



# Telekomunikaciona merenja TM P06 2018

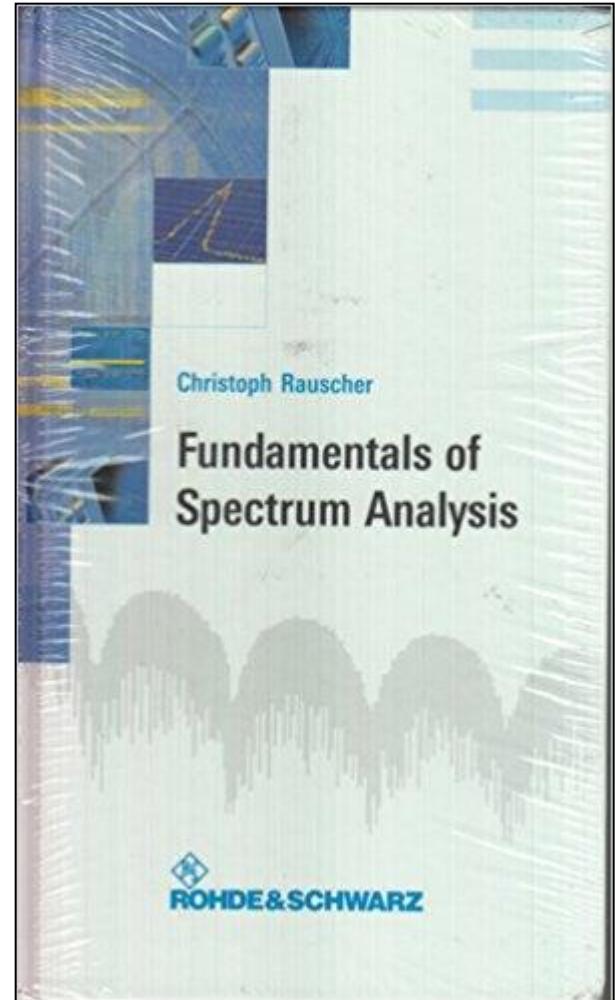
**Profesor dr Miroslav Lutovac**

*"This project has been funded with support from the European Commission. This publication [communication] reflects the views only of the author, and the Commission cannot be held responsible for any use which may be made of the information contained therein"*

