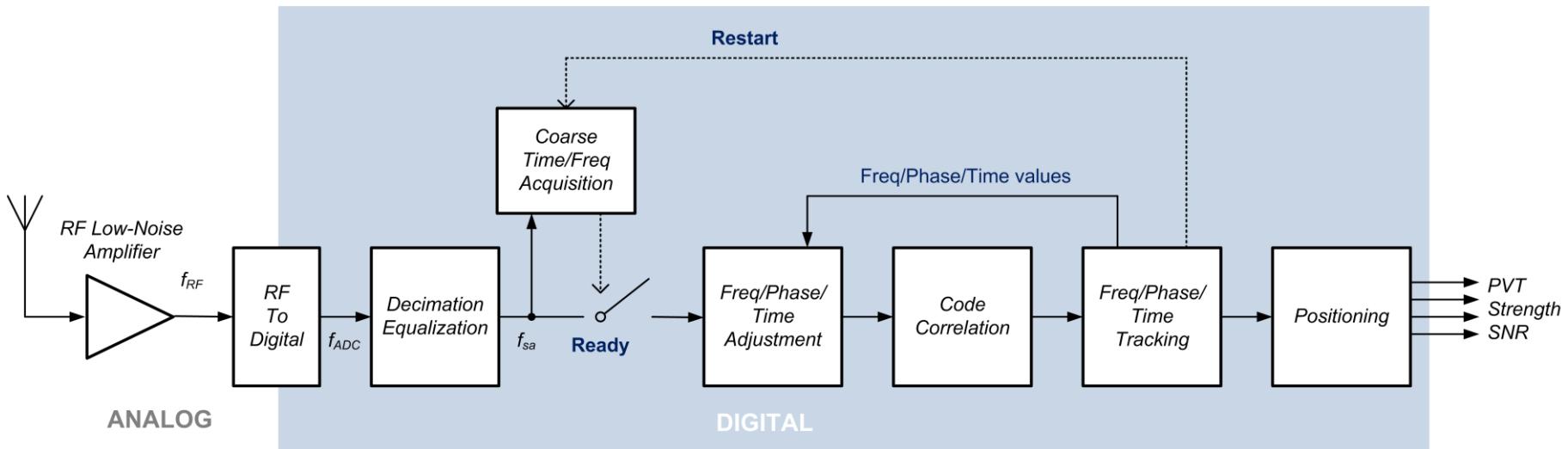
- GPS receiver
- Digital Maps
- Navigation SW
- User-Friendly GUI
- Voice Directions
- Touch-screen Display



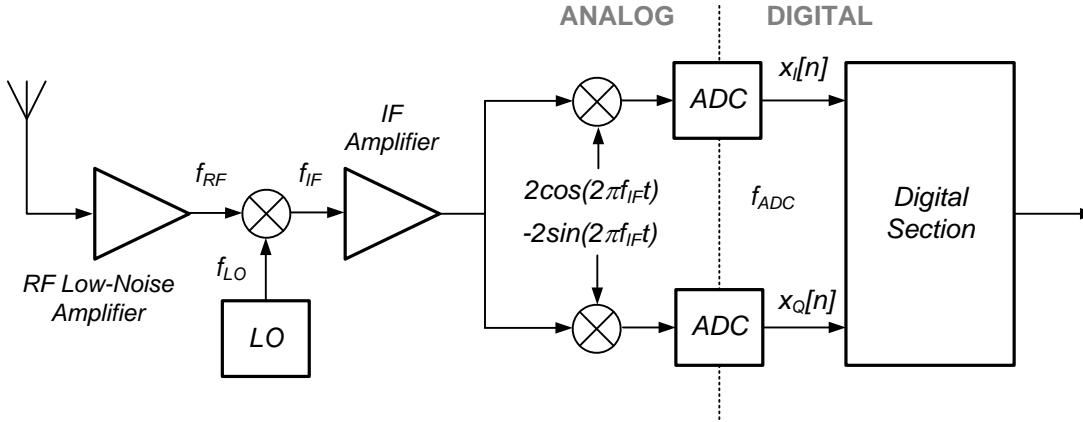


# Basic Receiver Architecture

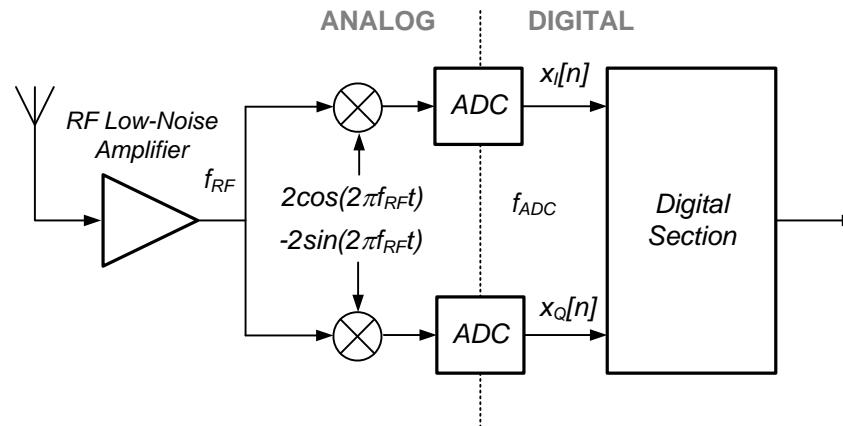
- Modern receivers perform all of the signal processing functions with DSP components and algorithms



# Form RF to Digital - I

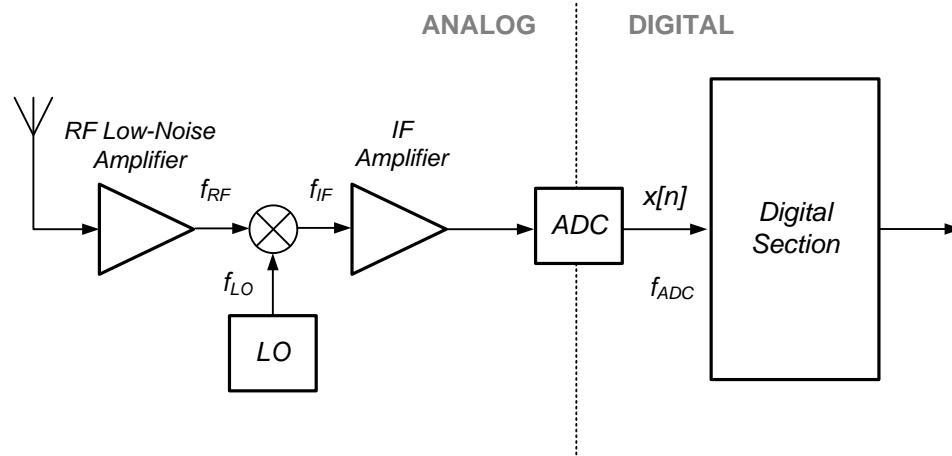


- **I/Q BaseBand Sampling with IF Conversion**

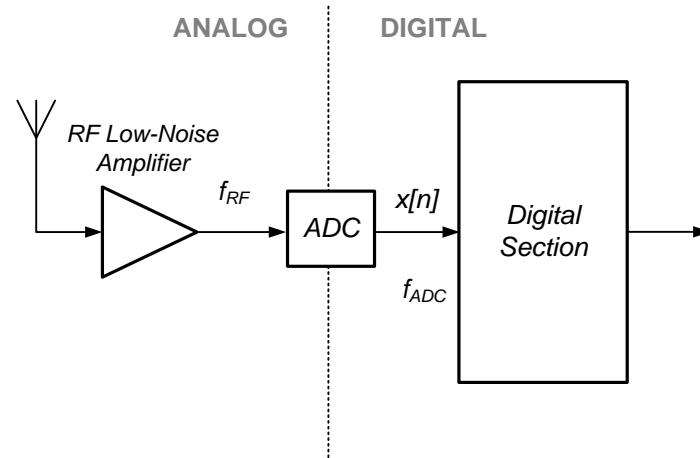


- **I/Q BaseBand Sampling with Direct Baseband Conversion (aka ZERO-IF Architecture)**

## Form RF to Digital - II



- **IF BandPass Sampling with Digital IF**

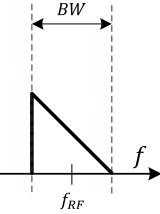
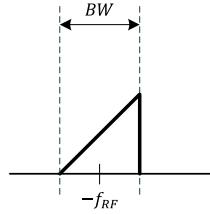


- **Direct RF Sampling with Digital IF**



# Band-Pass Sampling

- *Pass-band, real signal*
  - *RF (passband) bandwidth BW*
  - *carrier frequency  $f_{RF}$*



**HERMITIAN  
SYMMETRIC**

$$X(-f) = X(f)^*$$

- *Nyquist-sampling condition:*

$$f_s \geq 2B = 2\left(f_{RF} + \frac{BW}{2}\right)$$
- *For navigation signals in L-band, this is unrealistic!*

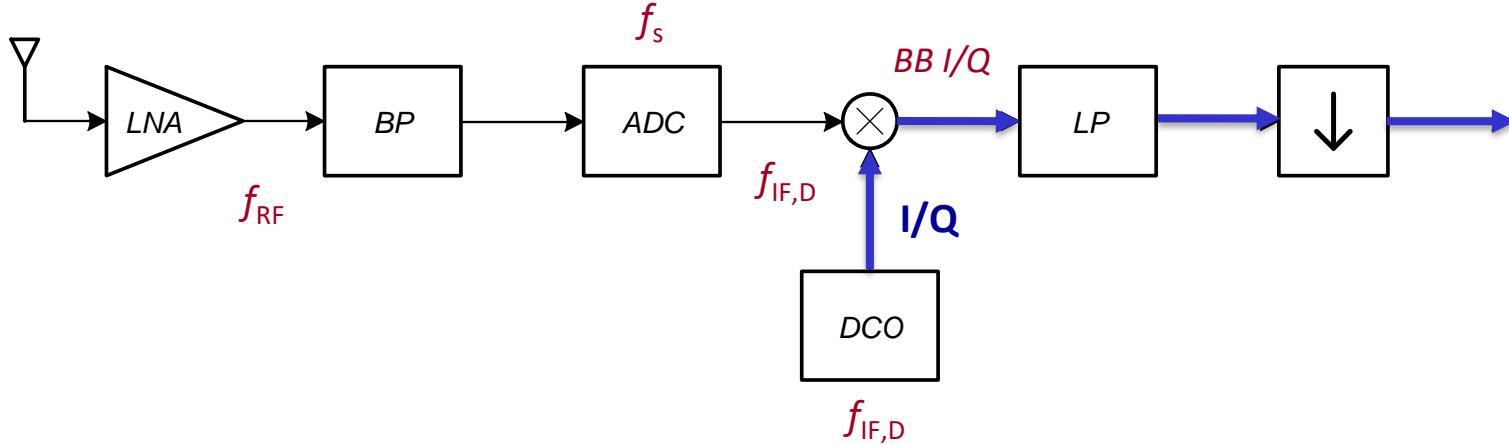
$$f_{RF} \in [1 \text{ GHz}, 2 \text{ GHz}] \quad \Rightarrow \quad f_s > 2 \div 4 \text{ GHz}$$
- *Solution: band-pass sampling at reduced rate :*

$$f_s \approx 2BW$$



# Direct Sampling Front-End Architecture

- Front-end architecture of direct-sampling receiver

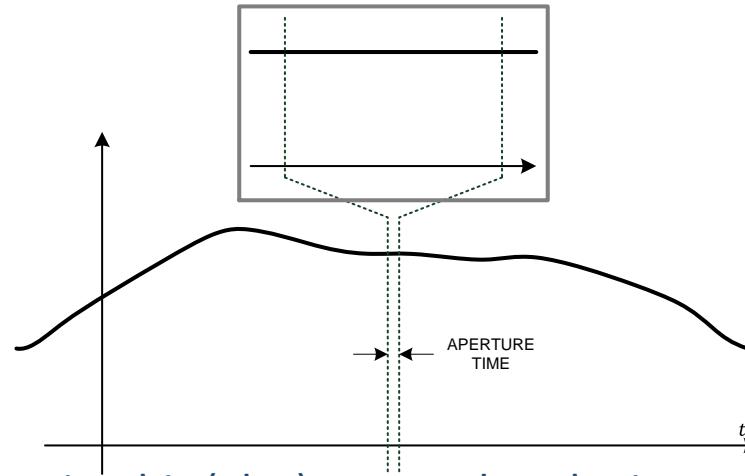


- RX signal from antenna is sampled **directly after LNA** and RF BP filter
  - no RF-to-IF downconversion stage / all-digital processing
- **Digital IF**  $f_{IF,D}$  resulting from the interplay between  $f_{RF}$  and  $f_s$  (later on)
- **Digital I/Q** oscillator to convert the received signal at  $f_{IF,D}$  to I/Q base-band
- low-pass filter to isolate the channel of interest
  - filter bandwidth is BW
- base-band decimation
  - base-band signal may be oversampled for many applications