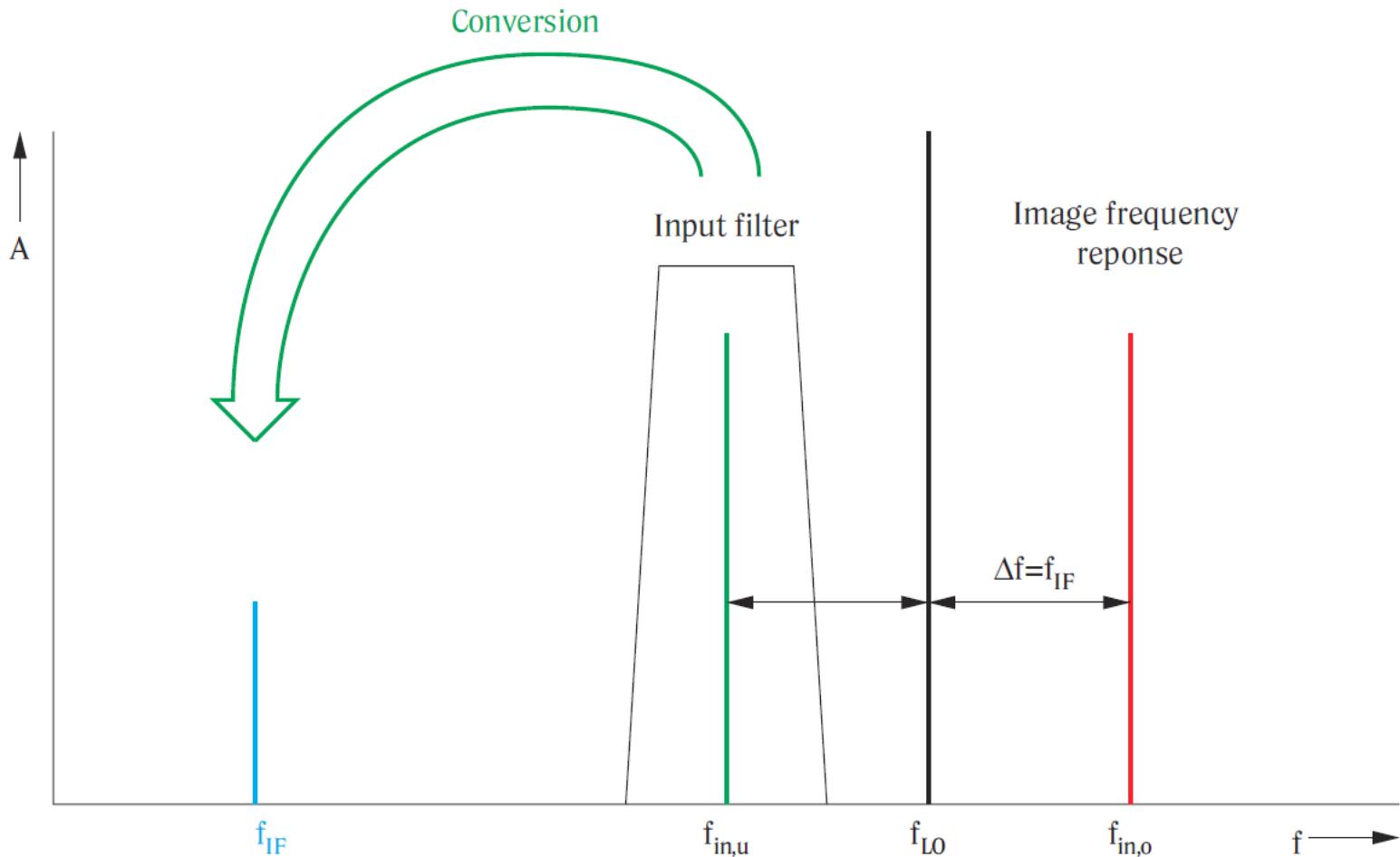
# Osnove analize spektra

Rauscher C.,  
Fundamentals of Spectrum  
Analysis,  
Rohde & Schwarz, 2006

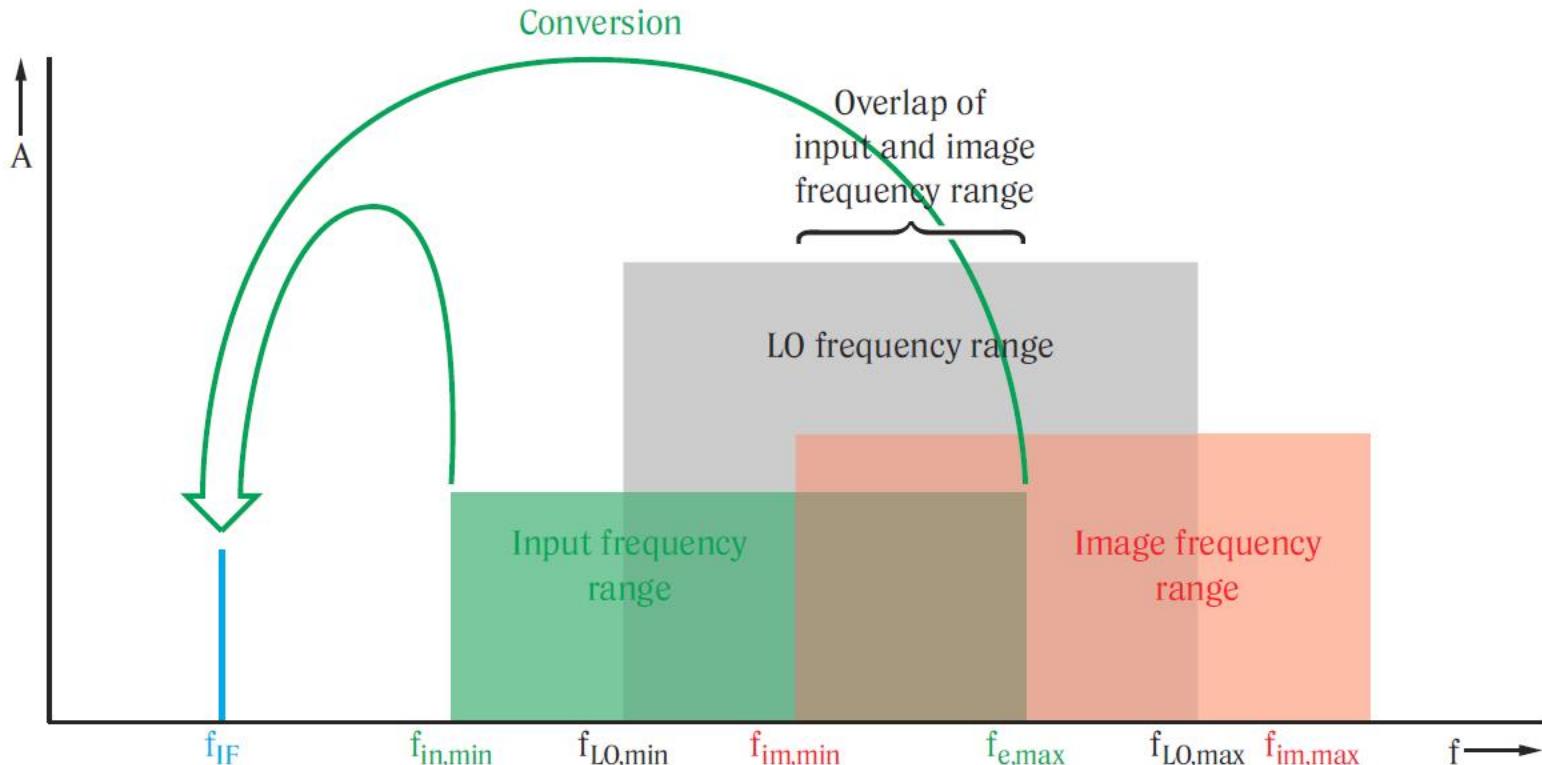
Poglavlje 4



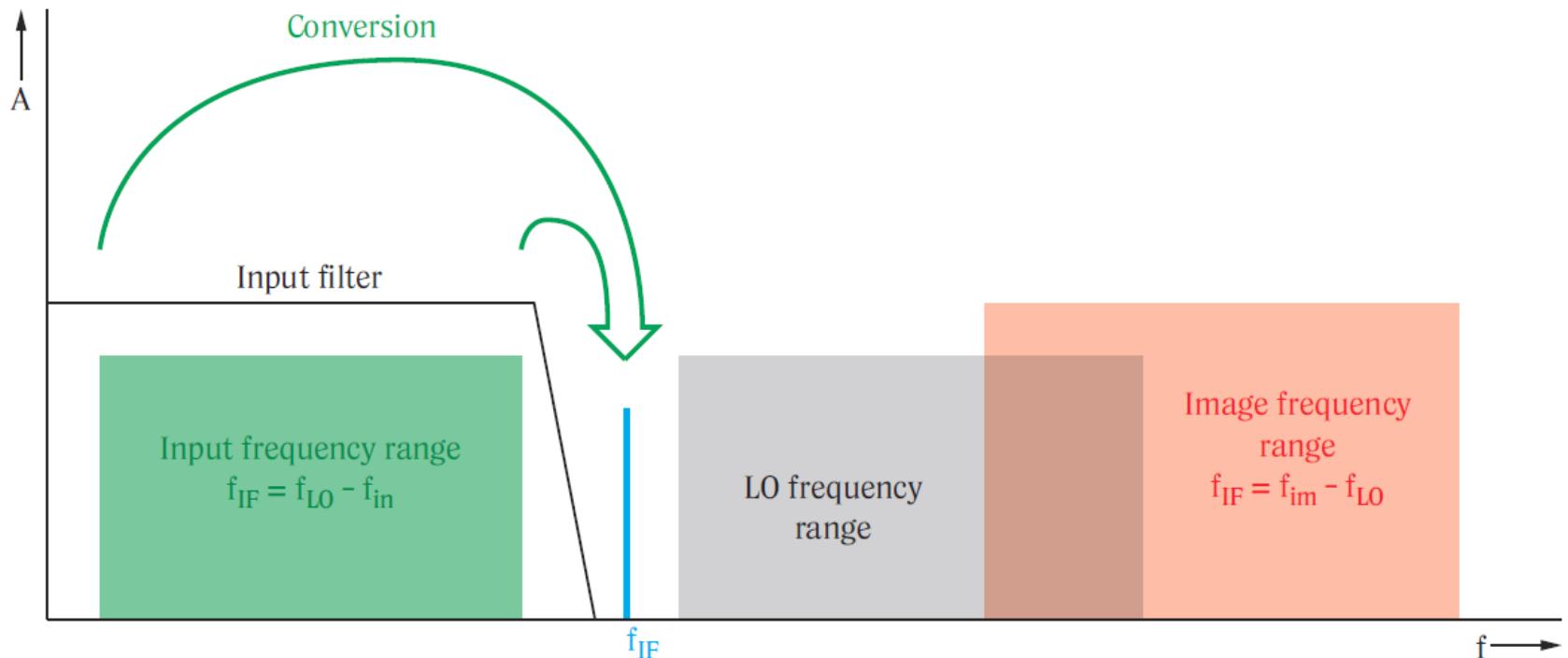
# Dvosmislenost heterodinskog principa



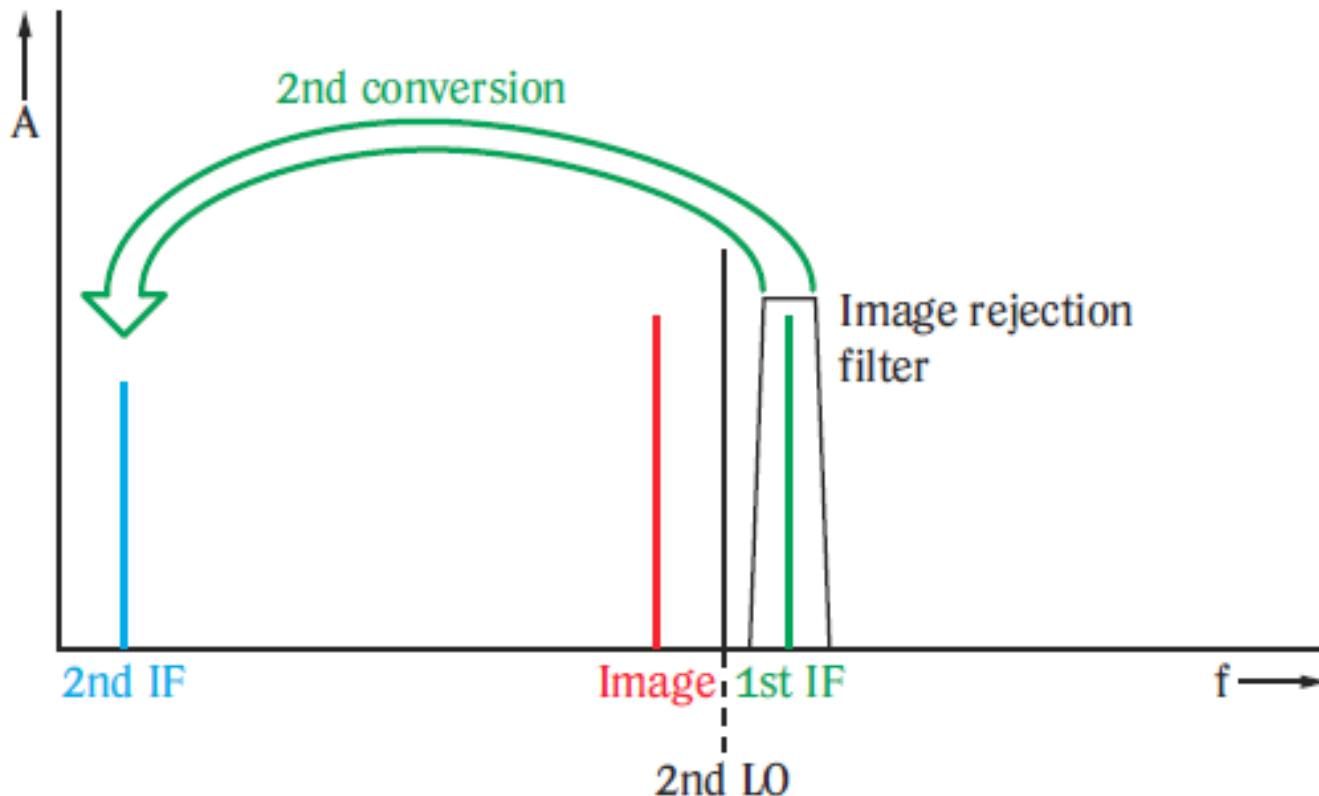
# Preklapanje spektra



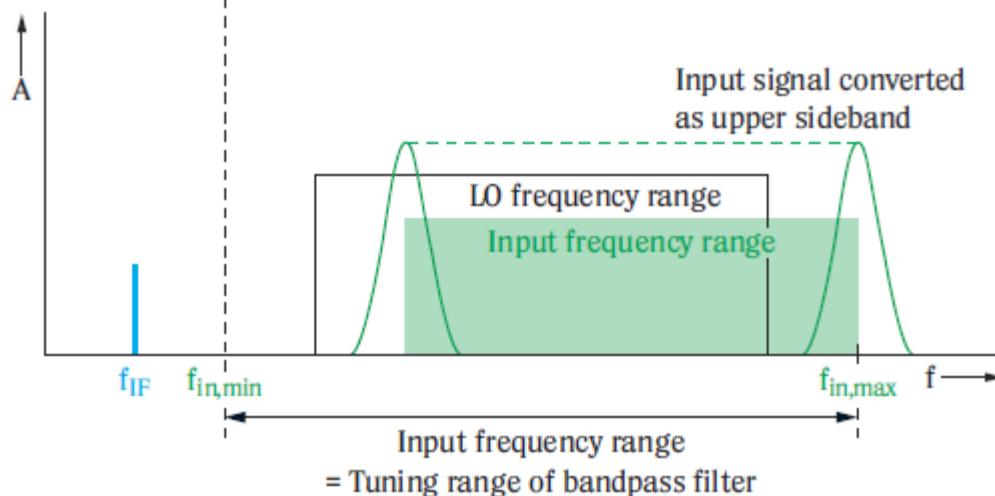
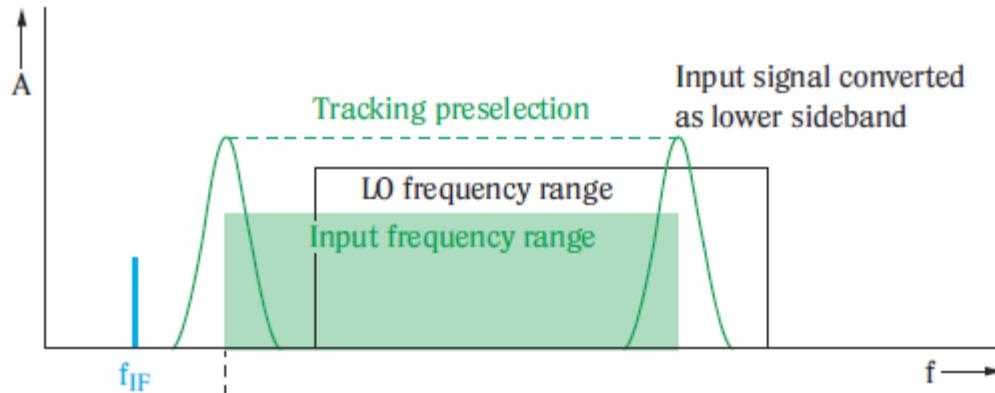
# Princip visoke IF



# Konverzija visoke 1. IF u nisku 2. IF



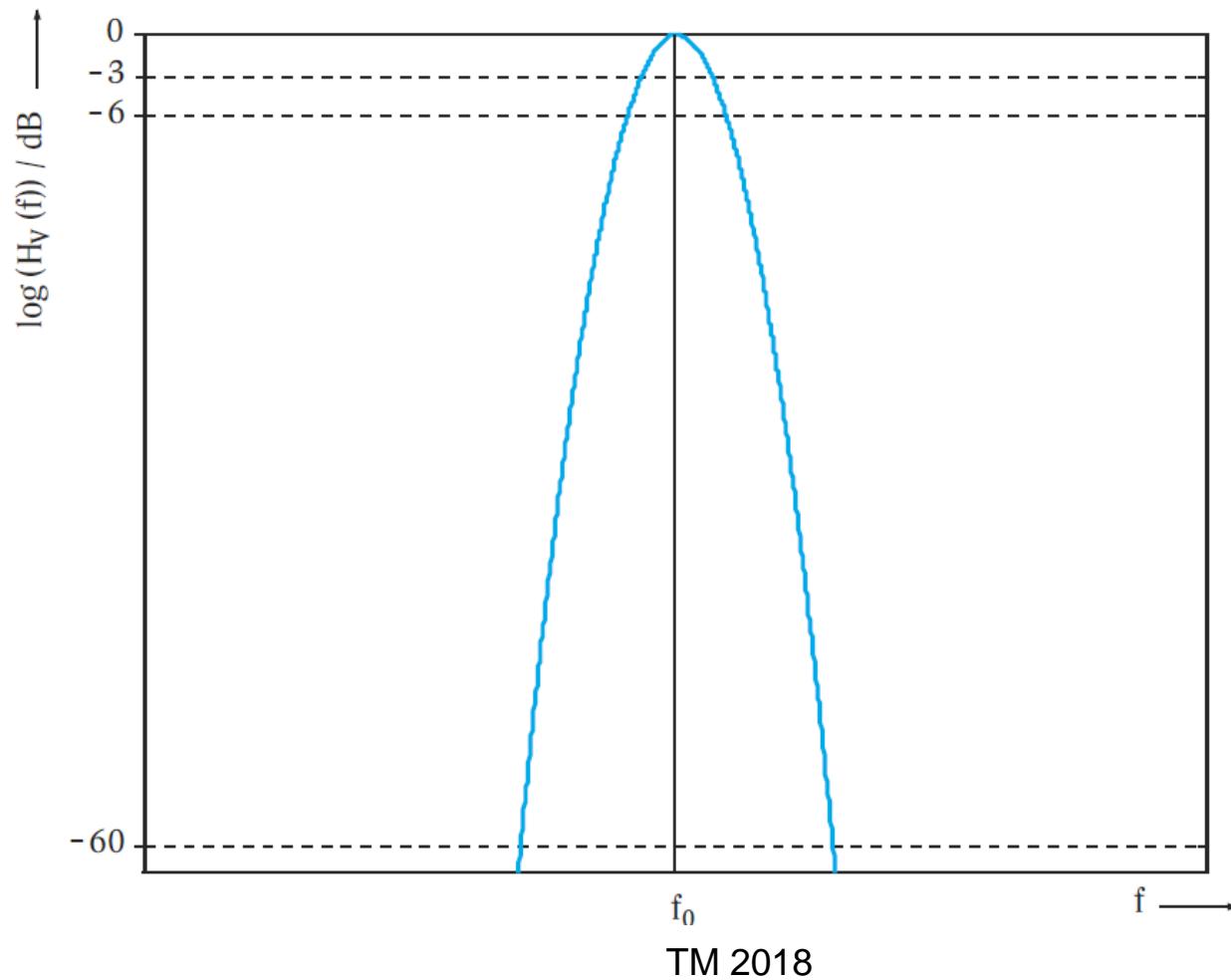
# Konverzija u nisku IF



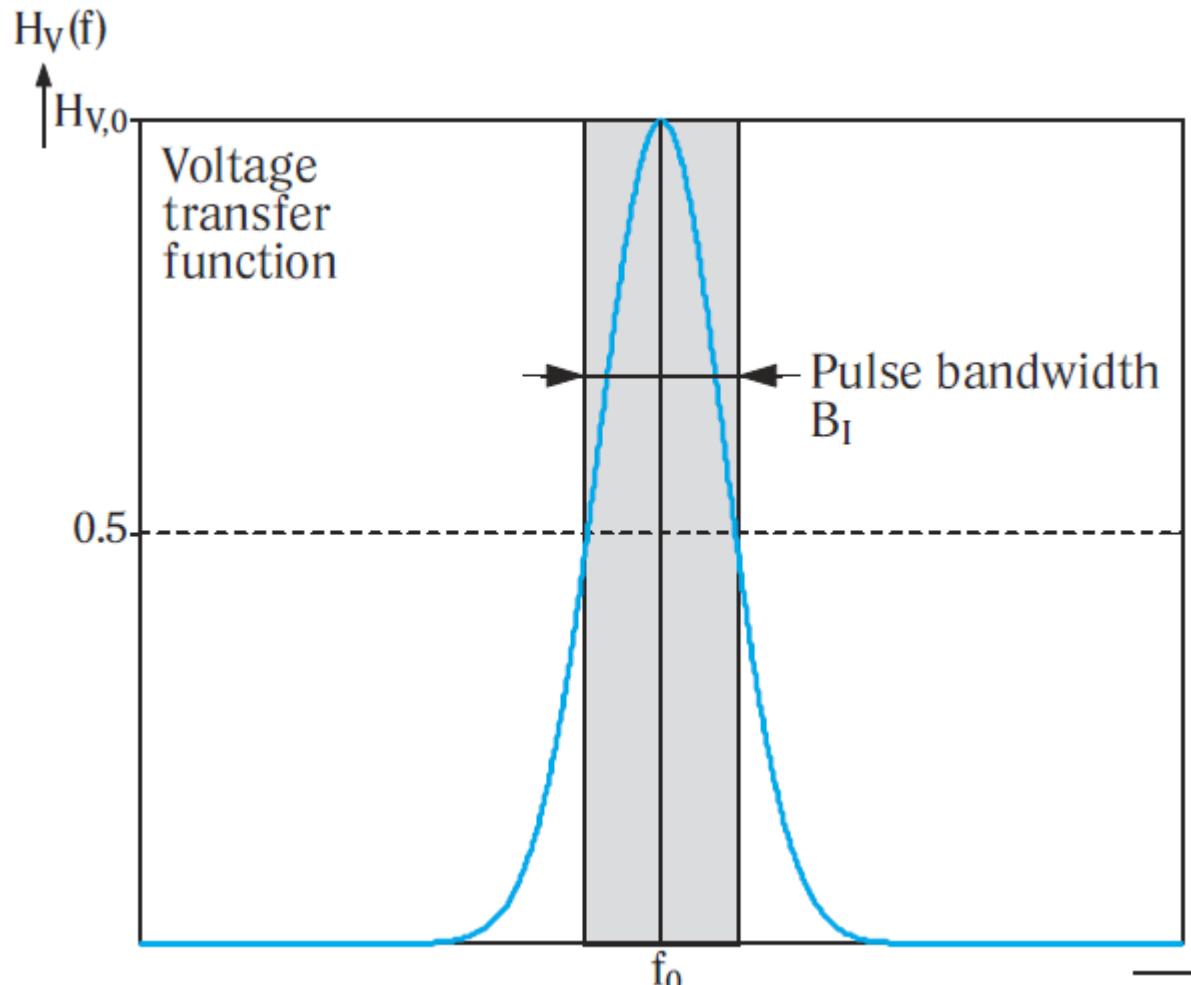
# Obrada signala na IF

- Obrada signala na IF 20.4 MHz
- Signal se pojačava i rezultantni propusni opseg zavisi od IF filtra
- Pojačanje može da se podeđava u koracima 0.1 dB
- Maksimalni nivo signala može se održavati da bude konstantan tokom obrade signala bez od postavljenog slabljenja
- Ako je slabljenje veliko, pojačanje treba da bude veliko tako da se dinamički opseg iskoristi za maksimalnu vrednost
- Koristi se Gausov (vremenski) prozor, zato što pravougani prozor ima veliki prelazni režim

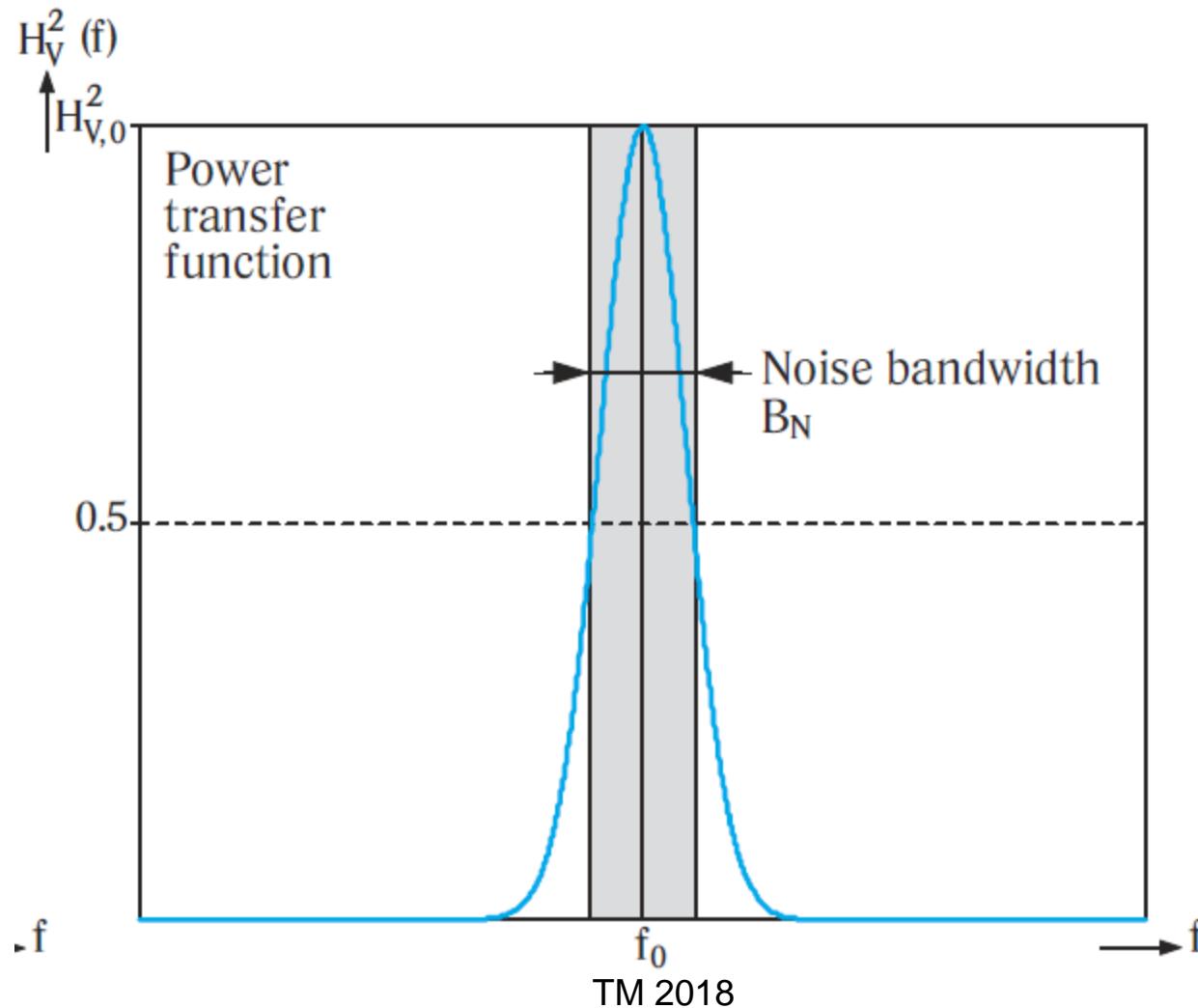
# Funkcija prenosa Gausovog filtra logaritamska razmera