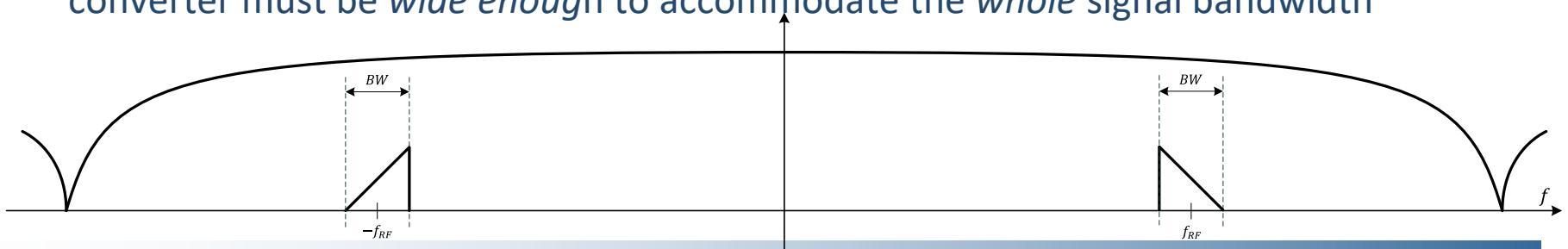


# Direct Sampling Requirements

- Any A/D converter is characterized by the *aperture time*, i.e., the time over which the signal must be stable (constant) to compute the digital output
  - it must be short enough to have correct signal sampling (constant level)



- In the frequency domain, this (also) means that the input bandwidth of the A/D converter must be *wide enough* to accommodate the *whole* signal bandwidth



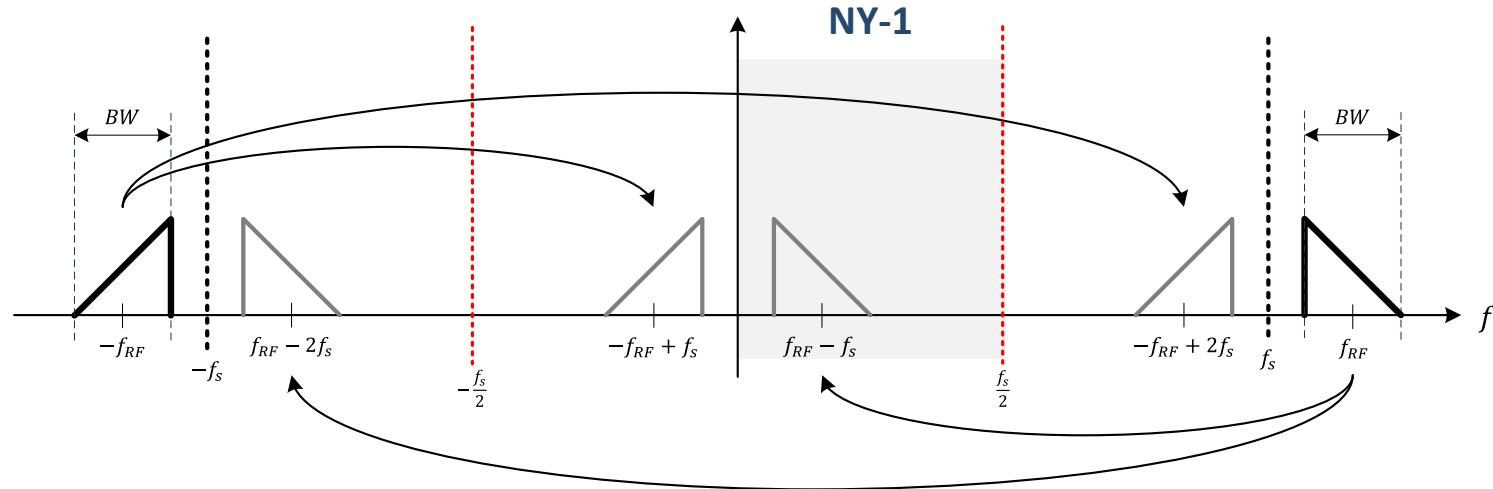


- $f_{RF} = 5 \text{ MHz}$ , BW=1 MHz
  - Conventional, Nyquist-rule  $f_s = 2 * (5 + 0.5) = 11 \text{ MHz}$
- Try  $f_{RF} = 4 \text{ MHz}$ 
  - ... and look at the digital bandwidth  $-2 \text{ MHz} < f < 2 \text{ MHz}$
  - Find  $f_{IF,D}$
- Then try  $f_{RF} = 7 \text{ MHz}$ 
  - ... and look at the digital bandwidth  $-3.5 \text{ MHz} < f < 3.5 \text{ MHz}$
  - Find  $f_{IF,D}$
  - Can you guess what happened ?

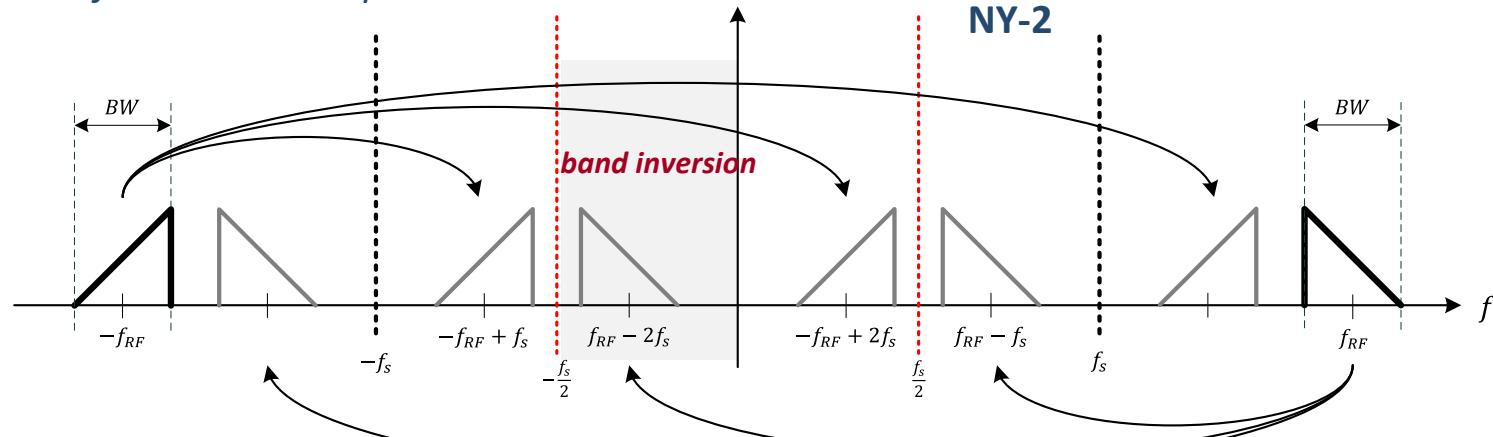


# Pass-Band Sampling: Band Inversion

- case I: fundamental replica in NY-1 zone:

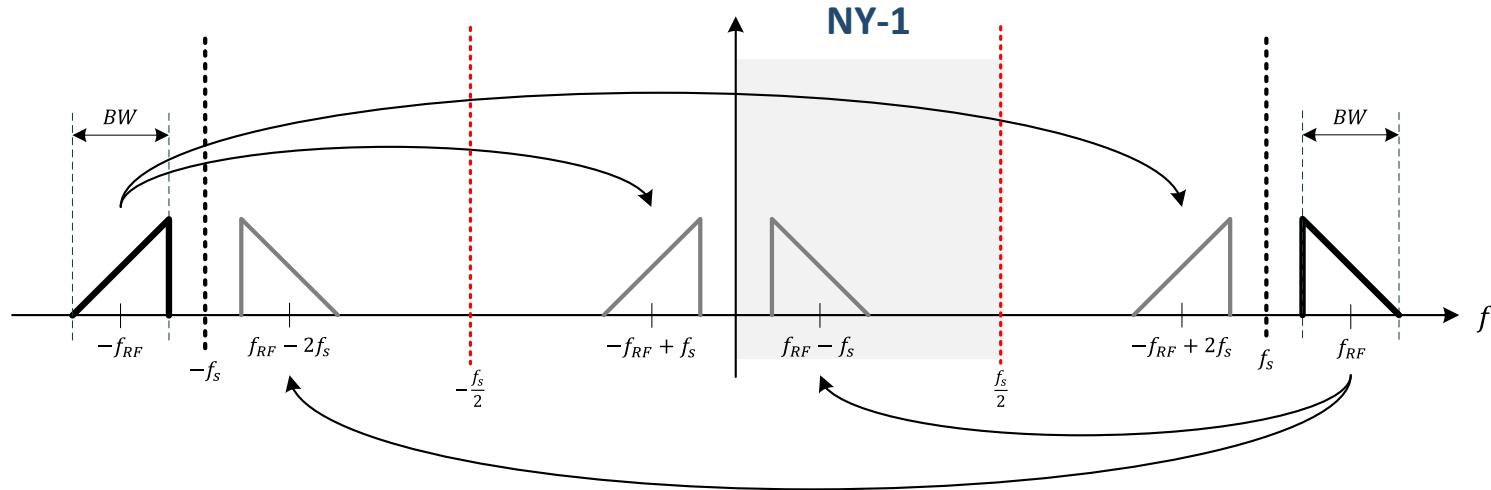


- case II: fundamental replica in NY-2 zone:



# Pass-Band Sampling 1

- case I: fundamental replica in NY-1 zone:

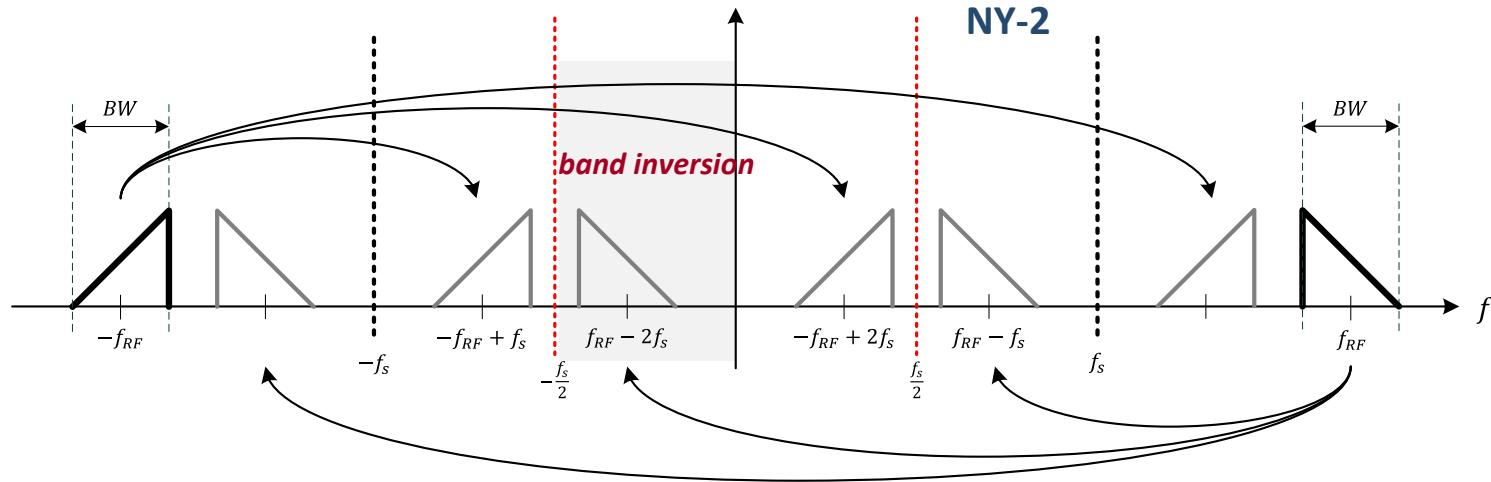


- The resulting Digital IF is derived from the relation  $f_{IF,D} = f_{RF} - mf_s$  where  $m$  is integer and is such that  $f_{IF,D}$  lies in the *POSITIVE HALF* of the digital bandwidth,  $0 \leq f < f_s/2$

$$f_{IF,D} = f_{RF} - mf_s \quad \Rightarrow \quad f_{IF,D} = |f_{RF}|_{f_s}$$

## Pass-Band Sampling 2: Band Inversion

- case II: fundamental replica in NY-2 zone:



The resulting Digital IF is again derived from the relation  $f_{IF,D} = f_{RF} - mf_s$  where  $m$  is integer, but now it is in the *NEGATIVE HALF* of the digital bandwidth  $-f_s/2 \leq f < 0$

$$f_{IF,D} = f_{RF} - mf_s \quad \Rightarrow \quad f_{IF,D} = |f_{RF}|_{f_s} - f_s$$

where the  $-f_s$  term has the purpose of re-mapping the “modulus- $f_s$ ” function to  $-f_s/2, f_s/2$  instead of  $0, f_s$  as in the mathematical definition (leading to a digital IF out of the digital bandwidth)

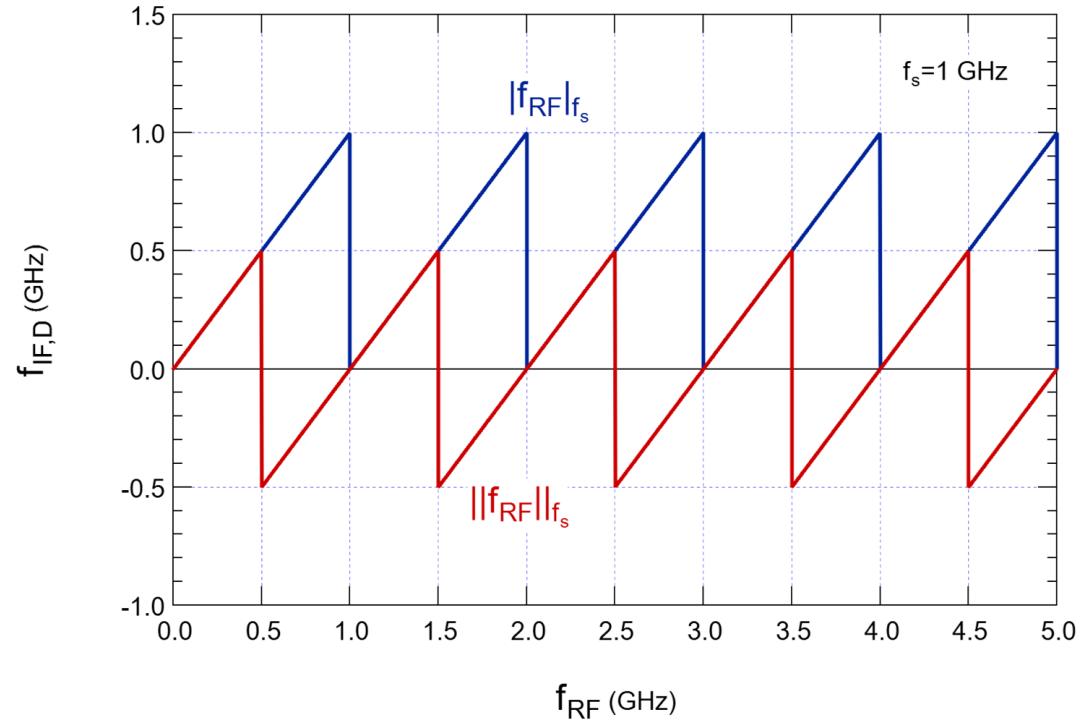
# Pass-Band Sampling: Computing $f_{IF,D}$

We define a “modified”, symmetrical modulus function

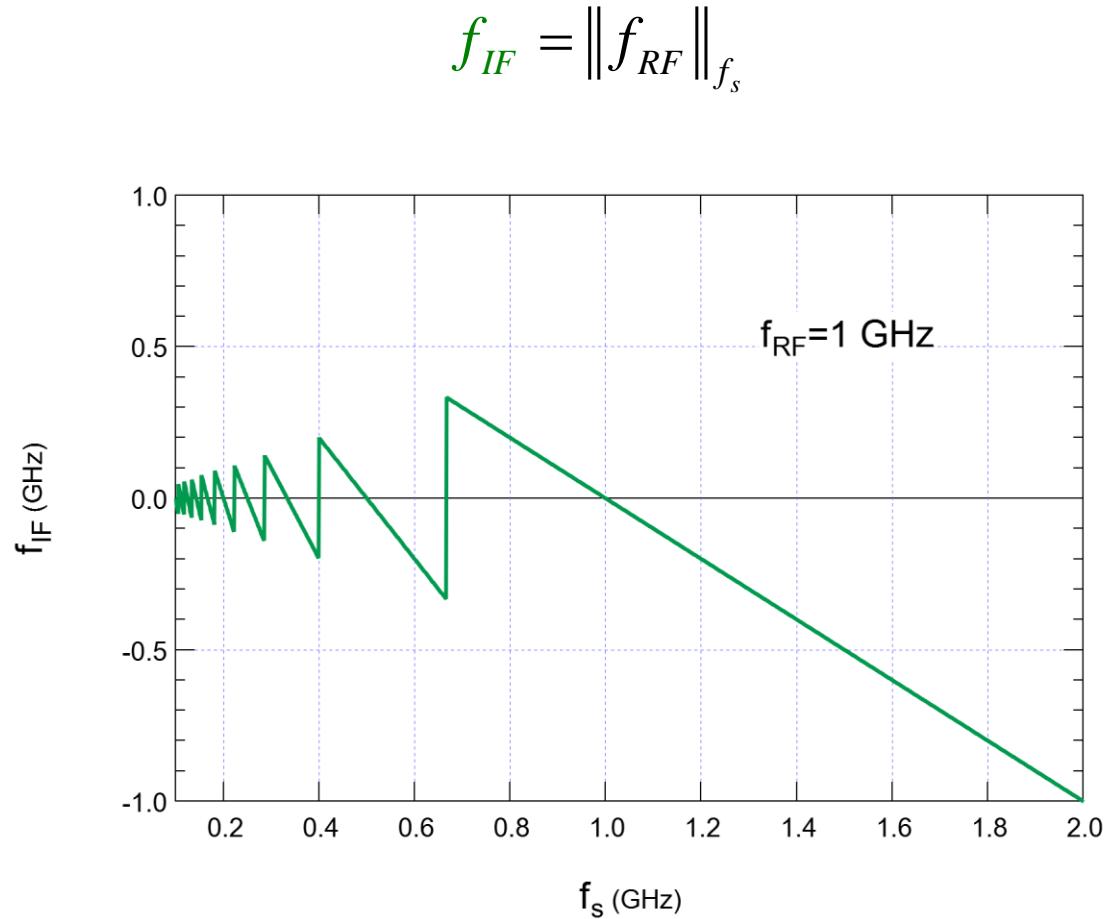
$$f_{IF,D} = \|f_{RF}\|_{f_s} \triangleq \left| f_{RF} + \frac{f_s}{2} \right|_{f_s} - \frac{f_s}{2}$$

where the  $-f_s/2$  term has the purpose of re-mapping the “modulus- $f_s$ ” function to  $[-f_s/2, f_s/2]$  instead of  $[0, f_s]$  as in the mathematical definition. For short, we will call  $f_{IF,D}$  just  $f_{IF}$

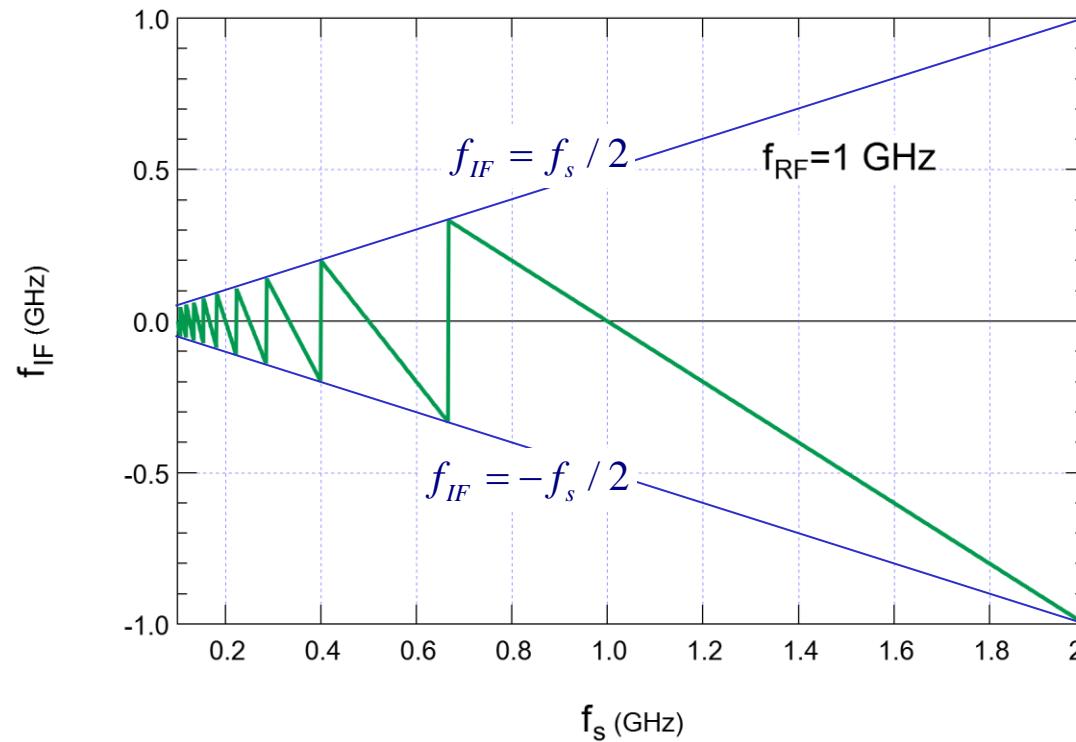
$$f_{IF,D} = \begin{cases} |f_{RF}|_{f_s} & , \quad |f_{RF}|_{f_s} \leq f_s / 2 \\ |f_{RF}|_{f_s} - f_s & , \quad |f_{RF}|_{f_s} > f_s / 2 \end{cases}$$



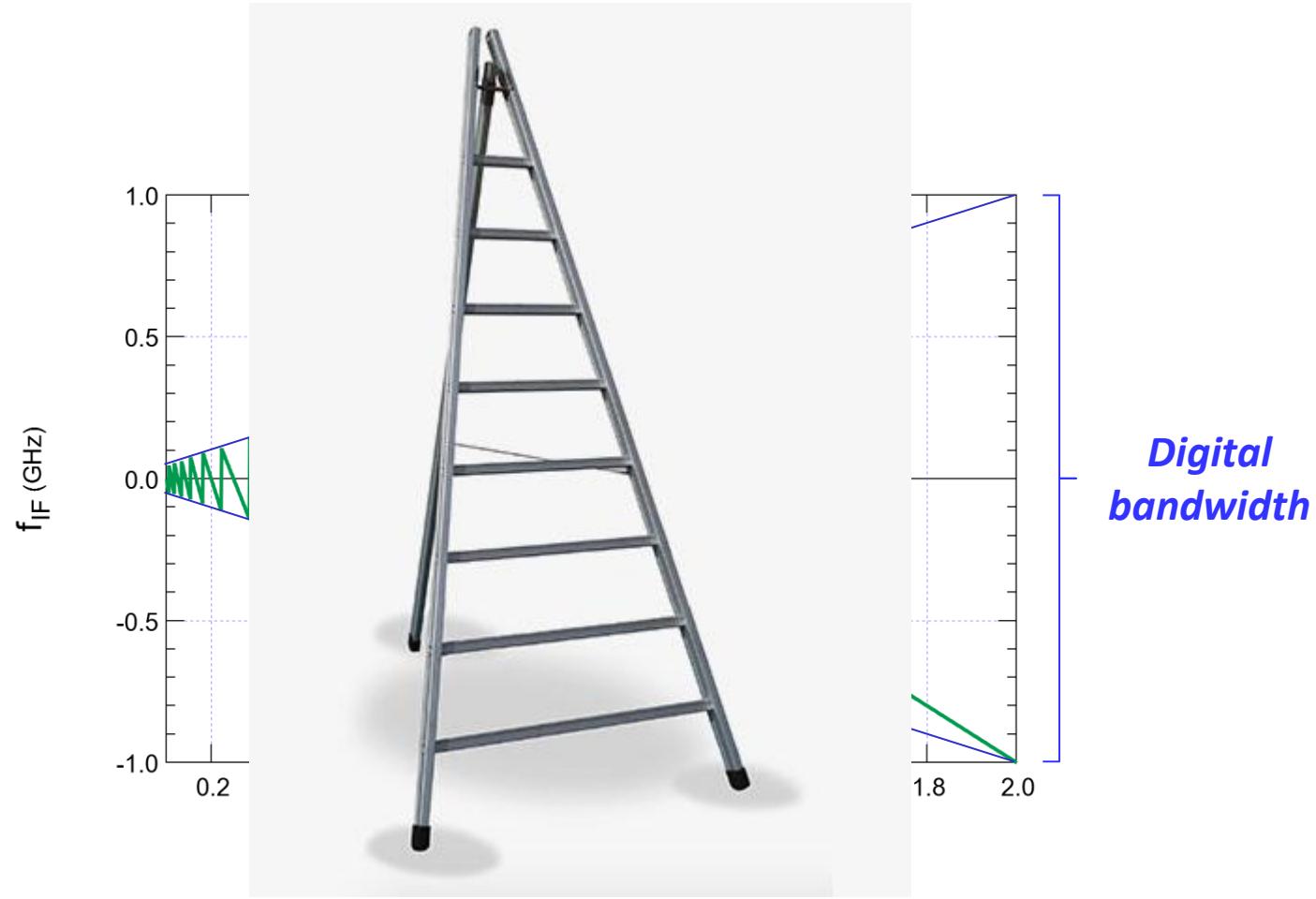
# The “Modified Modulus” as a function of $f_s$