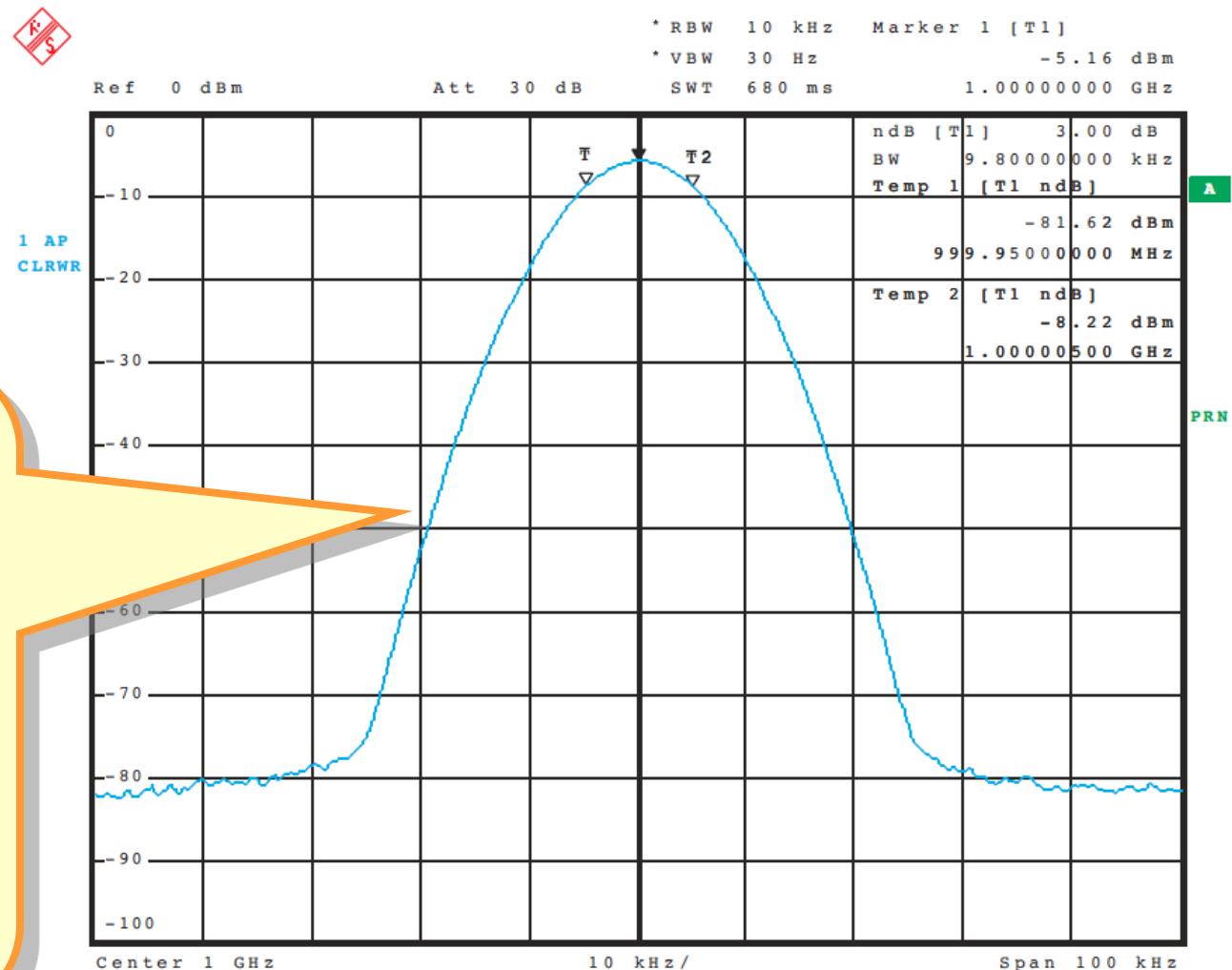
# Naponska funkcija prenosa Gausovog filtra, linearna razmera



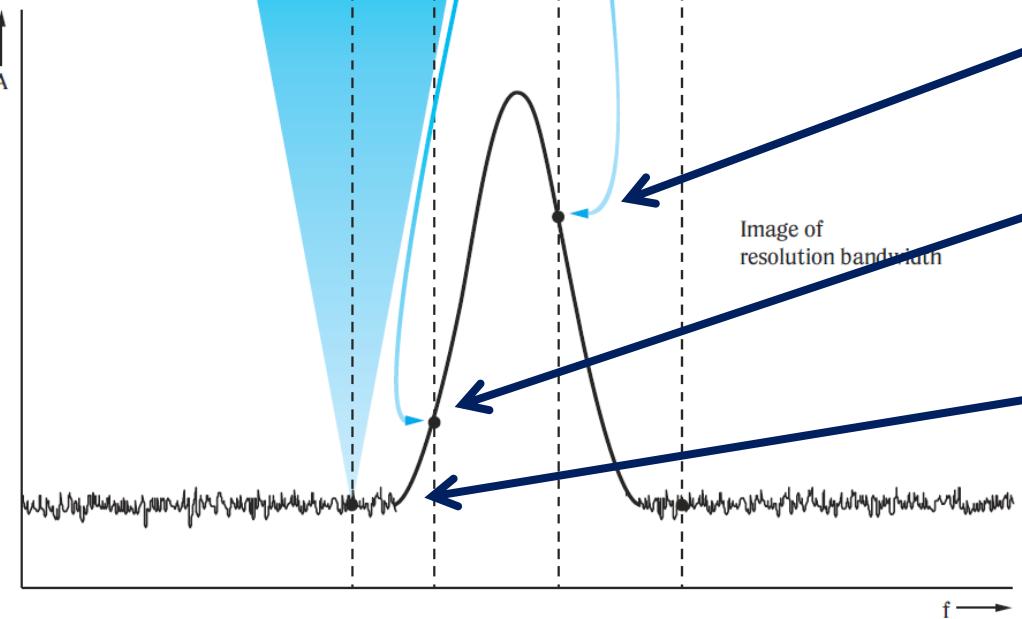
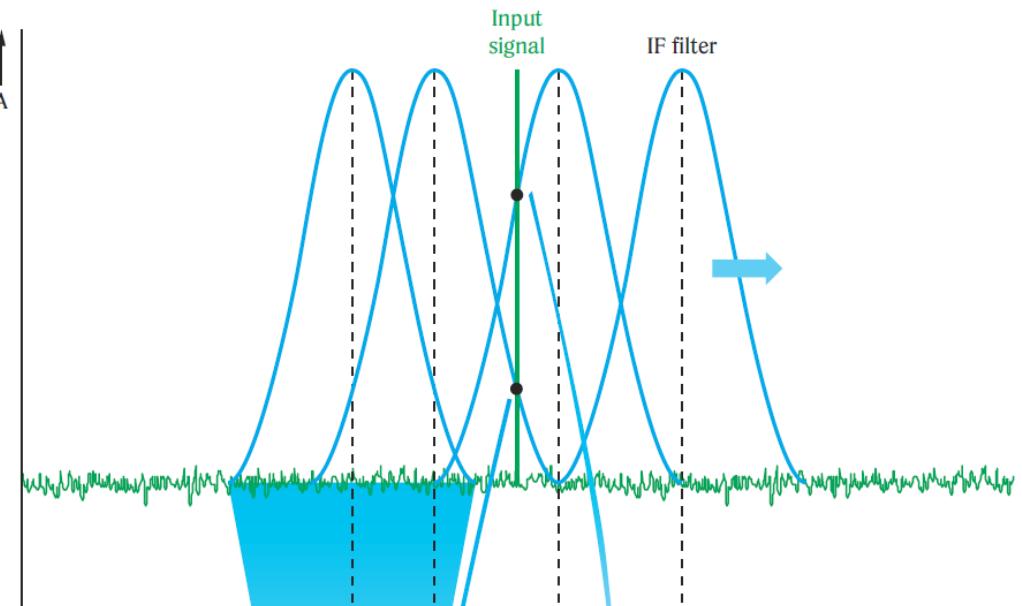
# Funkcija prenosa snage Gausovog filtra, linearna razmara



# Odraz sinusoidalnog signala u IF



Za heterodini princip rada analizatora spektra, kada se posmatra spektar sinusnog signala, ne dobija se vertikalna linija kao za FFT. Umesto toga se dobija



Odraz IF filtra za ulazni sinusoidalan signal zbog “swept past”

- ✓ Na rezonantnoj učestanosti trećeg spektra pojavljuje se veća vrednost
- ✓ Na rezonantnoj učestanosti drugog spektra pojavljuje se manja vrednost
- ✓ Na rezonantnoj učestanosti prvog spektra pojavljuje se 0

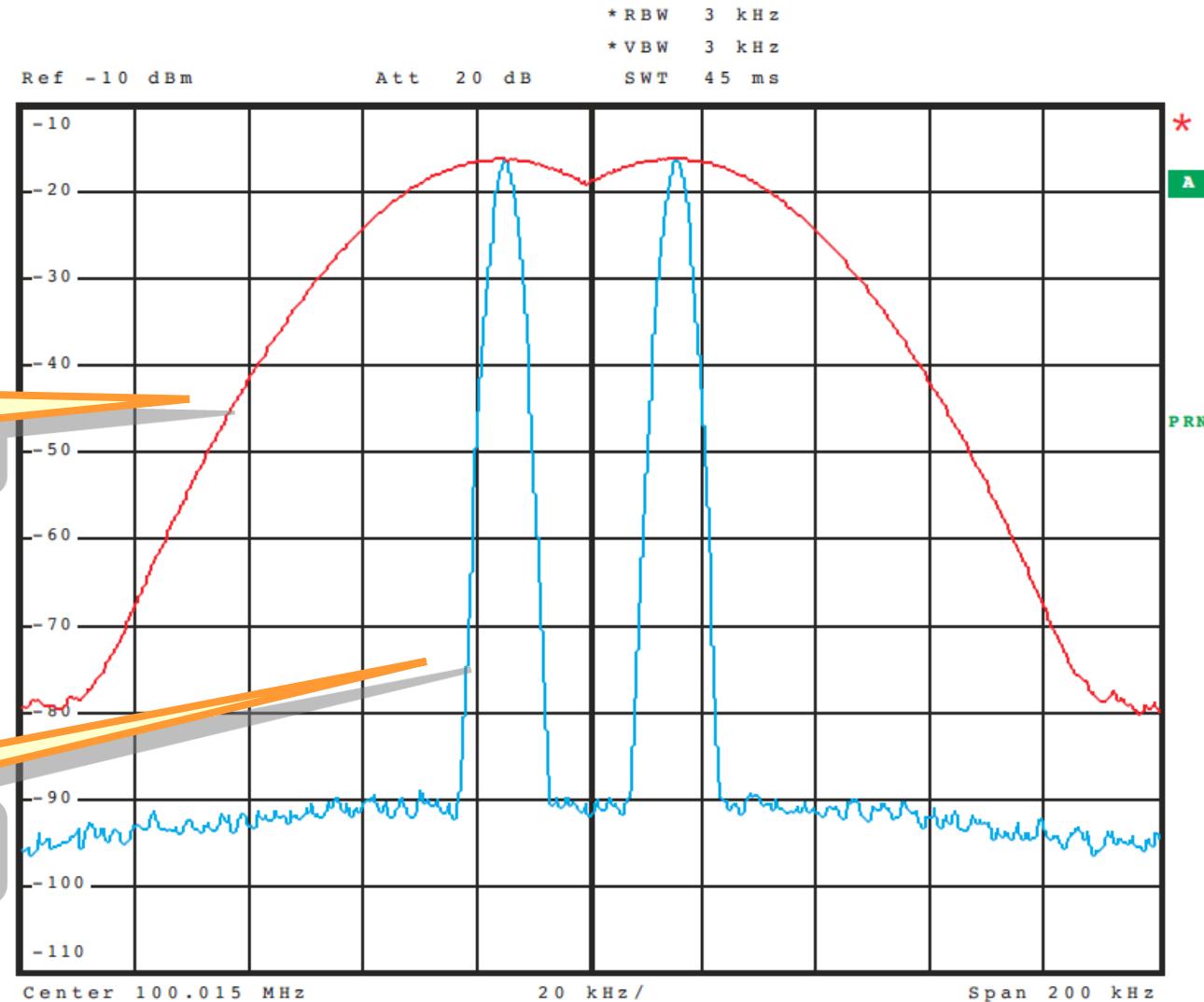
# Uazni signal zbir dve sinusoide

Različite rezolucije propusnih opsega

Crveno RBW=30kHz

Isti nivoi sinusoida

Plavo RBW = 3 kHz



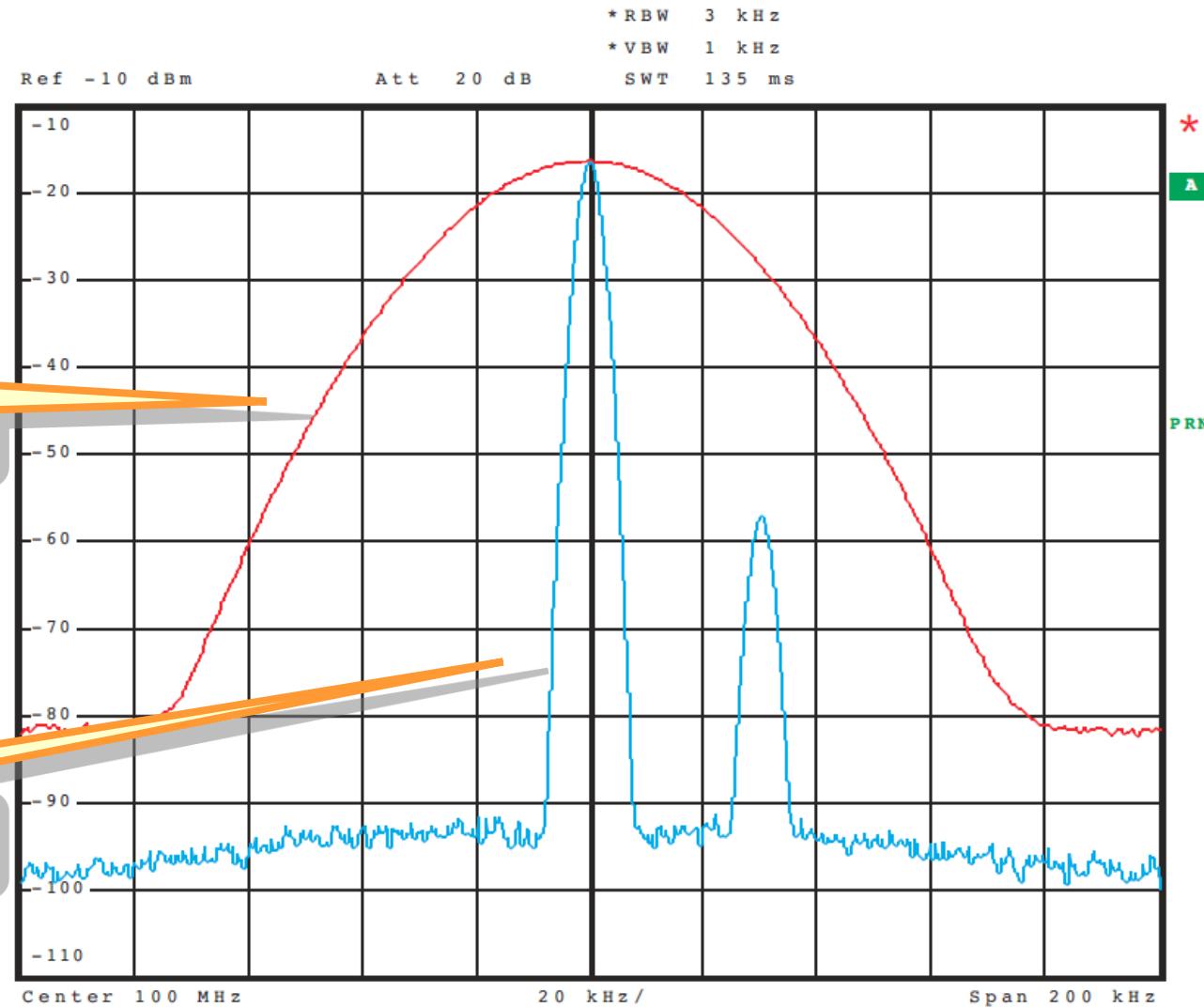
# Uazni signal zbir dve sinusoide

Različite rezolucije propusnih opsega

Crveno RBW=30kHz

Različiti nivoi sinusoida

Plavo RBW = 3 kHz



# Uazni signal zbir dve sinusoide

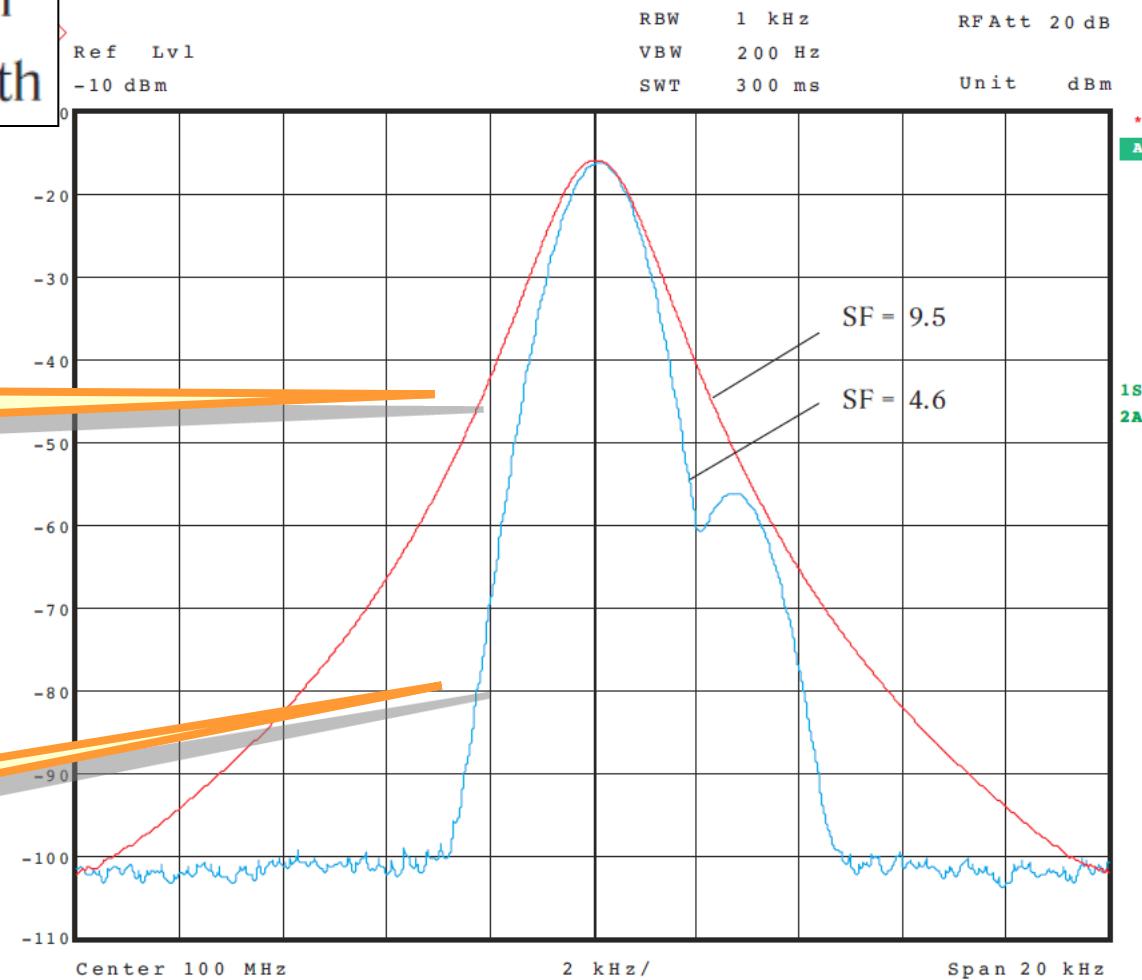
$$\begin{aligned} B_{3\text{dB}} &= 3 \text{ dB bandwidth} \\ B_{60\text{dB}} &= 60 \text{ dB bandwidth} \end{aligned}$$

$$SF_{60/3} = \frac{B_{60\text{dB}}}{B_{3\text{dB}}}$$

**Crveno SF = 9.5**

Različiti nivoi  
sinusoida i  
shape factor

**Plavo SF = 4.6**



**Rezolucija propusnog opsega 1 kHz**

# Vrste filtara

## 1. Analogni filtri (100 kHz - 10 MHz)

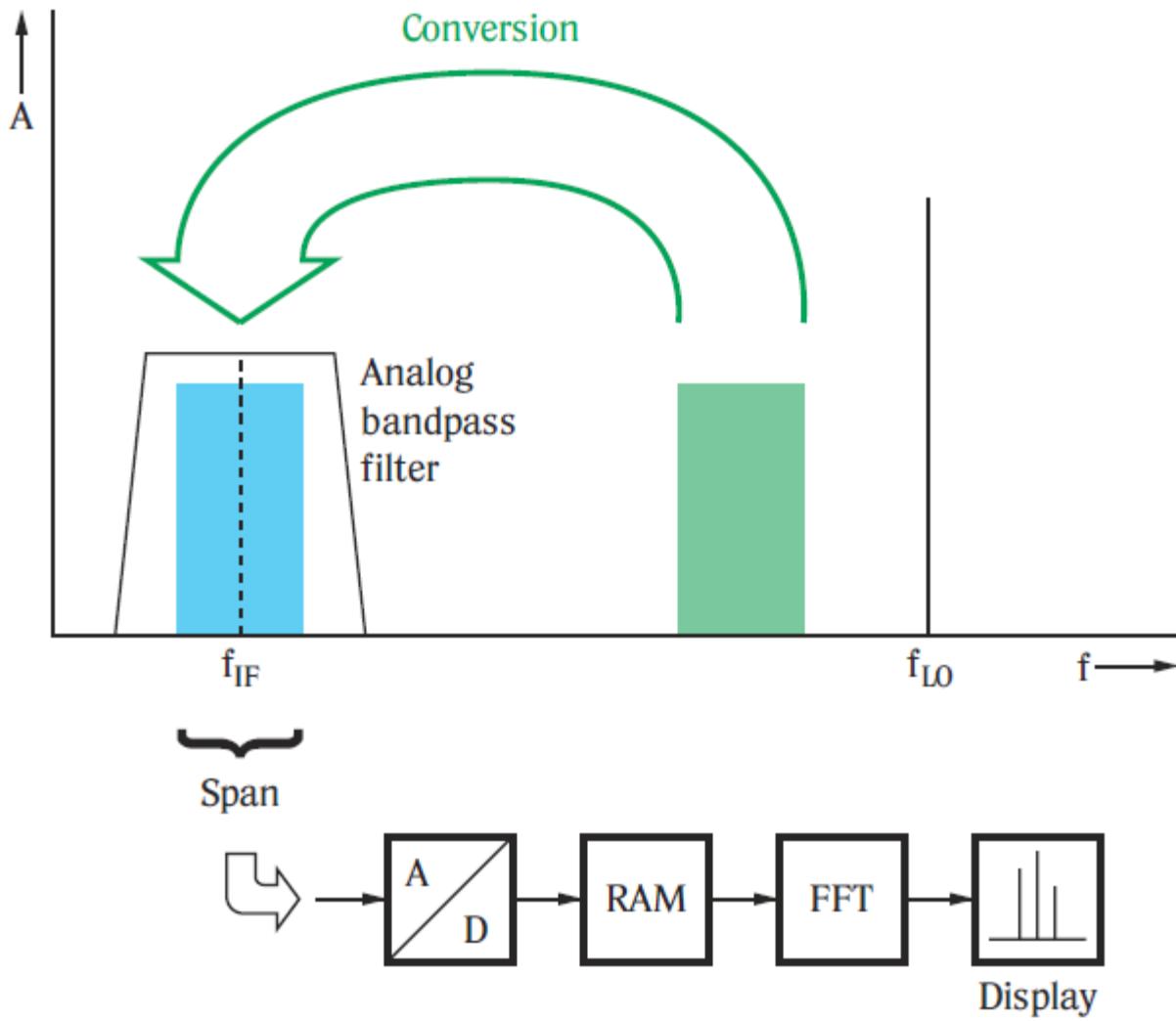
- ✓ Ne može da realizuje Gausov filter,
- ✓ Odlična aproksimacija samo do 20 dB,
- ✓ Odličan prelazni vremenski odziv,
- ✓ SF 10-14, pa i 4,6

## 2. Digitalni filtri

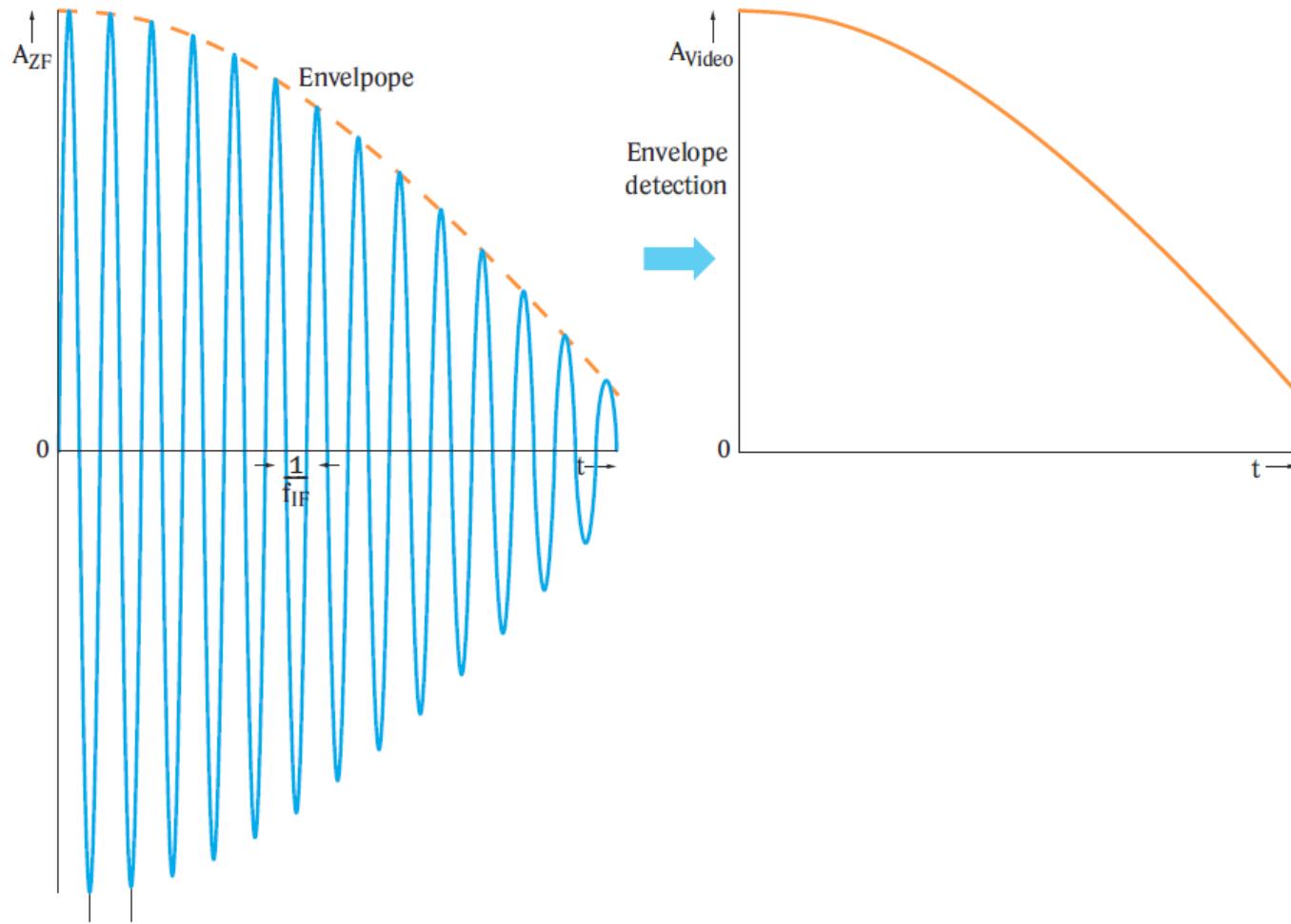
- ✓ Može da realizuje Gausov filter,
- ✓ Temperaturno stabilni nema efekata starenja,
- ✓ Bez podešavanja, kraće vreme sweep-a,
- ✓ Rezolucija od 10 Hz do 30 kHz.