# The “Modified Modulus” as a function of $f_s$



# The “Modified Modulus” as a function of $f_s$



# The Ladder Diagram I

- Digital IF frequency:

$$f_{IF} = \|f_{RF}\|_{f_s} = \left| f_{RF} + \frac{f_s}{2} \right| - \frac{f_s}{2} \quad (\text{Symmetric Modulus Function})$$

- Case-I (NY-1):

$$0 \leq f_{IF} < \frac{f_s}{2}$$

- Just frequency conversion

- Case II (NY-2):

$$-\frac{f_s}{2} \leq f_{IF} < 0$$

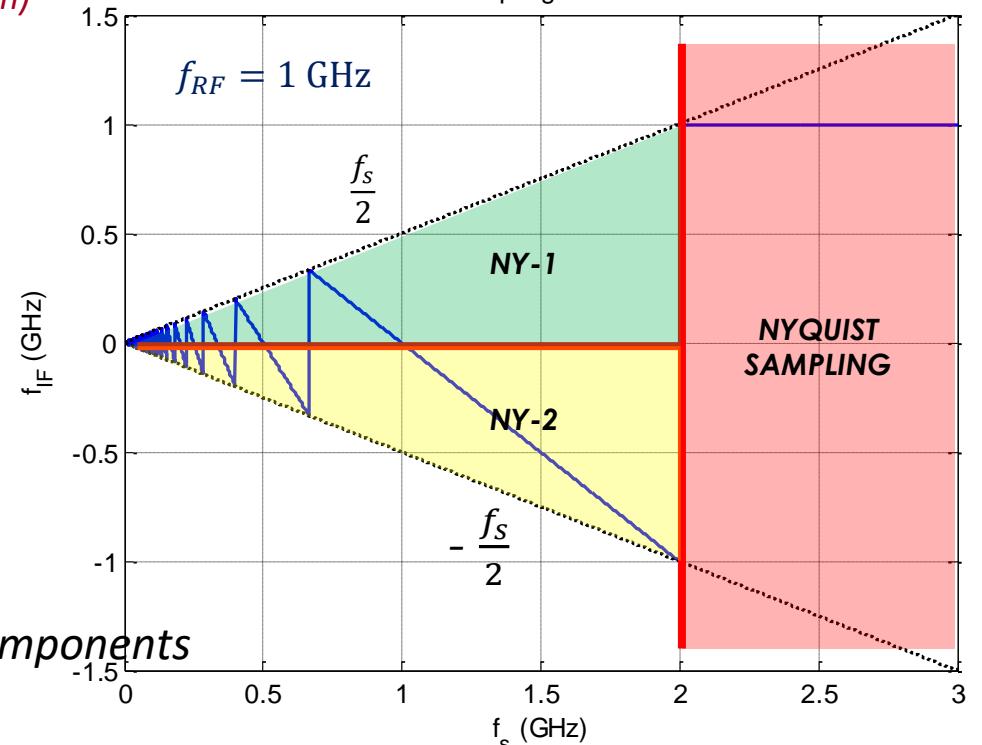
- Frequency conversion AND **band inversion**, i.e.,  $f_{IF}$  can be considered positive but the Q components is sign-changed

- OK with  $f_{IF}$ , but what is the relation of  $f_s$  with the (bandpass) signal BW ?

case study:

$$f_{RF} = 1 \text{ GHz}$$

direct sampling: local carrier



# Updated Nyquist Rule

- Selection of sampling rate
  - *no aliasing within the signal bandwidth*

$$\begin{cases} f_{IF} + \frac{BW}{2} \leq \frac{f_s}{2} \\ f_{IF} - \frac{BW}{2} \geq 0 \end{cases}$$

$$f_{IF} > 0$$

$$\begin{cases} f_{IF} + \frac{BW}{2} \leq 0 \\ f_{IF} - \frac{BW}{2} \geq -\frac{f_s}{2} \end{cases}$$

$$f_{IF} < 0$$

$$\begin{cases} |f_{IF}| \geq \frac{BW}{2} \\ |f_{IF}| \leq \frac{f_s}{2} - \frac{BW}{2} \end{cases}$$

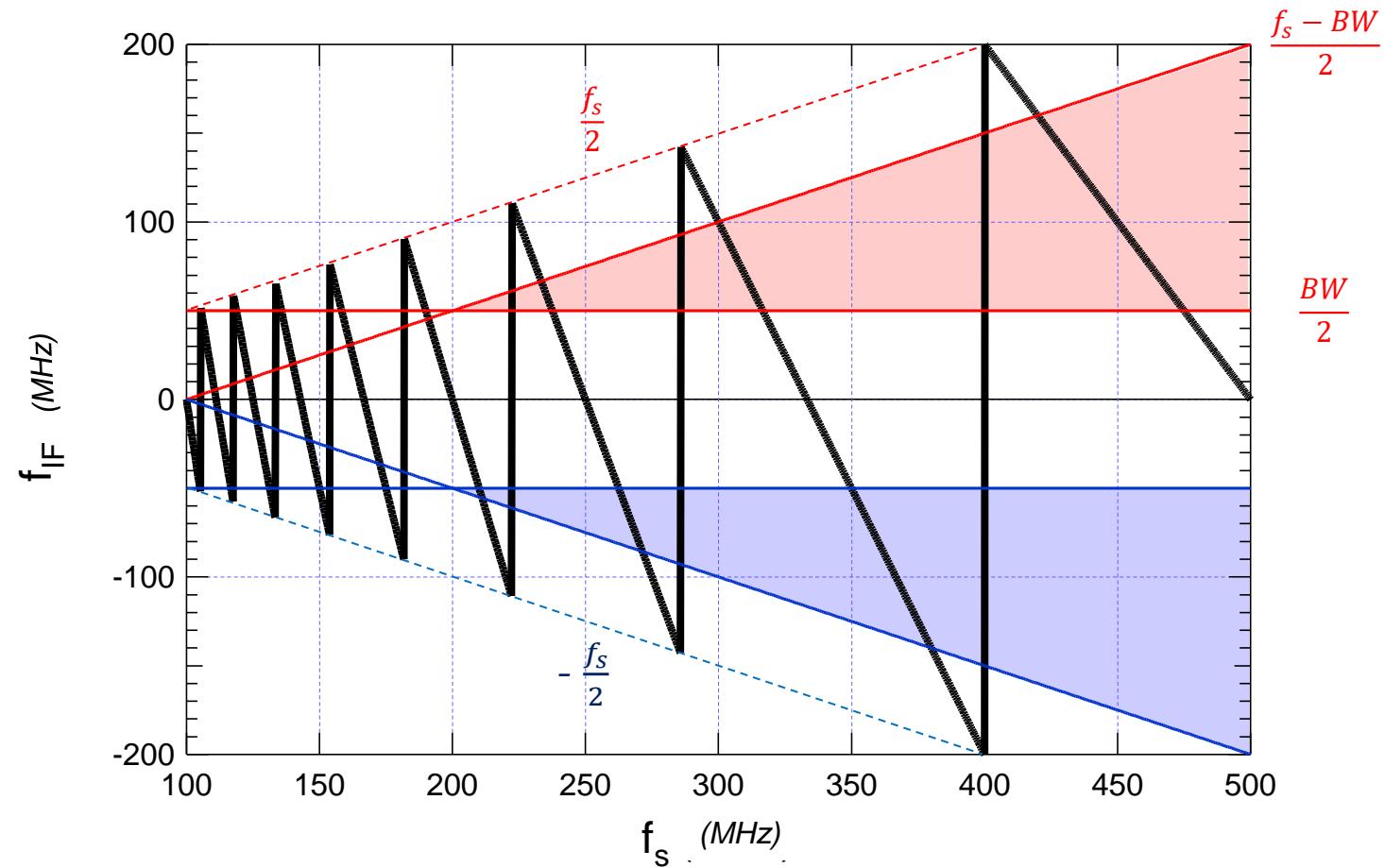


$$\frac{BW}{2} \leq |f_{IF}| \leq \frac{f_s}{2} - \frac{BW}{2}$$

- We can visualize the inequality on the Ladder diagram to actually find the allowed values of the digital carrier  $f_{IF}$  AND of the sampling frequency  $f_s$