## 3. Filtri na bazi FFT

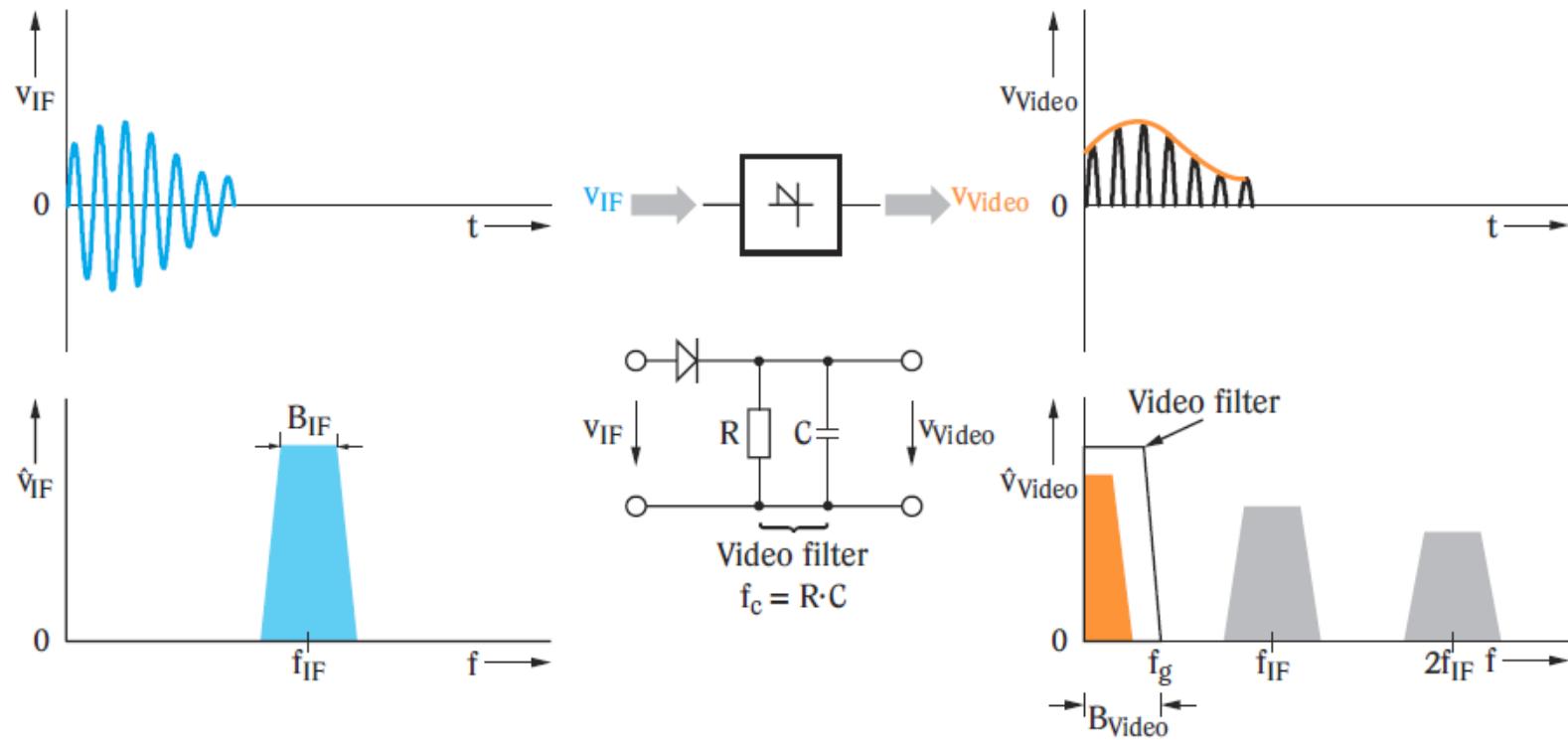
# Analizator spektra korišćenjem FFT



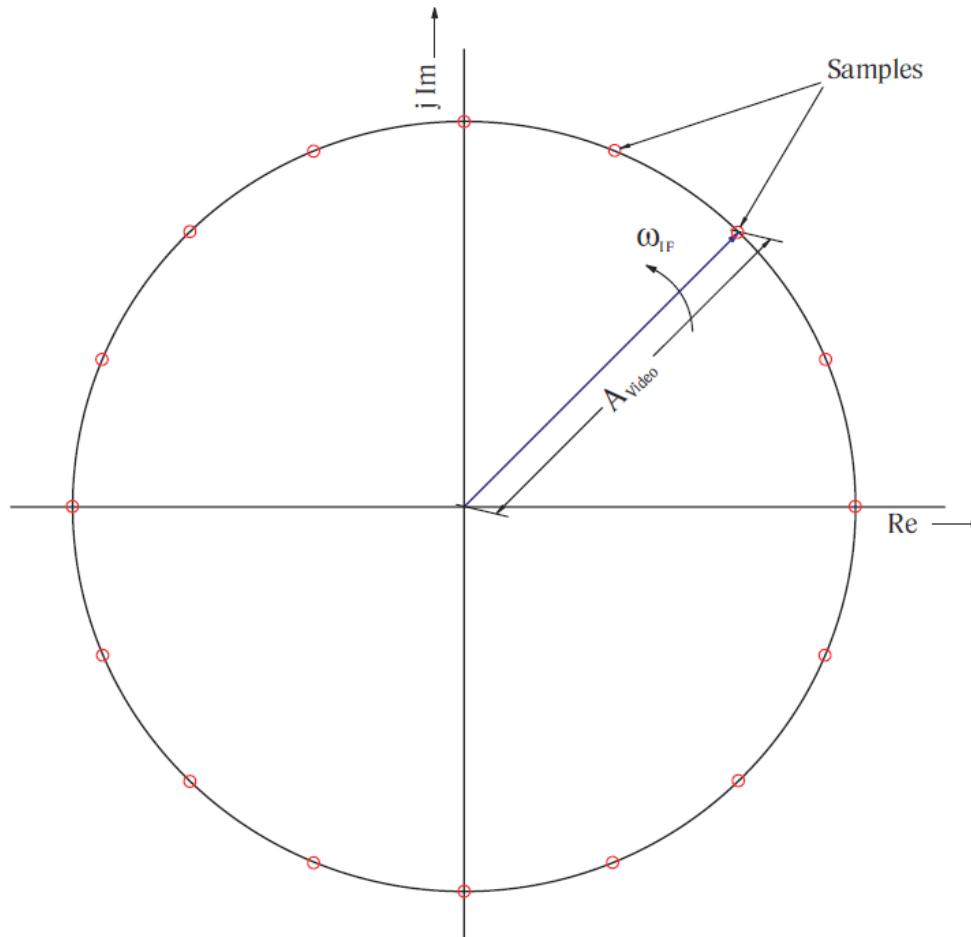
# Detekcija anvelope IF signala



# Detekcija IF anvelope

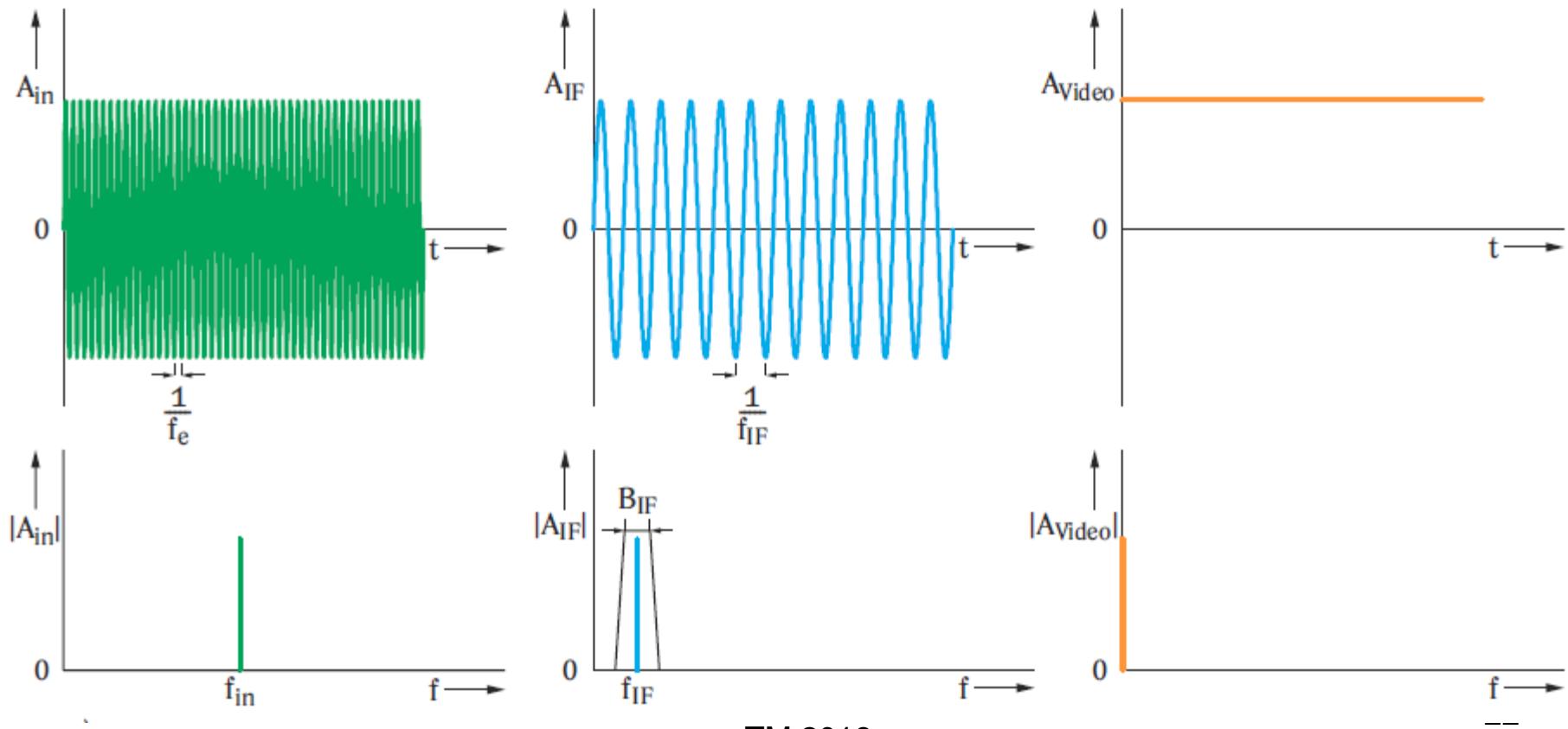


# Predstavljanje sinusoidalnog signala kompleksnim rotirajućim vektorom



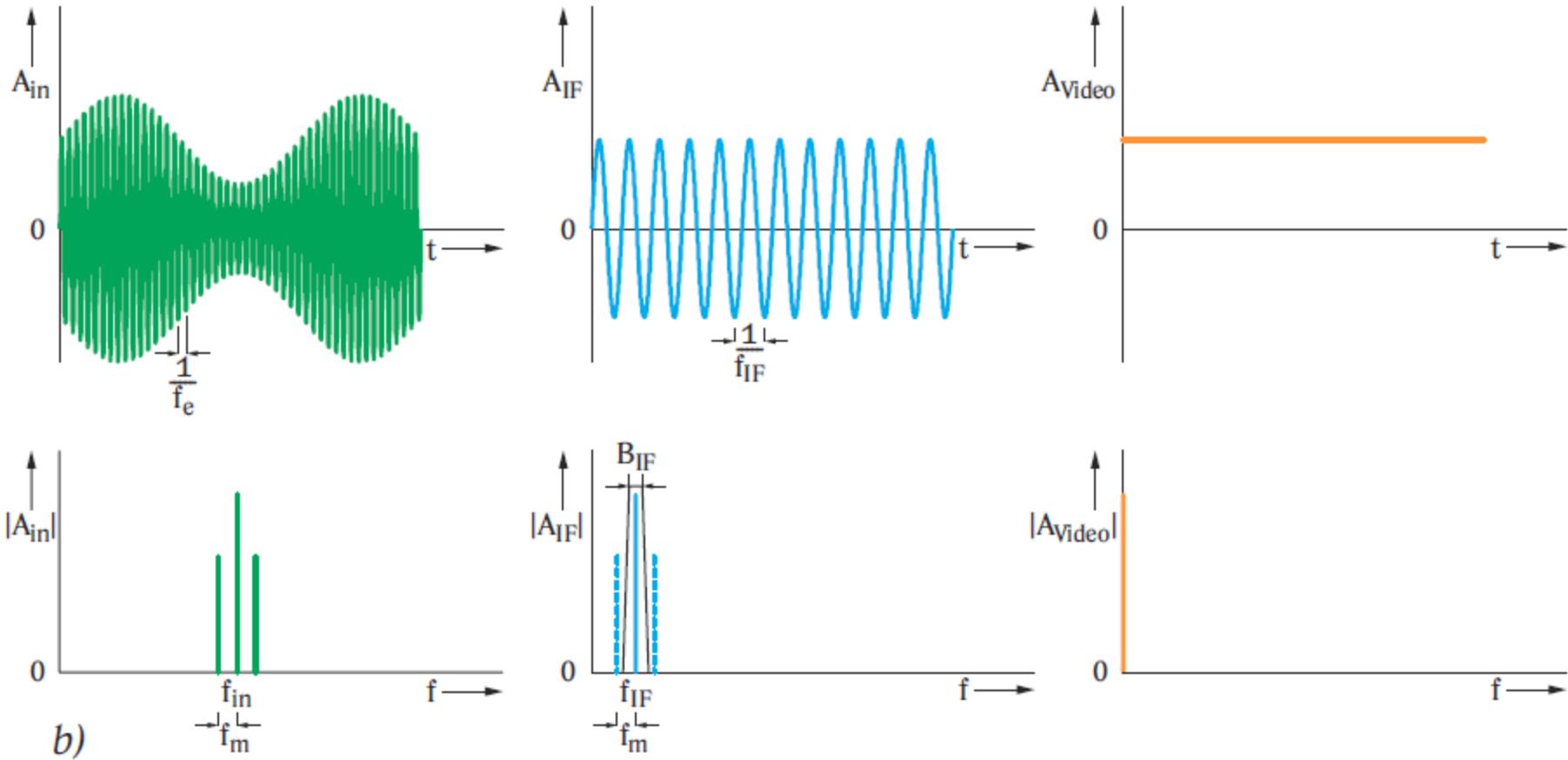
Video signal (žuto), IF signal posle IF filtra (plavo) za različite ulazne signale (zeleno) i rezultantni propusni opseg

sinusoidal signal



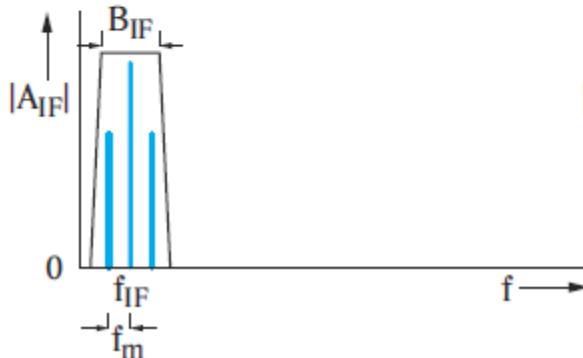
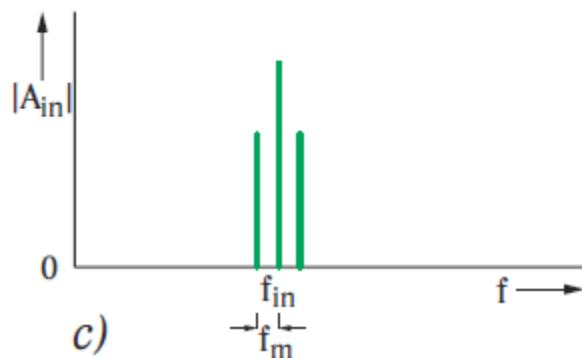
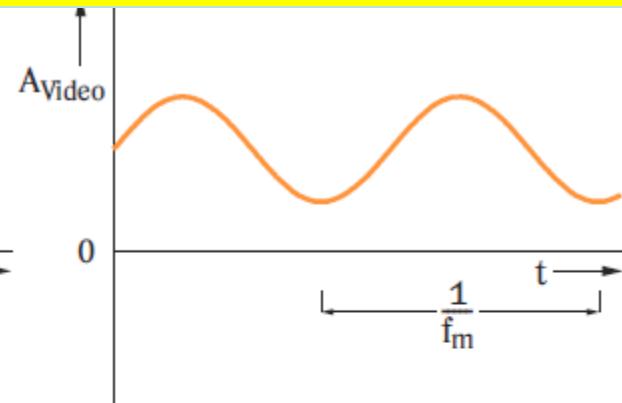
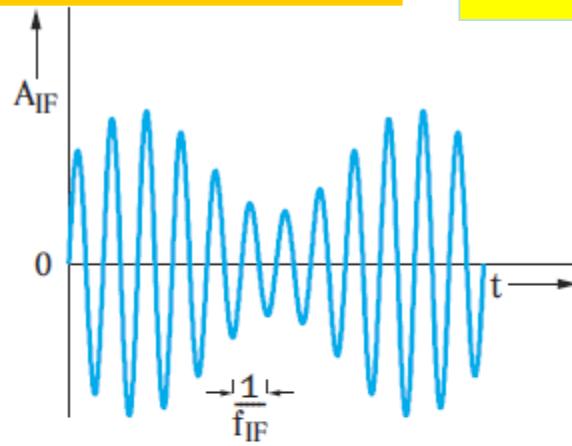
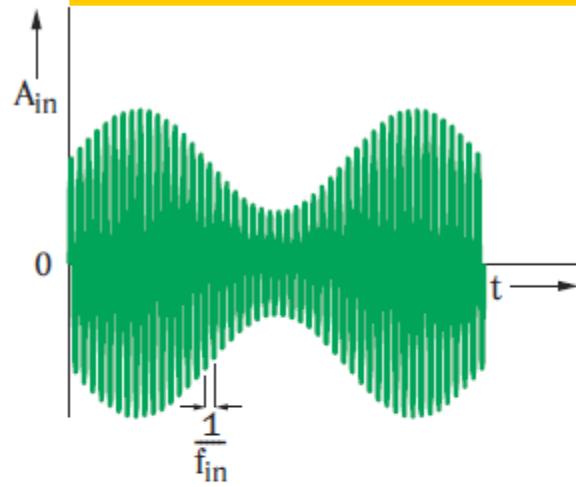
Video signal (žuto), IF signal posle IF filtra (plavo) za različite ulazne signale (zeleno) i resultantni propusni opseg

AM signal  
Opseg manji od 2 puta modulacionog opsega



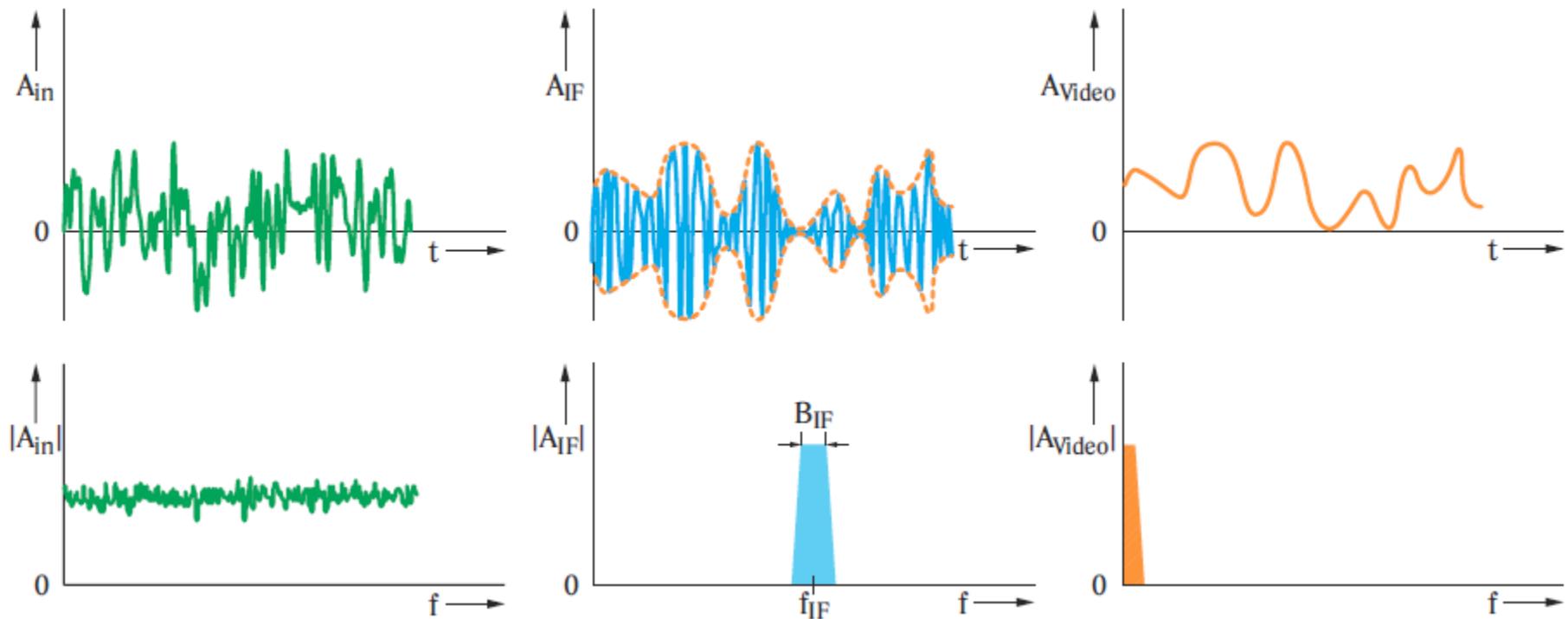
Video signal (žuto), IF signal posle IF filtra (plavo) za različite ulazne signale (zeleno) i resultantni propusni opseg

AM signal  
Opseg veći od 2 puta modulacionog opsega

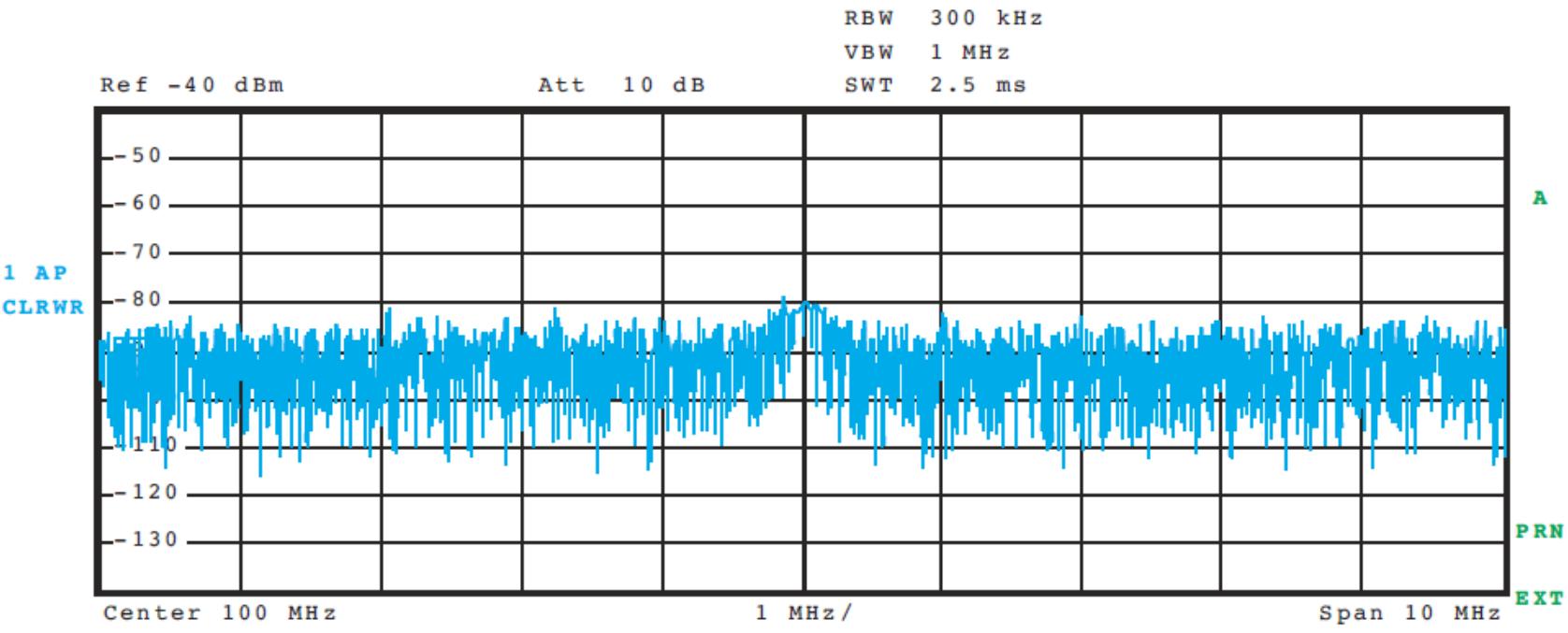


Video signal (žuto), IF signal posle IF filtra (plavo) za različite ulazne signale (zeleno) i resultantni propusni opseg

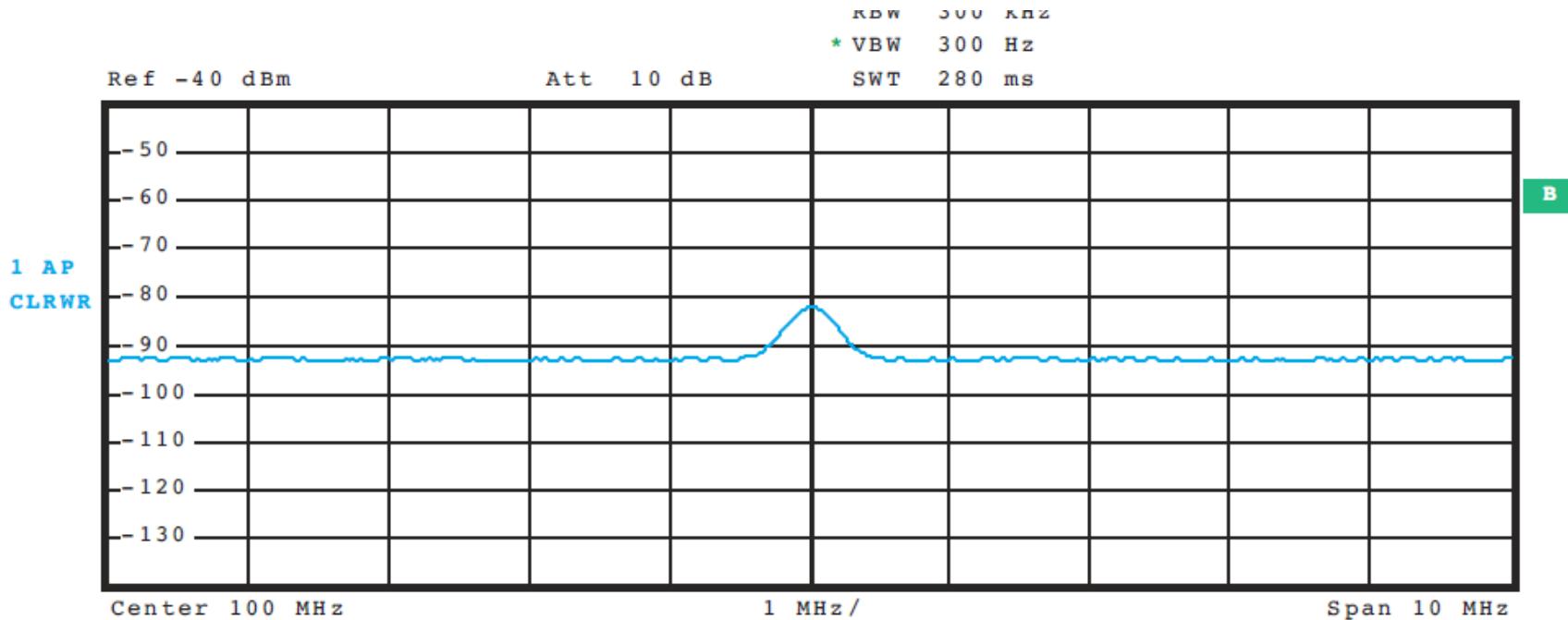
šum



# Sinusoidalan signal sa malim odnosom S/N za veliki propusni opseg

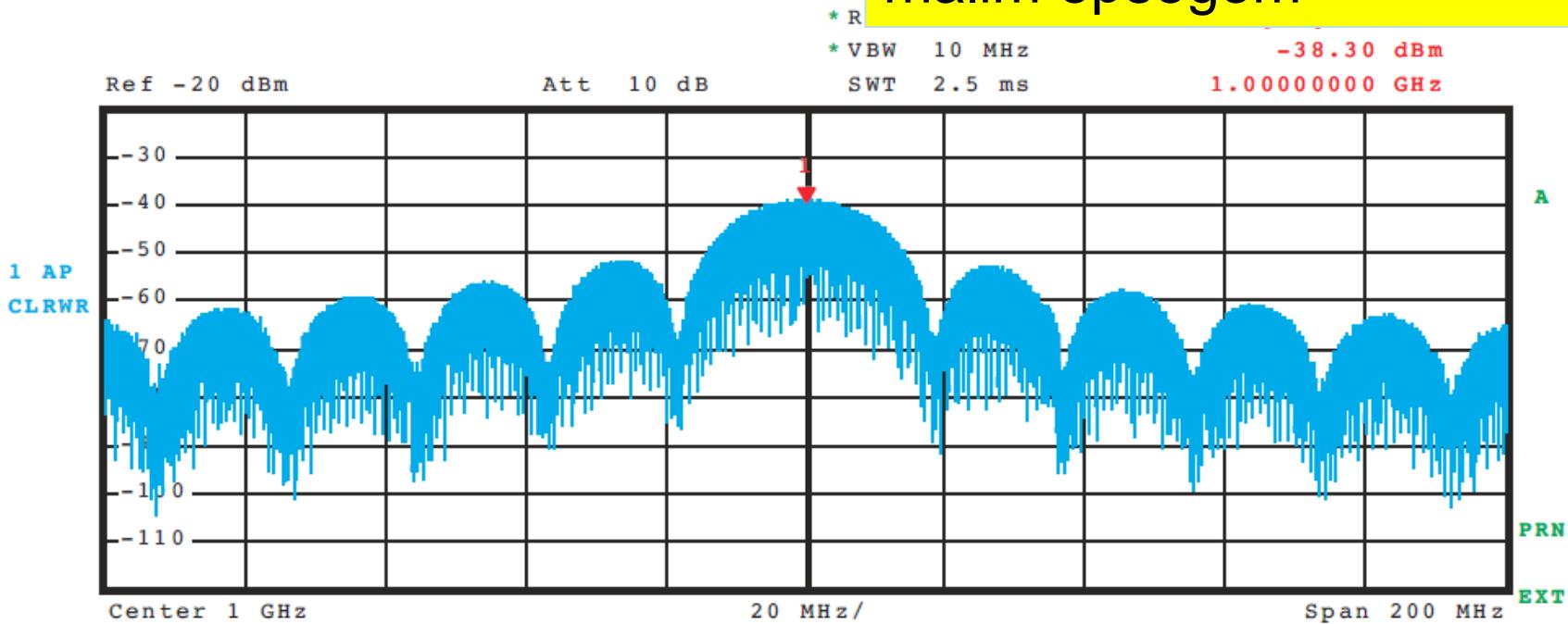


# Sinusoidalan signal sa malim odnosom S/N za mali propusni opseg



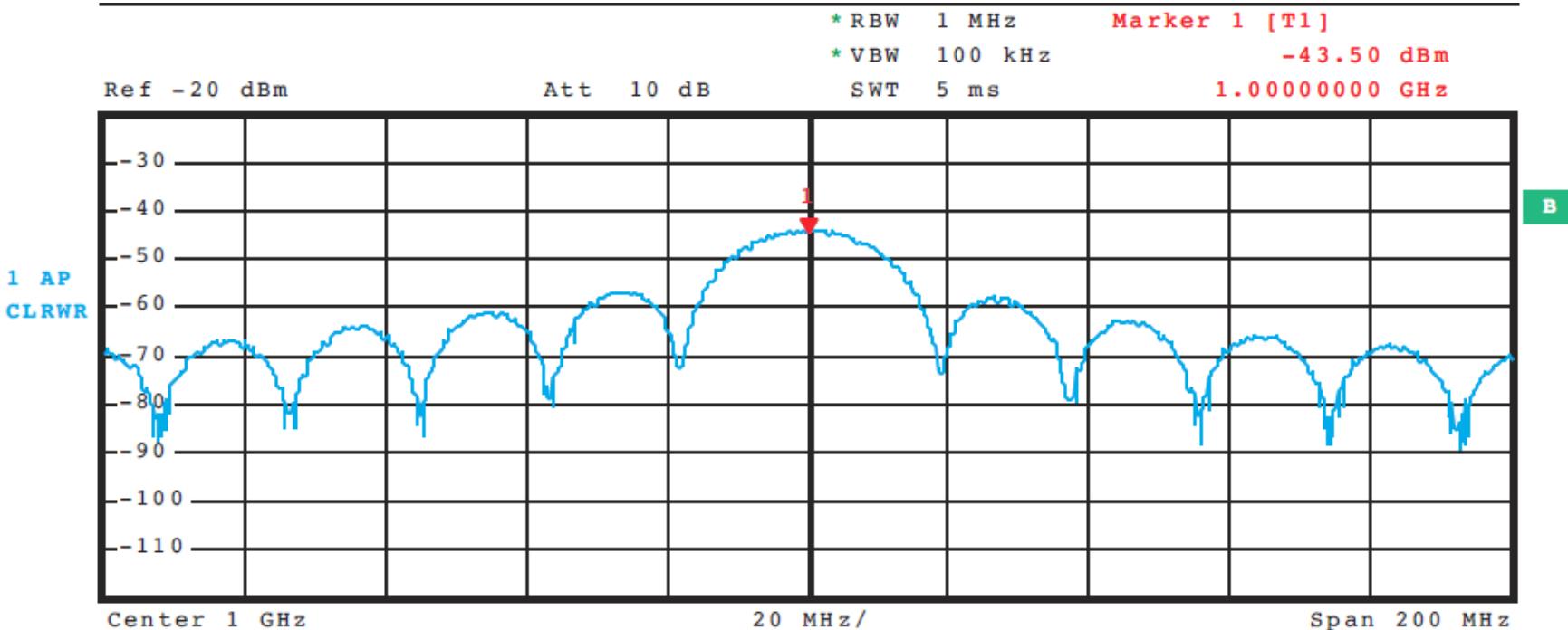
# Impulsni signal sa velikim video propusnim opsegom