## The Ladder Diagram II

- Case study:  $f_{RF}=1 \text{ GHz}$  and  $BW=100 \text{ MHz}$

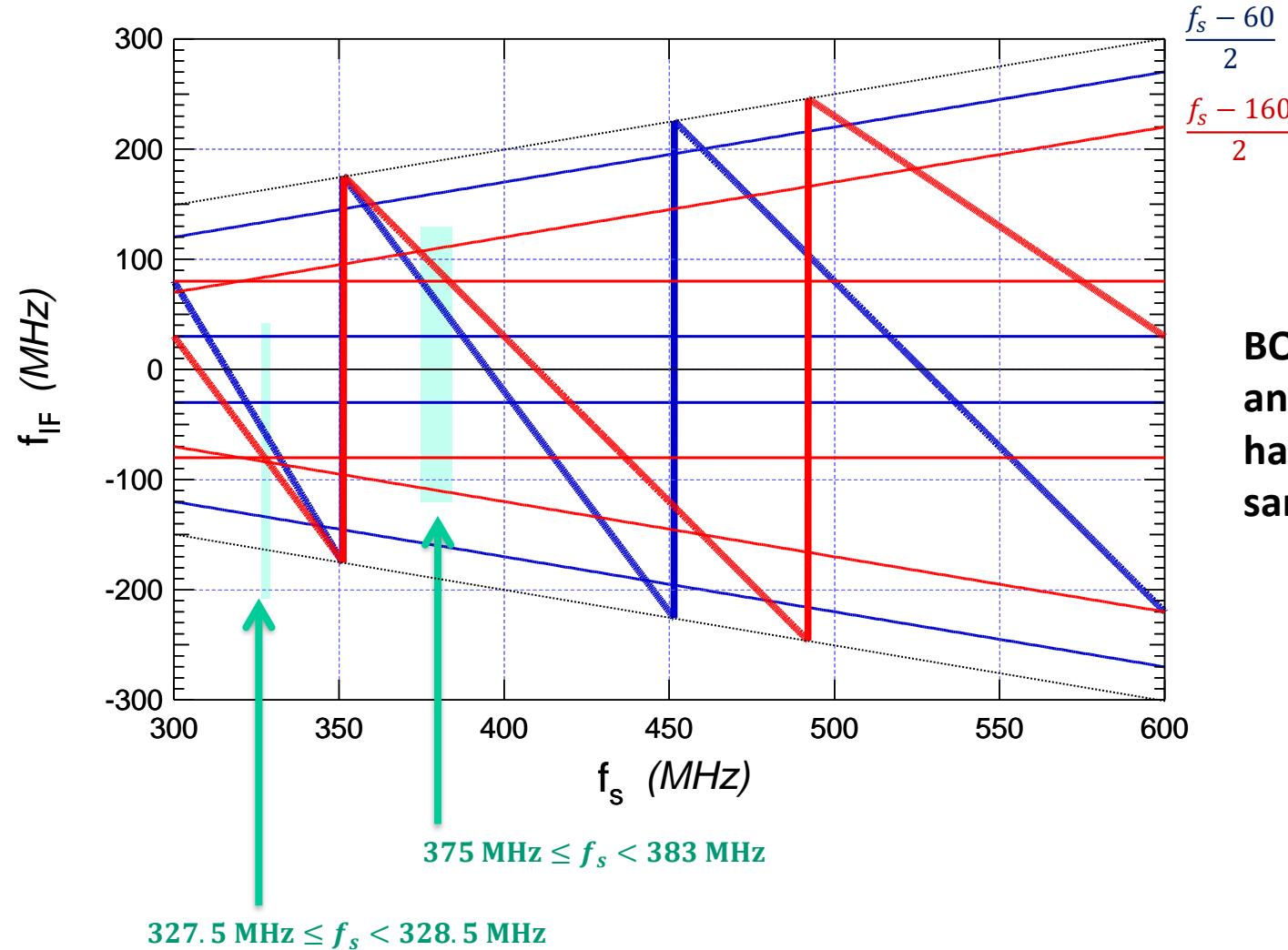


# Sample Design: GNSS Direct-Sampling Front-End

N.	RIMS V3 Req	Satellite System	Band	Lower/Upper	Signal	Carrier (MHz)	Carrier_BB (MHz)	ID_Comb_Carrier-BW	ID_BW_RF	BW @ RF (MHz)
1	Baseline	GALILEO	E5a	Lower	E5aI-E5aQ	1176,45	-135,55	1	1	20,46
2	Baseline	GPS	L5	Lower	I5-Q5	1176,45	-135,55	1	1	20,46
3	Baseline	GEO	L5	Lower	L5I-L5Q	1176,45	-135,55	1	1	20,46
4	Expandability	HEO	L5	Lower	L5I-L5Q	1176,45	-135,55	1	1	20,46
5	Expandability	GALILEO	E5b	Lower	E5bI-E5bQ	1207,14	-104,86	2	1	20,46
6	Optional	BEIDOU	E5b	Lower	B2I	1207,14	-104,86	2	1	20,46
7	Optional	GALILEO	E5	Lower	E5	1191,795	-120,205	3	2	40,92
8	Baseline	GPS	L2	Lower	L2CM/L2CL	1227,6	-84,4	4	1	20,46
9	Baseline	GPS	L2	Lower	L2P	1227,6	-84,4	4	1	20,46
10	Optional	GLONASS	L2	Lower	L2OF	1246	-66	5	3	7
11	Expandability	GALILEO	E6	Lower	E6B-E6C	1278,75	-33,25	6	4	10,23
12	Baseline	GALILEO	E1	Upper	E1B-E1C	1575,42	-64,58	7	5	24,552
13	Baseline	GPS	L1	Upper	L1CA	1575,42	-64,58	8	6	30,69
14	Expandability	GPS	L1	Upper	L1C-D/P	1575,42	-64,58	8	6	30,69
15	Baseline	GEO	L1	Upper	L1I-L1Q	1575,42	-64,58	8	6	30,69
16	Expandability	HEO	L1	Upper	L1I-L1Q	1575,42	-64,58	8	6	30,69
17	Optional	GLONASS	L1	Upper	L1OF	1602	-38	9	7	9
18	Optional	BEIDOU	L1	Upper	B1I	1561,098	-78,902	10	1	20,46

- *Lower-band:*
  - *SIS from 1 to 11 (1150 to 1310 MHz)*
  - $BW_L = 160 \text{ MHz}$ ;  $f_{RF,L} = 1230 \text{ MHz}$
- *Upper-band*
  - *SIS from 12 to 18 (1550 to 1610 MHz)*
  - $BW_U = 60 \text{ MHz}$ ;  $f_{RF,U} = 1580 \text{ MHz}$

# GNSS Direct-Sampling Front-End: Ladder(s)



BOTH ladders (lower and upper bands) has to work at the same time !

# GNSS Direct-Sampling Front-End: Final Choice



- Selection of sampling rate: Ladder diagram
  - requirement: no aliasing of the signal bandwidth

$$\text{NY-1} \begin{cases} f_{IF} + \frac{BW}{2} \leq \frac{f_s}{2} \\ f_{IF} - \frac{BW}{2} \geq 0 \end{cases}$$

$$\text{NY-2} \begin{cases} f_{IF} + \frac{BW}{2} \leq 0 \\ f_{IF} - \frac{BW}{2} \geq -\frac{f_s}{2} \end{cases}$$



$$\begin{cases} |f_{IF}| \geq \frac{BW}{2} \\ |f_{IF}| \leq \frac{f_s}{2} - \frac{BW}{2} \end{cases}$$



*Lowest-frequency solution (NY-2):*

$$327.5 \text{ MHz} \leq f_s < 328.5 \text{ MHz}$$



$$f_s = 328 \text{ MHz}$$

$$\frac{BW}{2} \leq |f_{IF}| \leq \frac{f_s}{2} - \frac{BW}{2}$$

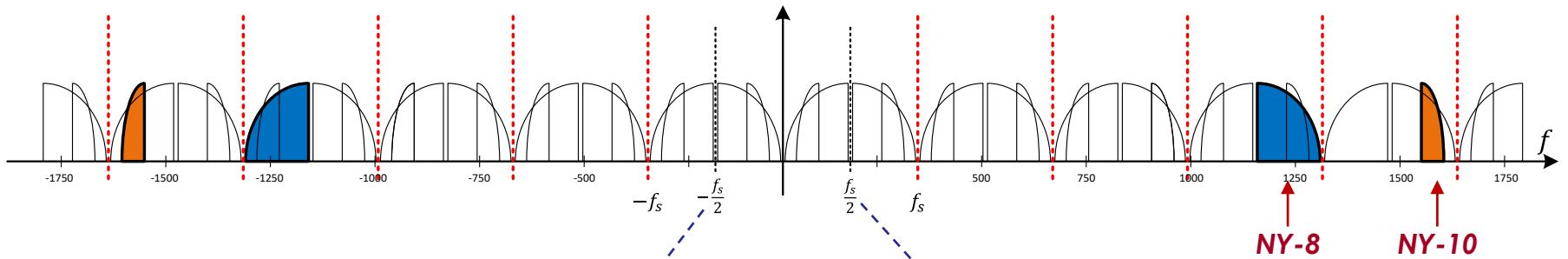
*next solution (NY-1):*

$$375 \text{ MHz} \leq f_s < 383 \text{ MHz}$$

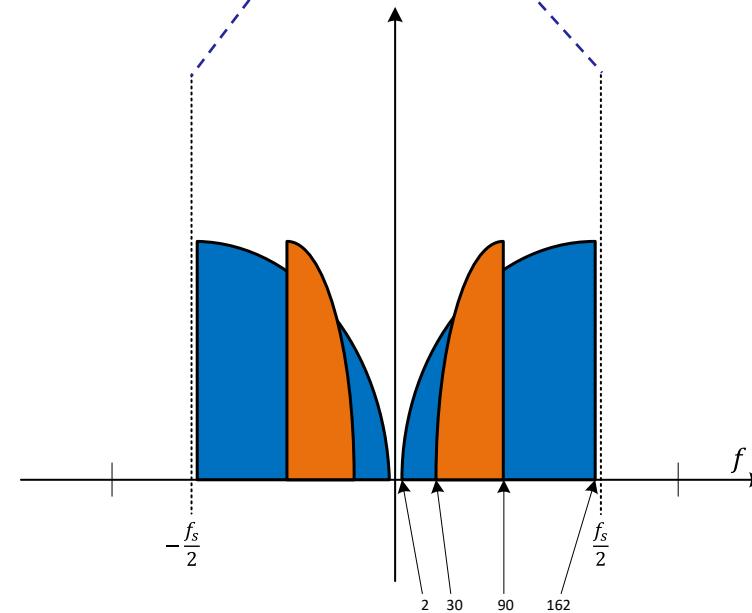


# GNSS Direct-Sampling Front-end: Spectra

- Sampling frequency  $f_s = 328$  MHz



- Upper-band:*
  - $[-90; -30]$  MHz
  - $f_{IF,U} = -60$  MHz
- Lower-band:*
  - $[-162; -2]$  MHz
  - $f_{IF,L} = -82$  MHz



# Basic Modeling of the I/Q received signal



- $N_{sat}$  : number of satellites in visibility (*elevation larger than  $\approx 10$  degrees*)
- $i$  : satellite identifier
- $C_i$  : received signal power
- $\tau_i$  : time of flight in the user time scale
- $\Delta f_i$  : Doppler shift
- $\theta_i$  : carrier phase

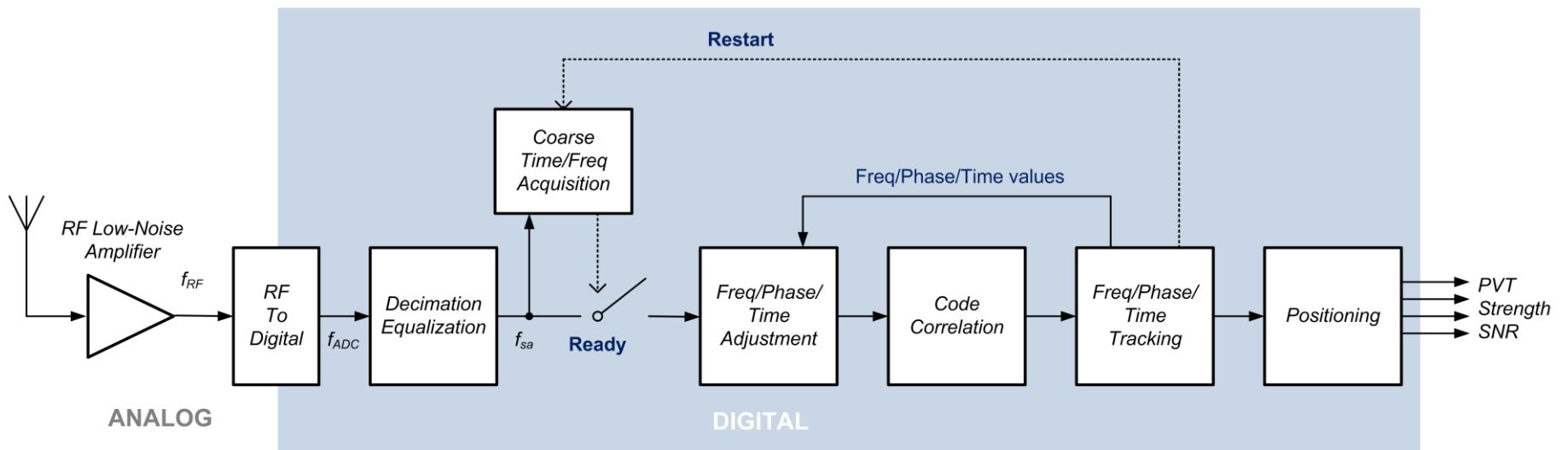
$$r(t) = \sum_{i=1}^{N_{sat}} \sqrt{2C_i} s_i(t - \tau_i) \exp\left[j(2\pi\Delta f_i t + \theta_i)\right] + w(t)$$

Whatever the architecture, at the output of the **RF to Digital Section**:

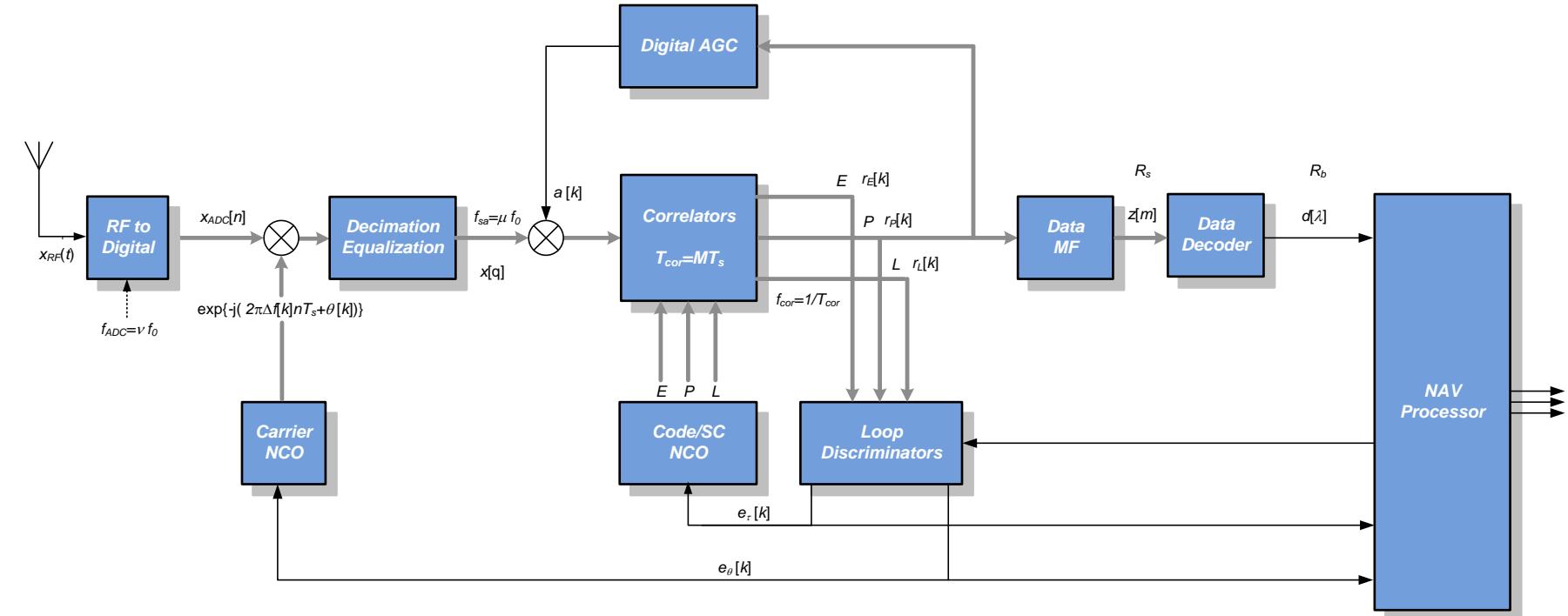
$$x_{ADC}[n] = \sum_{i=1}^{N_{sat}} \sqrt{2C_i} s_i(nT_s - \tau_i) \exp\left[j(2\pi\Delta f_i nT_s + \theta_i)\right] + w[nT_s]$$

# Basic Receiver Architecture

- Modern receivers perform all of the signal processing functions with DSP components and algorithms



# Detailed Receiver Architecture

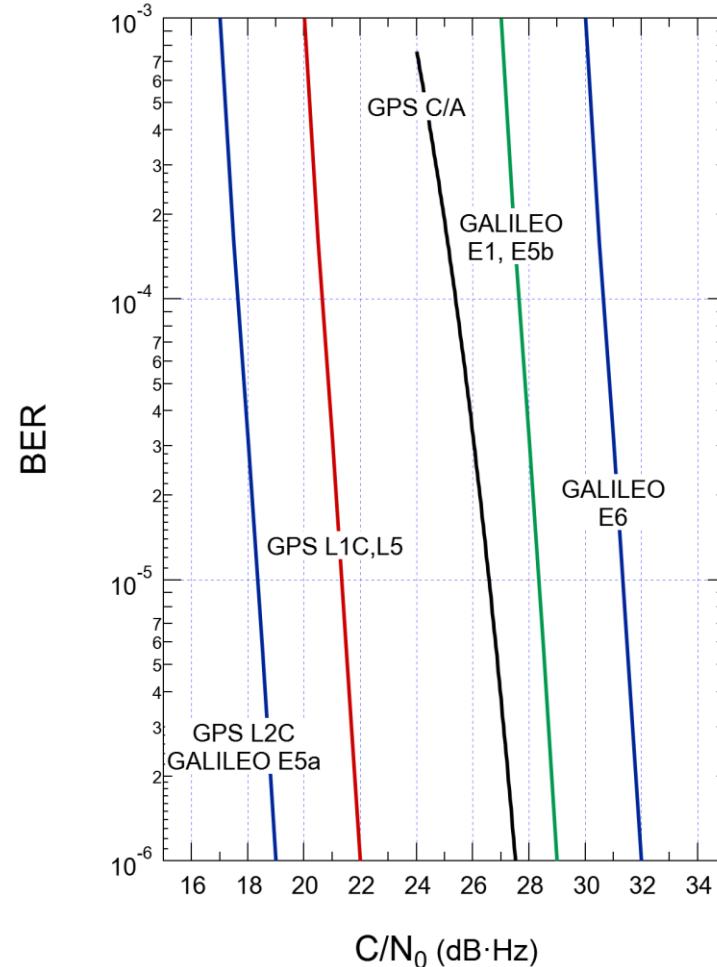


$E$ =Early,  $L$ =Late, correlations for the DLL

$P$ =Prompt for Data Detection (may also be used for the DLL)

# NAV Message Data Reception

- Data detection performance depend on the bit-rate and on the kind of channel coding that is used on the different carriers/channels/systems



# A Case Study: Code Delay Tracking



- $N_{sat}=1$
- $i=1$  satellite identifier
- $C_1=1/2$  received signal power
- $\tau_1=\tau$  time of flight in the user time scale
- $\Delta f_i=0$  Doppler shift
- $\theta_i=0$  carrier phase
- Pilot signal only (no Q component needed)

$$r(t) = \sqrt{2C} c(t - \tau) + w(t)$$

Let us concentrate on delay tracking...



# Ranging Code Tracking

**To maximize**  $\int_0^{T_0} r(t)c(t - \tilde{\tau})dt$  **we do:**

$$\frac{d}{d\tilde{\tau}} \int_0^{T_0} r(t)c(t - \tilde{\tau})dt = 0 \Rightarrow \int_0^{T_0} r(t) \frac{d}{d\tilde{\tau}} c(t - \tilde{\tau})dt = 0$$

$$\frac{d}{d\tilde{\tau}} c(t - \tilde{\tau}) \cong -\frac{c(t - \tilde{\tau} + \Delta) - c(t - \tilde{\tau} - \Delta)}{2\Delta} \Rightarrow \int_0^{T_0} r(t) \frac{c(t - \tilde{\tau} + \Delta) - c(t - \tilde{\tau} - \Delta)}{2\Delta} dt = 0$$

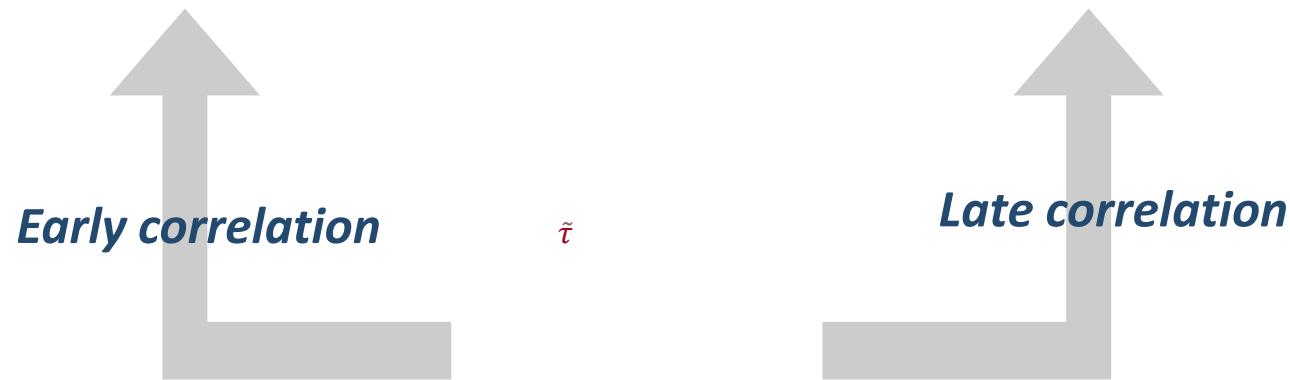
$$\int_0^{T_0} r(t)c(t - \tilde{\tau} + \Delta)dt - \int_0^{T_0} r(t)c(t - \tilde{\tau} - \Delta)dt = 0$$

$T_0$  is the correlation (observation) time

# Error Function



$$e(\tilde{\tau}) = \int_0^{T_0} r(t)c(t - \tilde{\tau} + \Delta)dt - \int_0^{T_0} r(t)c(t - \tilde{\tau} - \Delta)dt$$



*Can be computed Real-Time on the k-th **correlation interval**  $[(k-1)T_0, kT_0]$*

$$e(\tilde{\tau}[k-1]) = \int_{(k-1)T_0}^{kT_0} r(t)c(t - \tilde{\tau}[k-1] + \Delta)dt - \int_{(k-1)T_0}^{kT_0} r(t)c(t - \tilde{\tau}[k-1] - \Delta)dt$$

*the variable  $k$  runs at the **correlation time**  $T_0$*

# Recursive delay estimation



- We know that the current estimate is correct (optimum) *iff* the feedback signal  $e(\hat{\tau}[k-1])$  is zero, so that the estimate is maximum-likelihood.
- BUT, the feedback signal is in general NOT 0 - we can use this feedback signal (aka *discriminator output*) to recursively update the current estimate and find a better one:

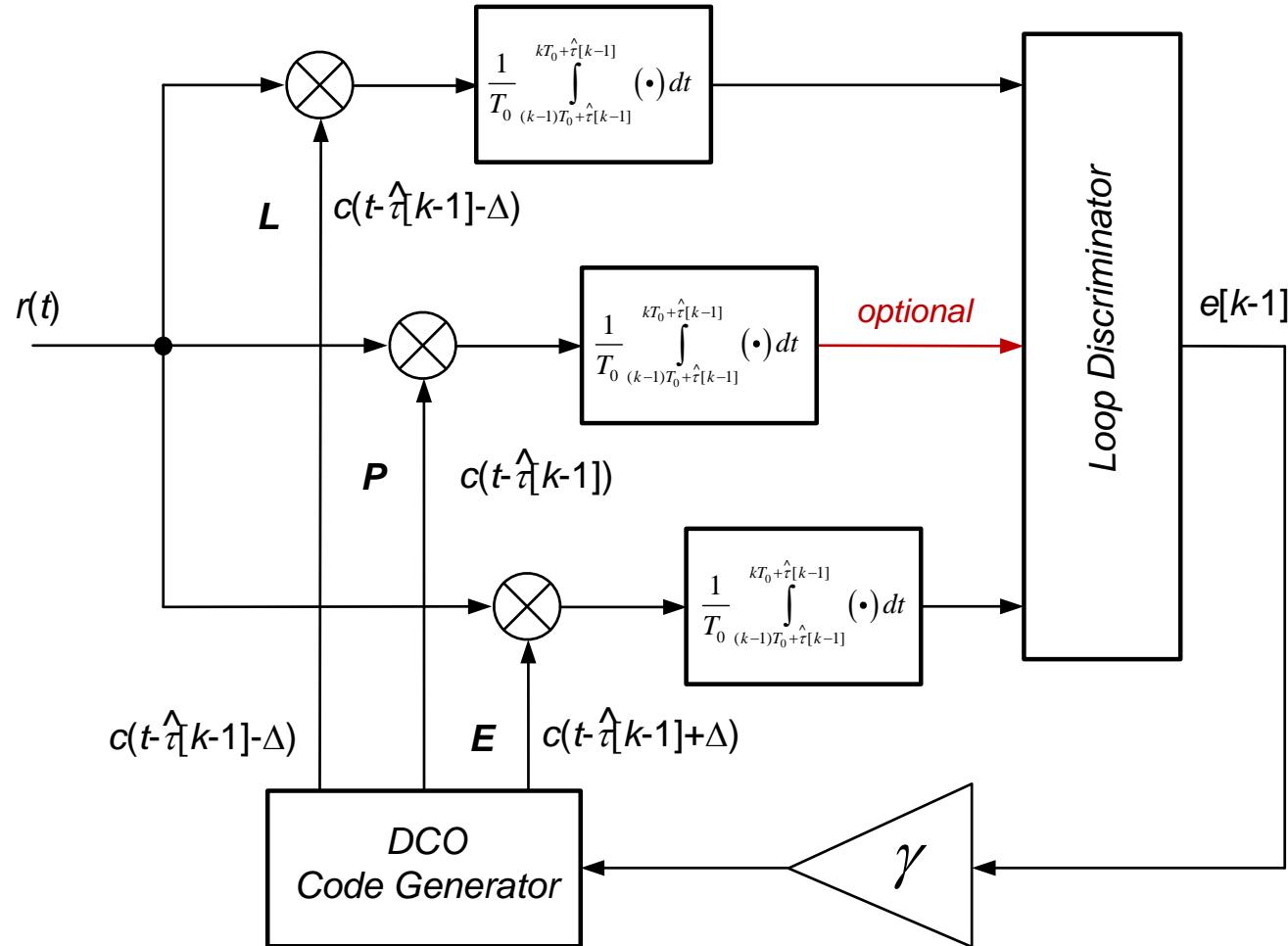
$$\hat{\tau}[k] = \hat{\tau}[k-1] - \gamma e[k-1]$$

where, for short,

$$e[k-1] \triangleq e(\hat{\tau}[k-1])$$

- This is the equation of a *negative feedback control loop* that finds an equilibrium at steady state whenever  $e[k-1]=0 \rightarrow$  finds the (optimum) ML delay estimate !
- The constant  $\gamma$  is a *stepsize* (the *gain* of the feedback loop) that dictates the magnitude of the feedback correction term