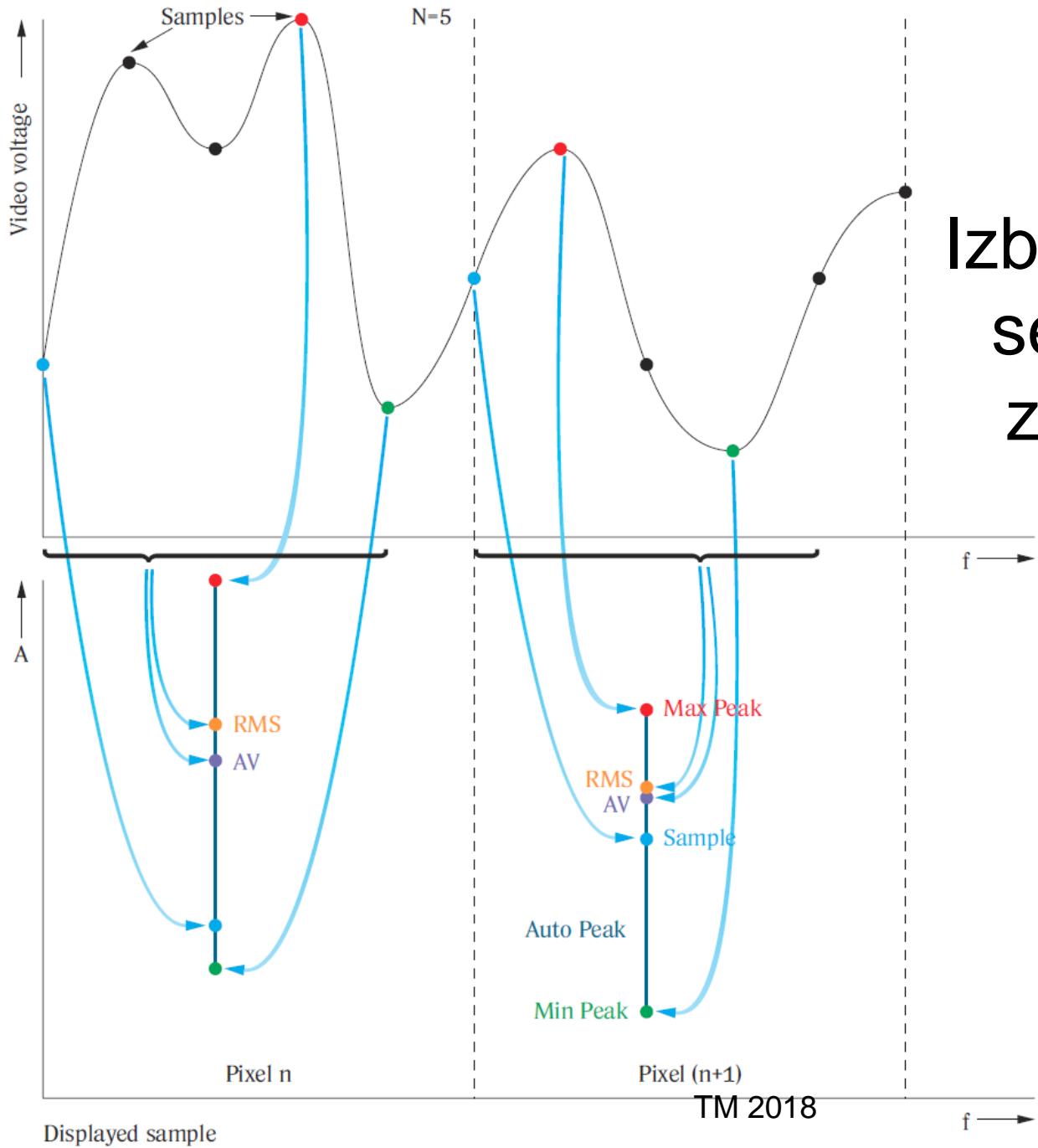
Marker  
Amplituda je oslabljena sa  
malim opsegom



# Impulsni signal sa malim video propusnim opsegom

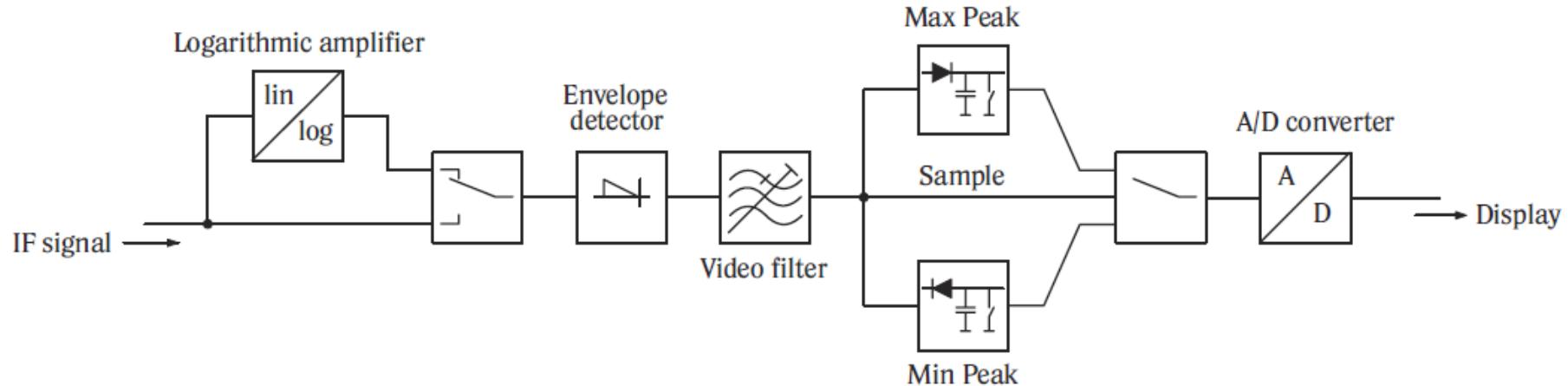
Marker  
Amplituda je oslabljena sa  
malim opsegom