# The Delay-Lock “Loop” (DLL)



# Early/Late Correlation Values

*Received signal*

$$r(t) = \sqrt{2C} \cdot c(t - \tau) + w(t)$$

$$e^\pm(\hat{\tau}[k-1]) = \int_{(k-1)T_0}^{kT_0} r(t)c(t - \hat{\tau}[k-1] \pm \Delta)dt \quad \text{Early/Late Correlations}$$

$$\begin{aligned} &= \sqrt{2C} \int_{(k-1)T_0}^{kT_0} c(t - \tau)c(t - \hat{\tau}[k-1] \pm \Delta)dt + \int_{(k-1)T_0}^{kT_0} w(t)c(t - \hat{\tau}[k-1] \pm \Delta)dt \\ &= \sqrt{2C} R_c(\hat{\tau}[k-1] - \tau \mp \Delta) + W^\pm[k-1] \end{aligned}$$

Code Autocorrelation Function

Gaussian RVs



- Let us disregard noise for the moment, and let us define the estimation *error*

$$\varepsilon[k] \triangleq \hat{\tau}[k] - \tau$$

- The feedback signal (discriminator output) is therefore

$$\begin{aligned} e[k-1] &= e^+(\hat{\tau}[k-1]) - e^-(\hat{\tau}[k-1]) \\ &= \sqrt{2C} \cdot [R_c(\varepsilon[k-1] - \Delta) - R_c(\varepsilon[k-1] + \Delta)] = e(\varepsilon[k-1]) \end{aligned}$$

that depends on the estimation error  $\varepsilon[k-1]$  – it is an *estimation error detector*

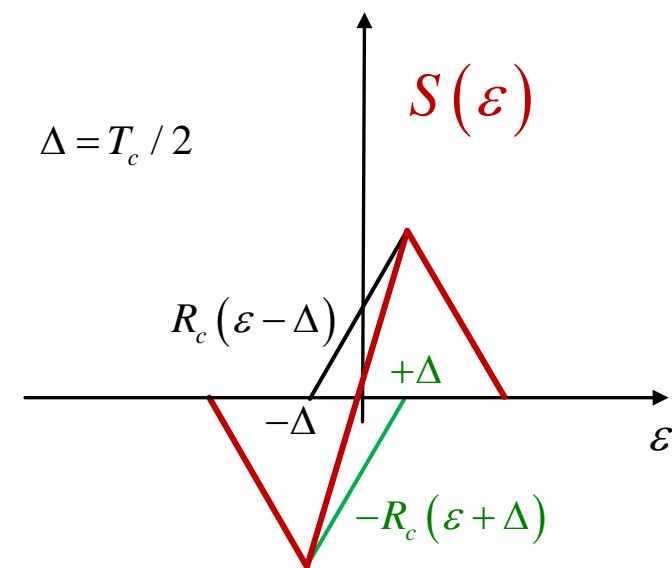
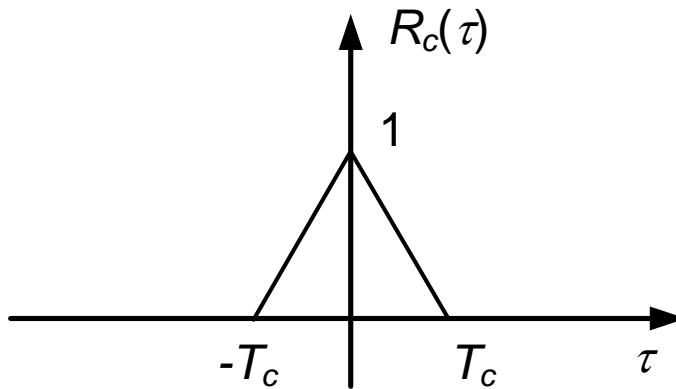
- The value of  $e$  as a function of a *fixed* estimation error  $\varepsilon$  (and without noise) is called the *discriminator characteristics* or *S-curve* of the estimator (loop)

$$S(\varepsilon) = e[k-1] \Big|_{\varepsilon[k-1]=\varepsilon} = \sqrt{2C} \cdot [R_c(\varepsilon - \Delta) - R_c(\varepsilon + \Delta)]$$

# The S-curve

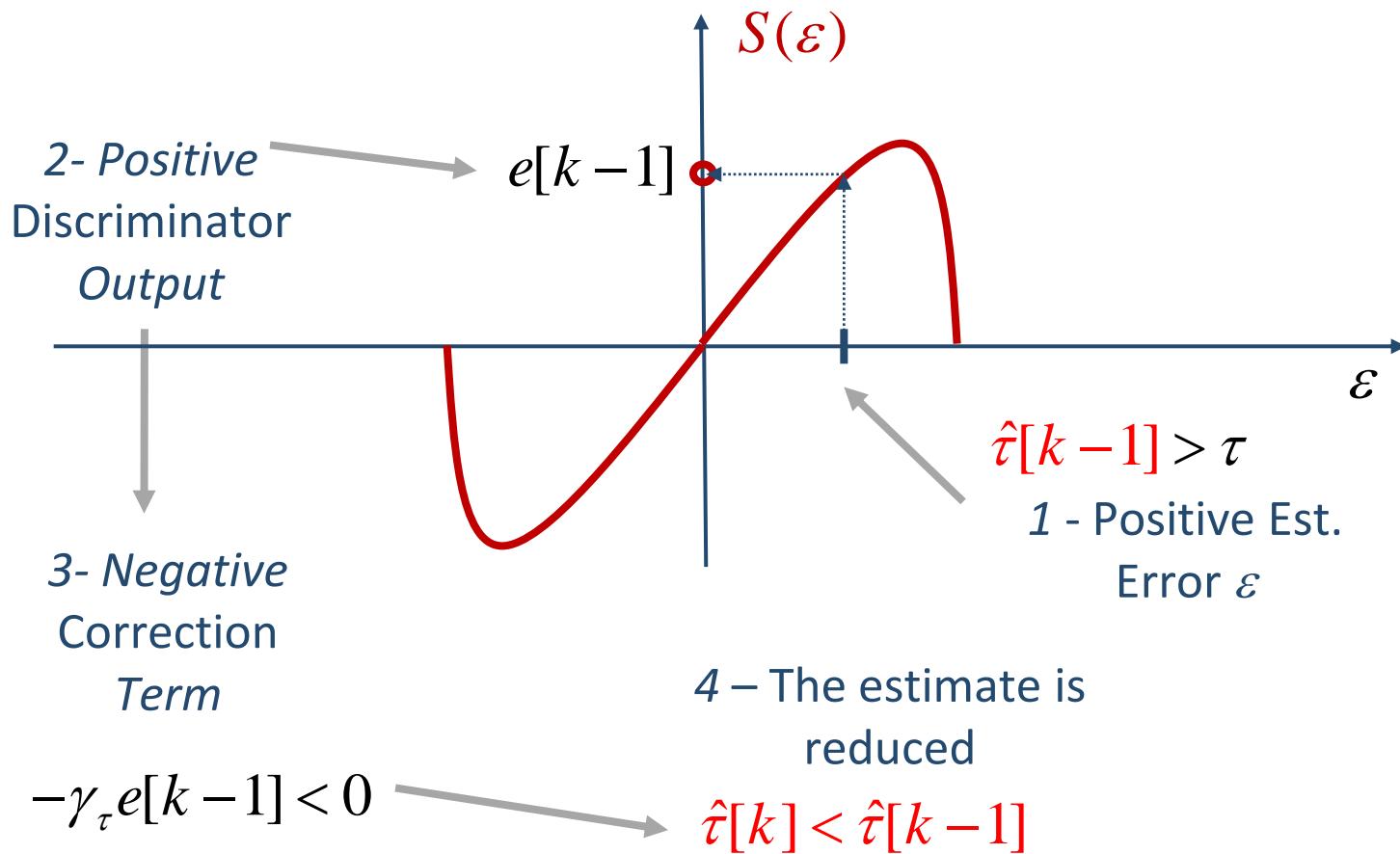


$$S(\varepsilon) = e[k-1] \Big|_{\varepsilon[k-1]=\varepsilon} = \sqrt{2C} \cdot \left[ R_c(\varepsilon - \Delta) - R_c(\varepsilon + \Delta) \right]$$



# Negative Feedback Loop

$$\hat{\tau}[k] = \hat{\tau}[k-1] - \gamma_\tau e[k-1]$$





- In addition to the useful term, the discriminator output also contains *noise*

$$\begin{aligned}
 e[k-1] &= e^+(\hat{\tau}[k-1]) - e^-(\hat{\tau}[k-1]) \\
 &= \sqrt{2C} \cdot [R_c(\varepsilon[k-1] - \Delta) - R_c(\varepsilon[k-1] + \Delta)] \\
 &\quad + [W^+[k-1] - W^-[k-1]] = S(\varepsilon[k-1]) + W[k-1]
 \end{aligned}$$

so the successive estimates  $\hat{\tau}[k]$  will be *noisy* as well

$$\hat{\tau}[k] = \hat{\tau}[k-1] - \gamma(S(\varepsilon[k-1]) + W[k-1])$$

- Large values of  $\gamma$  give a “fast” loop that is ready to react to large errors and drive back the estimate towards the correct value, but also enhance the effect of noise and provide a larger *error variance*  $\sigma_\varepsilon^2$



# Iterative Estimation ERROR Analysis

$$\hat{\tau}[k] = \hat{\tau}[k-1] - \gamma e[k-1]$$

Subtracting the true value  $\tau$  we get the (recursive) ESTIMATION ERROR EQUATION

$$\varepsilon[k] = \varepsilon[k-1] - \gamma e[k-1]$$

still running at estimation time  $T_0$

LOOP Noise

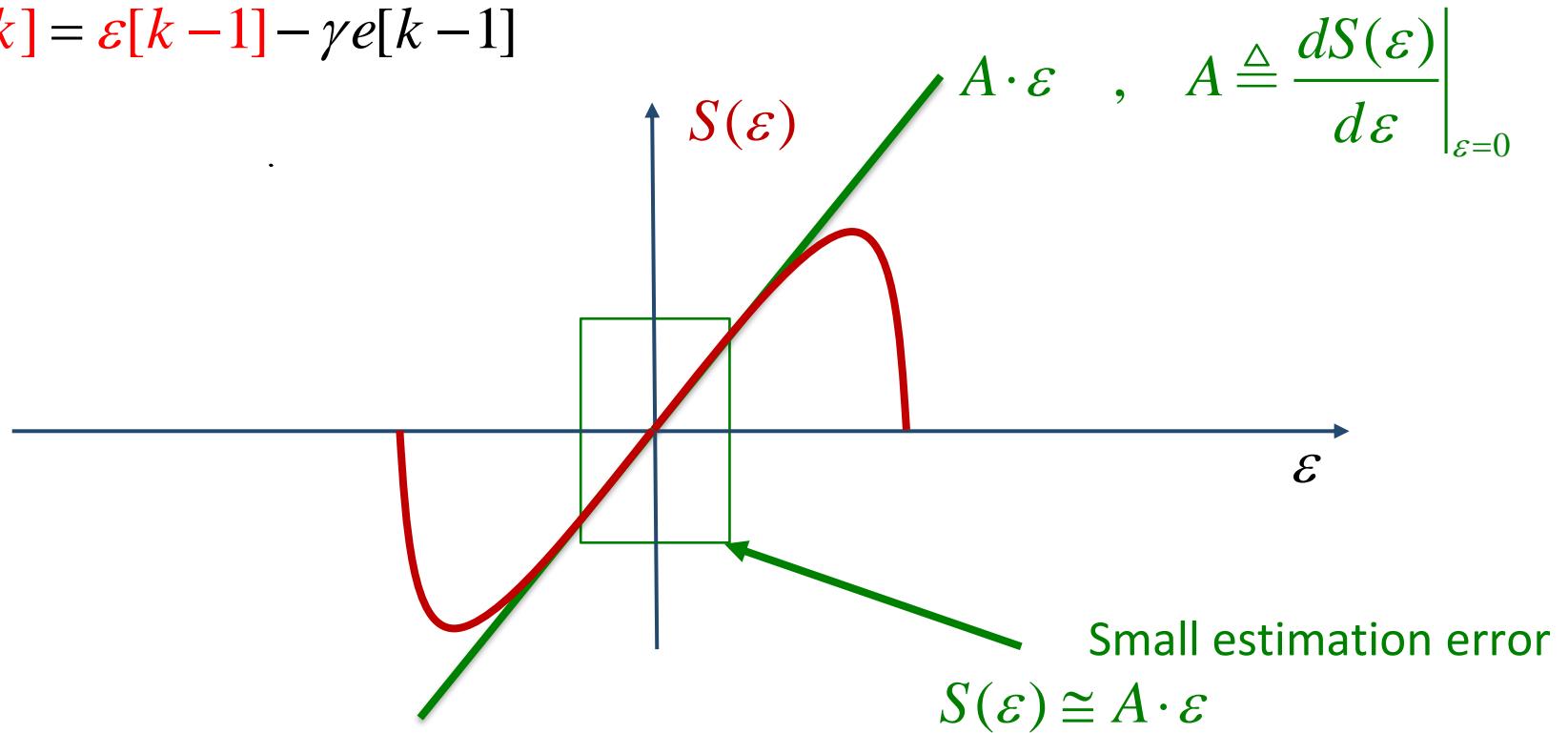
$$\varepsilon[k] = \varepsilon[k-1] - \gamma S(\varepsilon[k-1]) - \gamma W[k-1]$$

LOOP Discriminator  
Function



# Estimate Variance: Linearized Loop

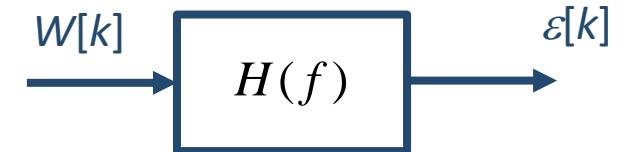
$$\varepsilon[k] = \varepsilon[k-1] - \gamma e[k-1]$$



$$\varepsilon[k] \approx \varepsilon[k-1] - \gamma (A\varepsilon[k-1] + W[k-1]) , \quad W[k-1] = W^+[k-1] - W^-[k-1]$$

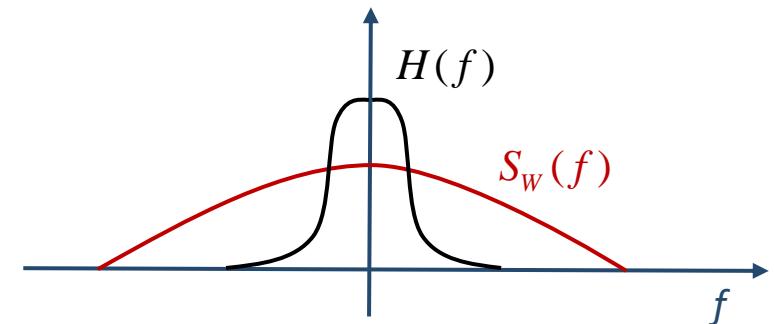
# Linearized First Order Loop

$$\varepsilon[k] = (1 - \gamma A) \varepsilon[k-1] - \gamma A \frac{W[k-1]}{A}$$



It is a (narrowband) FIRST-ORDER IIR Filter operating at a rate  $1/T_0$

$$\sigma_\tau^2 = \sigma_\varepsilon^2 \approx 2B_L T_0 \left. \frac{S_w(f)}{A^2} \right|_{f=0}$$



with an equivalent (normalized) *noise bandwidth*

$$B_L T_0 = \frac{\gamma A}{2(2 - \gamma A)} \approx \frac{\gamma A}{4}$$



For GPS C/A,  $\Delta = T_c / 2$

$$A = \frac{2}{T_c} \sqrt{2C} \quad , \quad S_w(f) = \sigma_w^2$$

$$\sigma_w^2 = E\{W^2\} = E\{(W^+ - W^-)^2\} = E\{(W^+)^2\} + E\{(W^-)^2\} - 2E\{W^+W^-\}$$

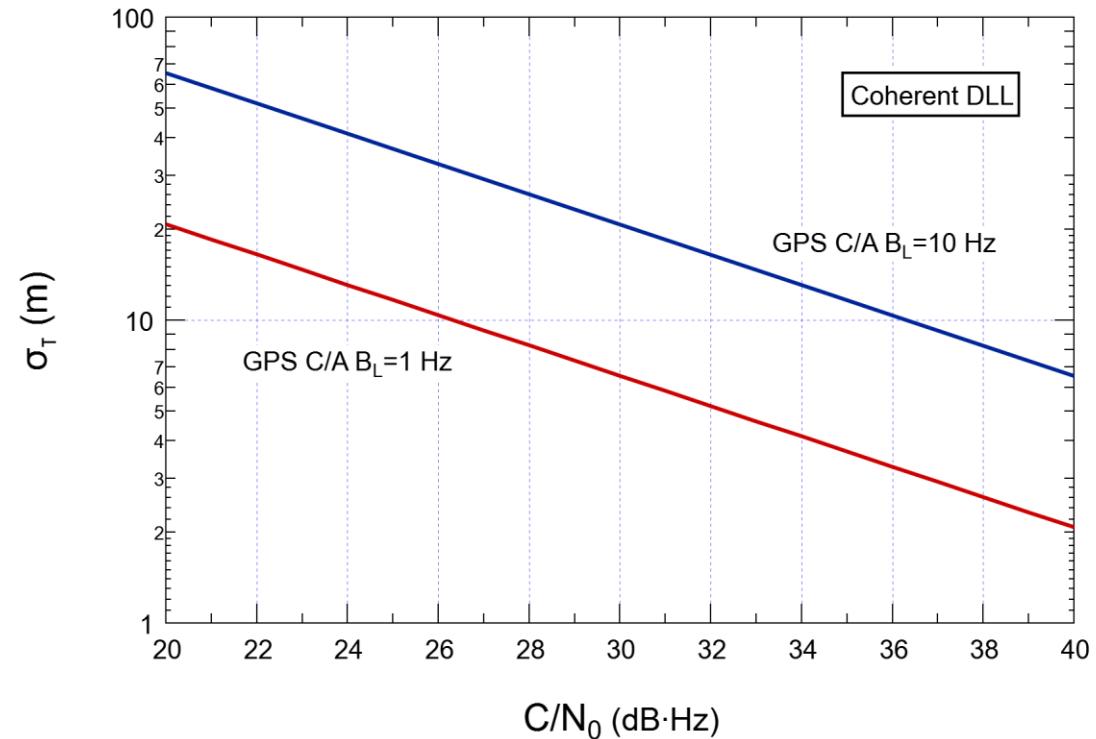
$$= \frac{N_0}{T_0} + \frac{N_0}{T_0} - 0$$

$$\sigma_\varepsilon^2 = 2B_L T_0 \frac{2N_0/T_0}{2C} \frac{T_c^2}{4} = T_c^2 \frac{B_N}{2C/N_0}$$

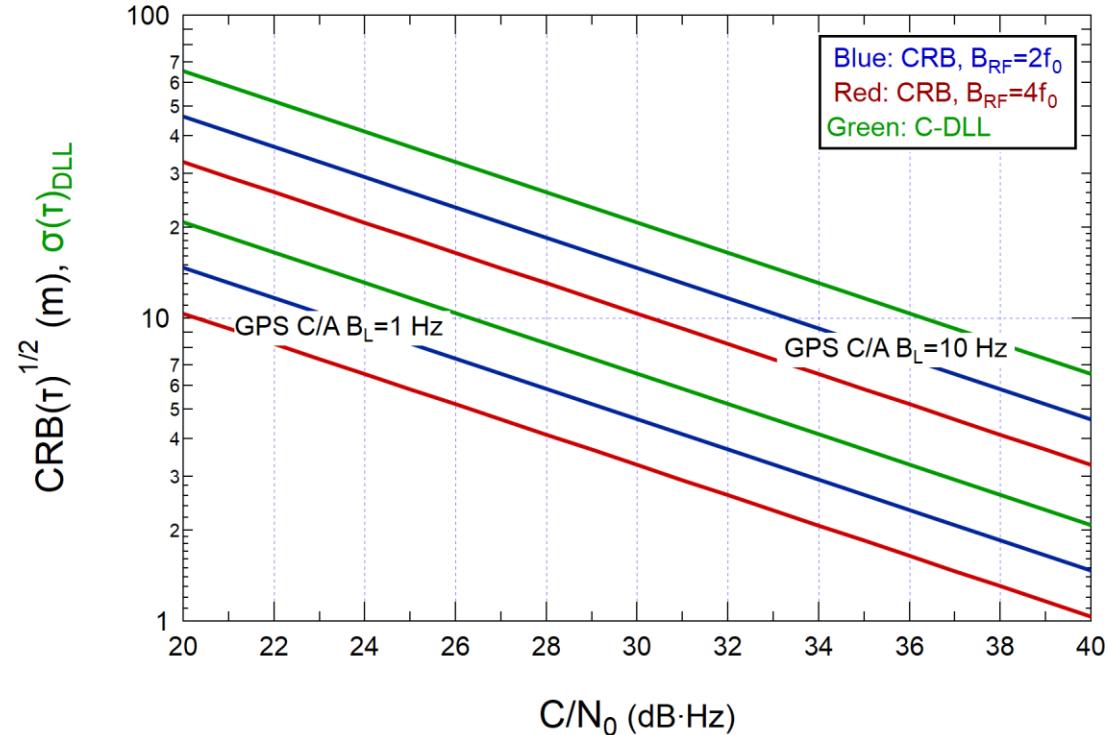
$$\sigma_\tau[m] = c T_c \sqrt{\frac{B_L}{2C/N_0}}$$



# Coherent DLL Accuracy



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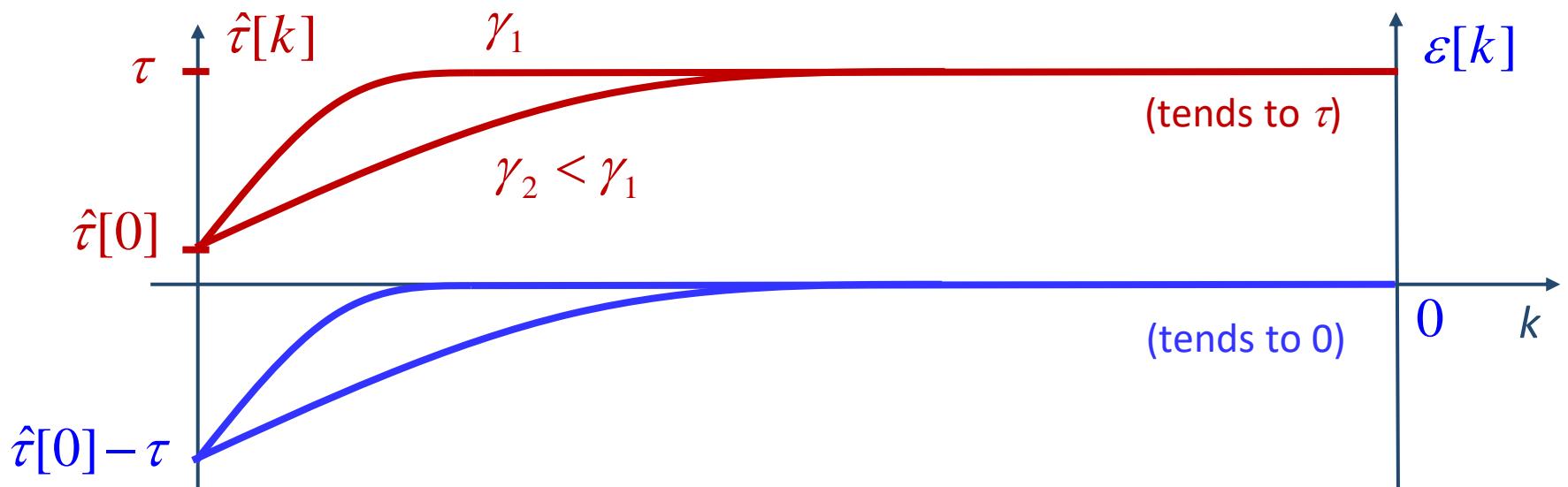
# Visualizing the Loop Bandwidth

- Linearized loop error equation with **no noise**

$$\varepsilon[k] = (1 - \gamma A) \varepsilon[k-1] , \quad \gamma A \cong 4B_L T_0 \ll 1$$

The loop evolution starts from  $\varepsilon[0] = \varepsilon_0 < T_c / 2$  as resulting from the acquisition procedure. Then from

$$\varepsilon[1] = (1 - \gamma A) \varepsilon_0 , \quad \varepsilon[2] = (1 - \gamma A)^2 \varepsilon_0 , \dots , \quad \varepsilon[k] = (1 - \gamma A)^k \varepsilon_0$$



# Visualizing the Loop Bandwidth

- Linearized loop error equation **with noise**

$$\varepsilon[k] = (1 - \gamma A)\varepsilon[k-1] + \frac{\gamma}{A} W[k-1] \quad , \quad \gamma A \approx 4B_N T_0 \ll 1$$

The loop bandwidth also impacts on the steady-state error variance (aka *jitter*)