Izbor sempla koji se prikazuje u zavisnosti od detektora

# Analogna realizacija detektora



Max peak detektor

Min peak detektor

Auto peak detektor

Sample detektor

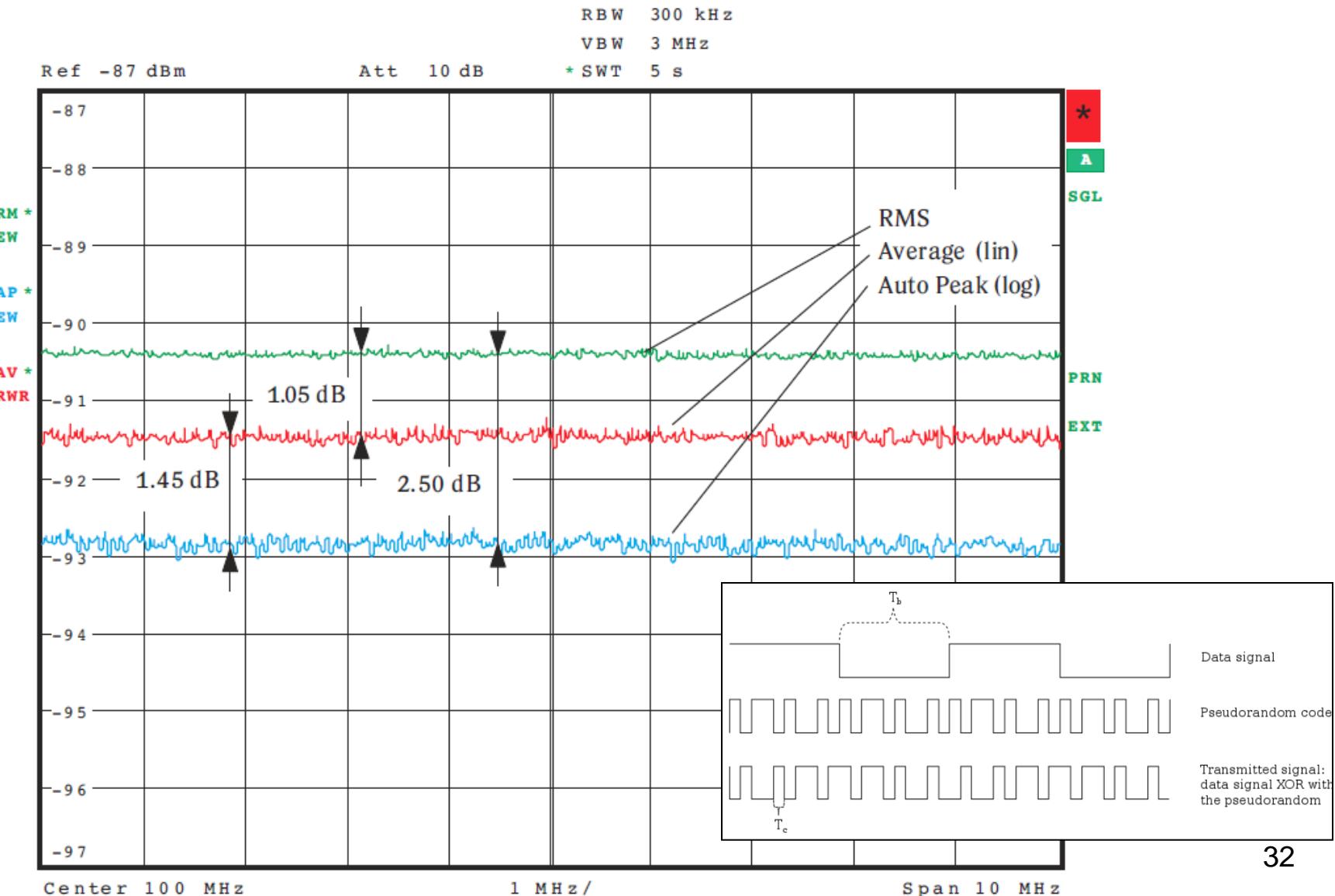
RMS (root mean square) detektor

AV (average) detektor

Quasi peak detector

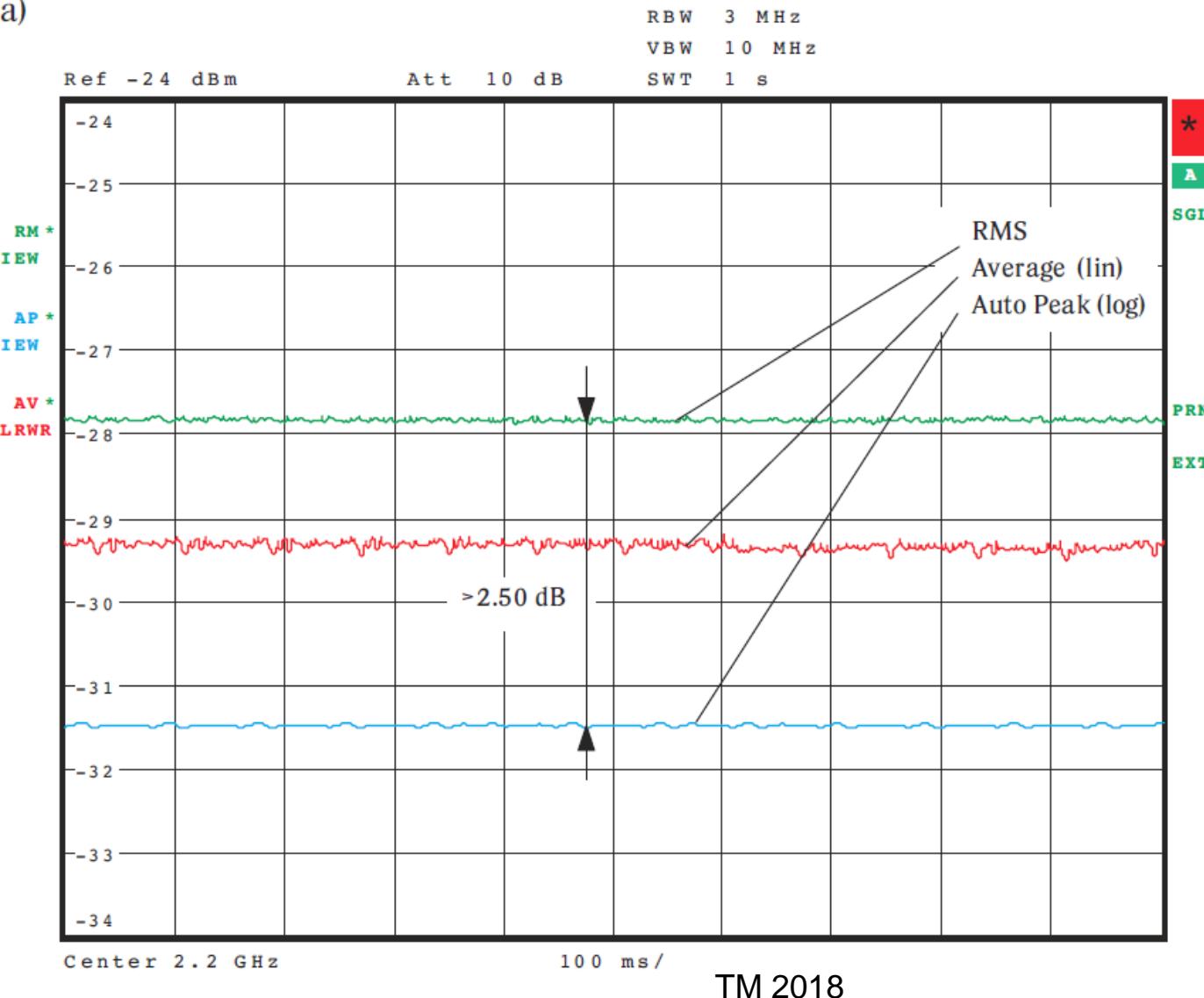
# CDMA Code-division multiple access

## Gausov šum i CDMA signal

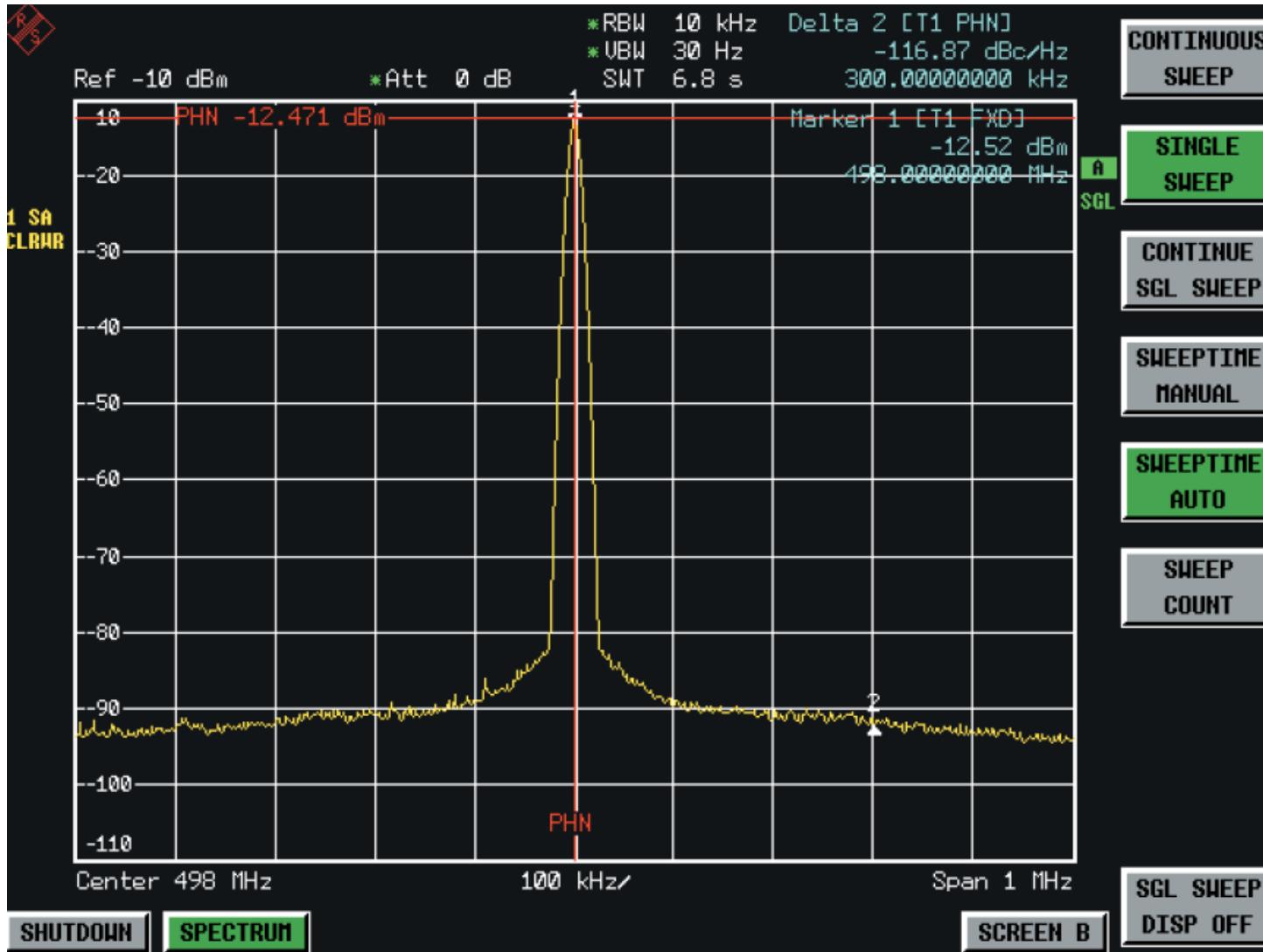


## Gausov šum i CDMA signal

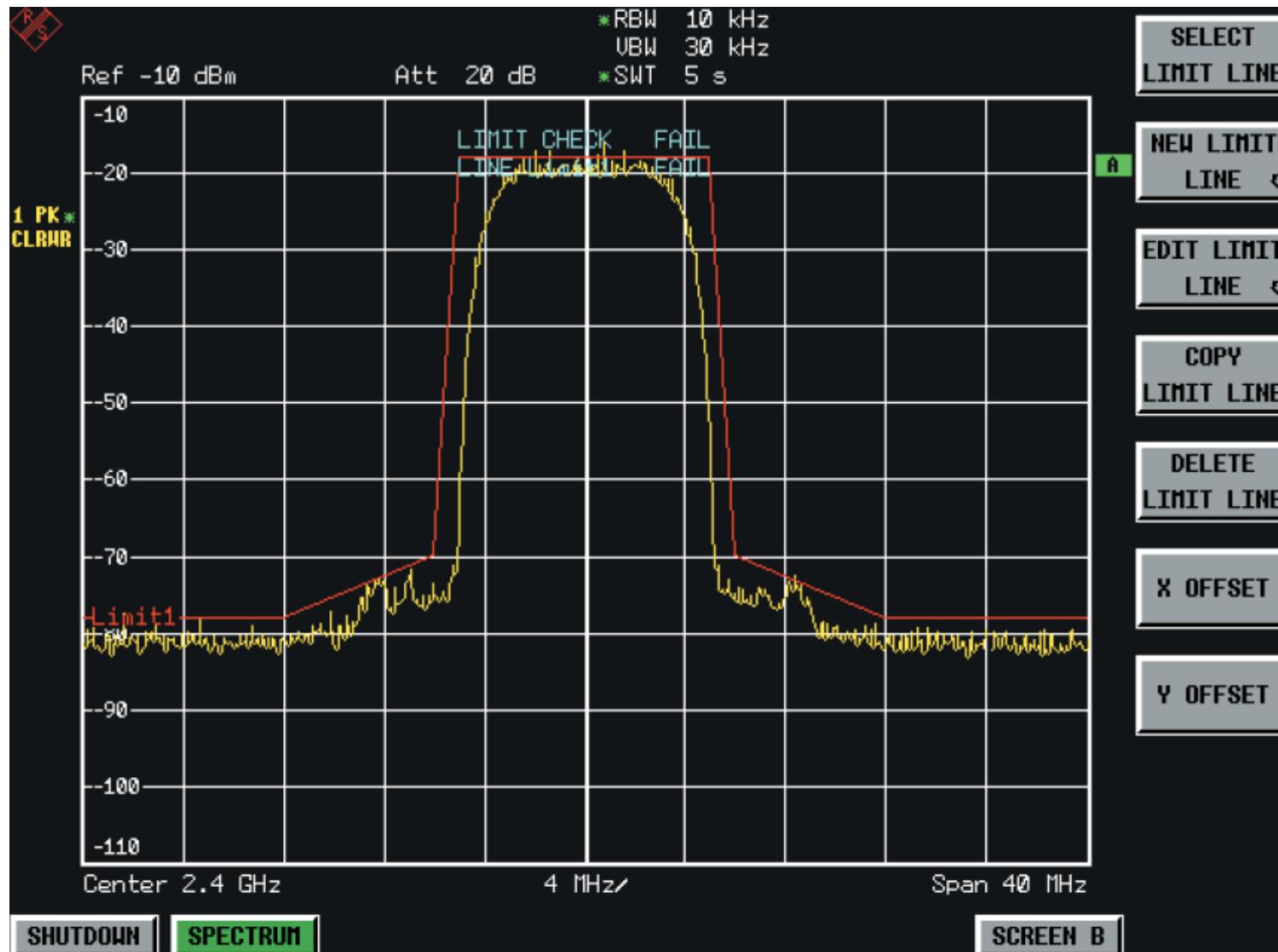
a)



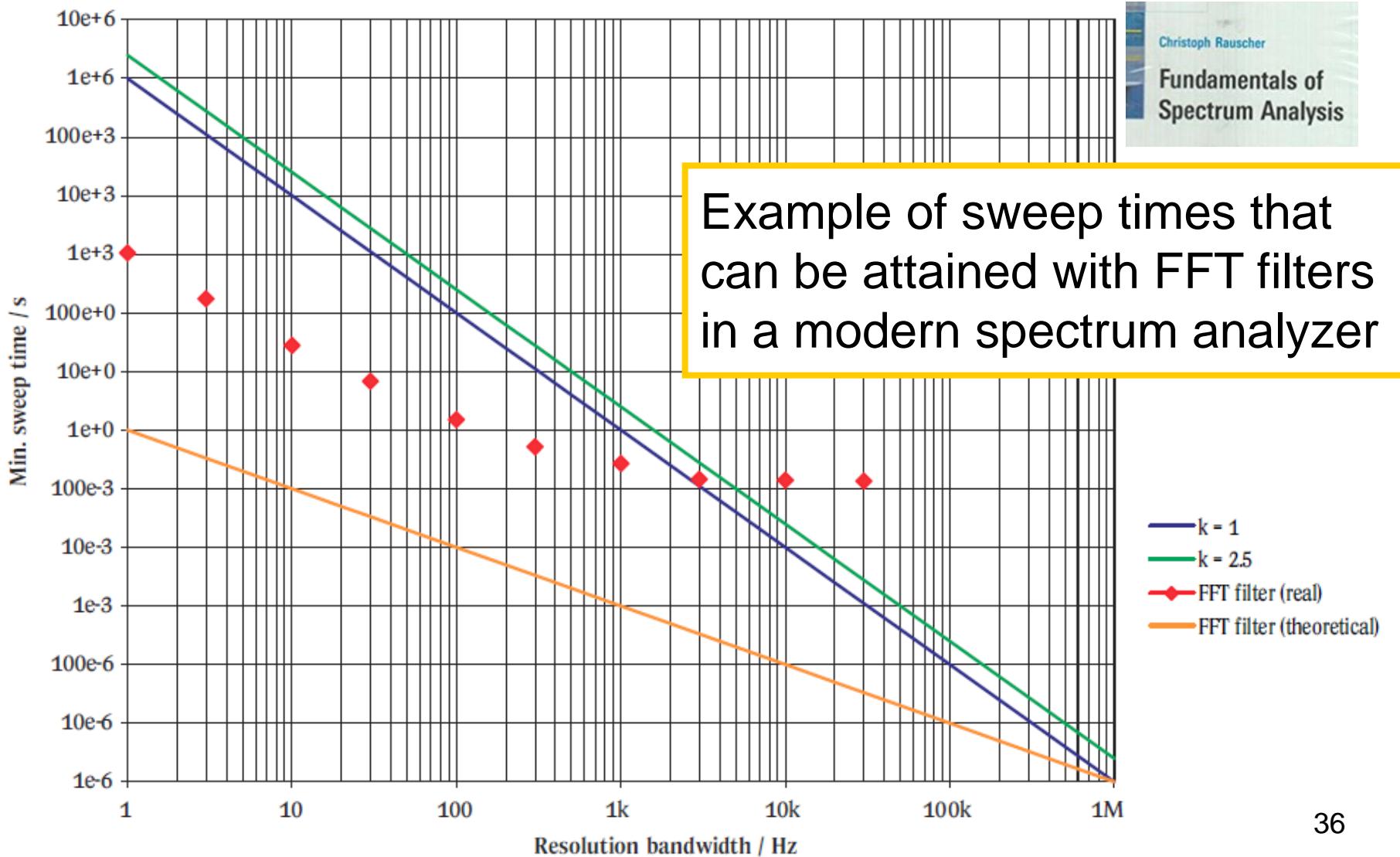
# Marker funkcije za jednostavno merenje faznog šuma ulaznog signala



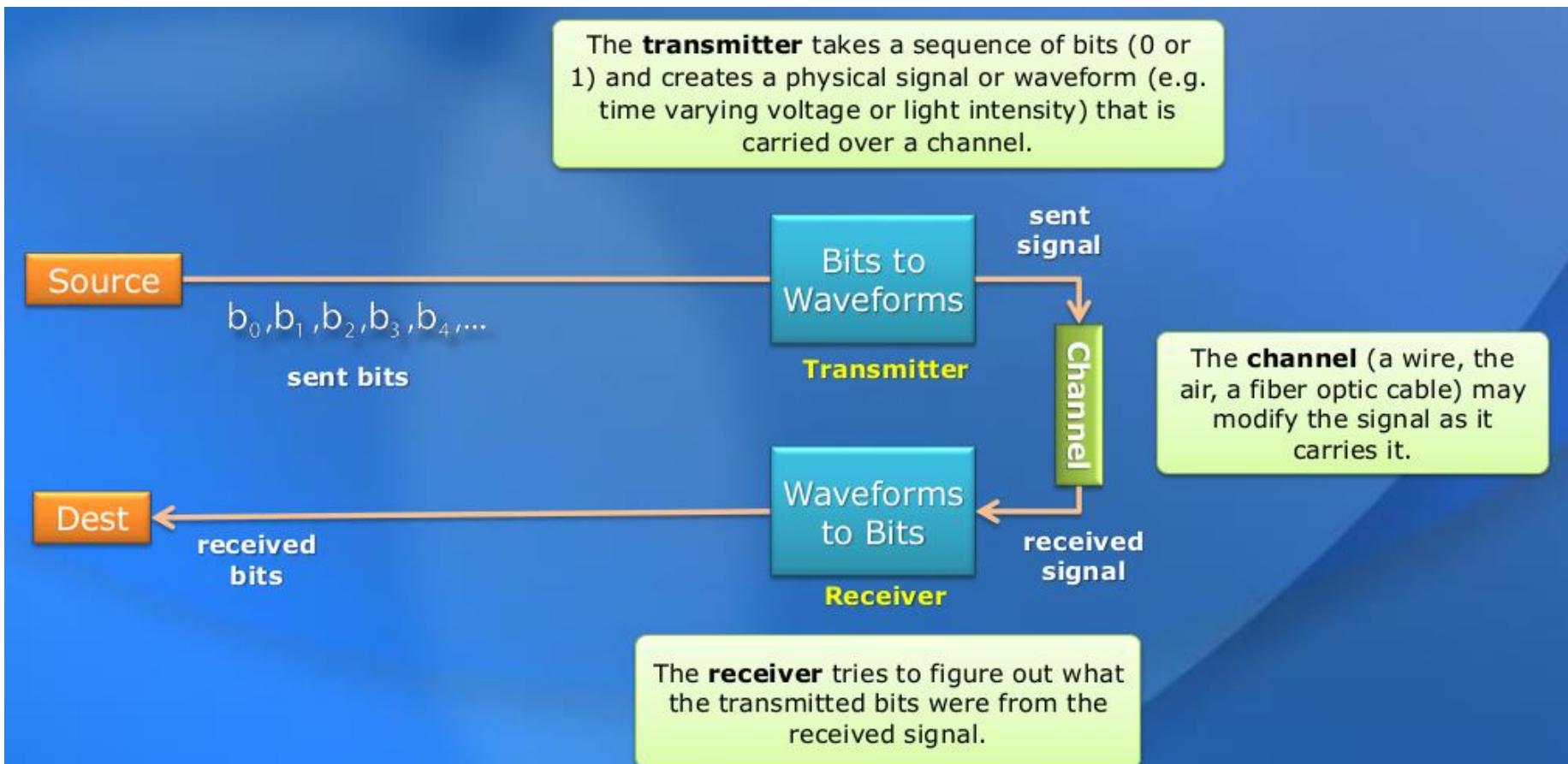
# Evaluation of traces with the aid of limit lines



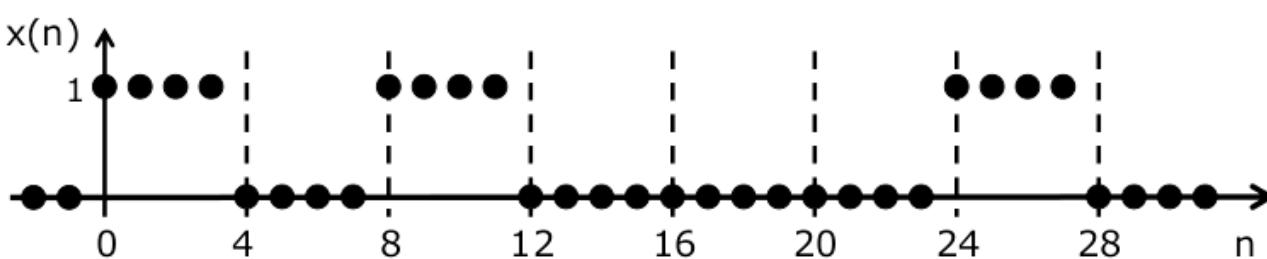
# Theoretically required sweep time as a function of resolution bandwidth at a span of 1 MHz



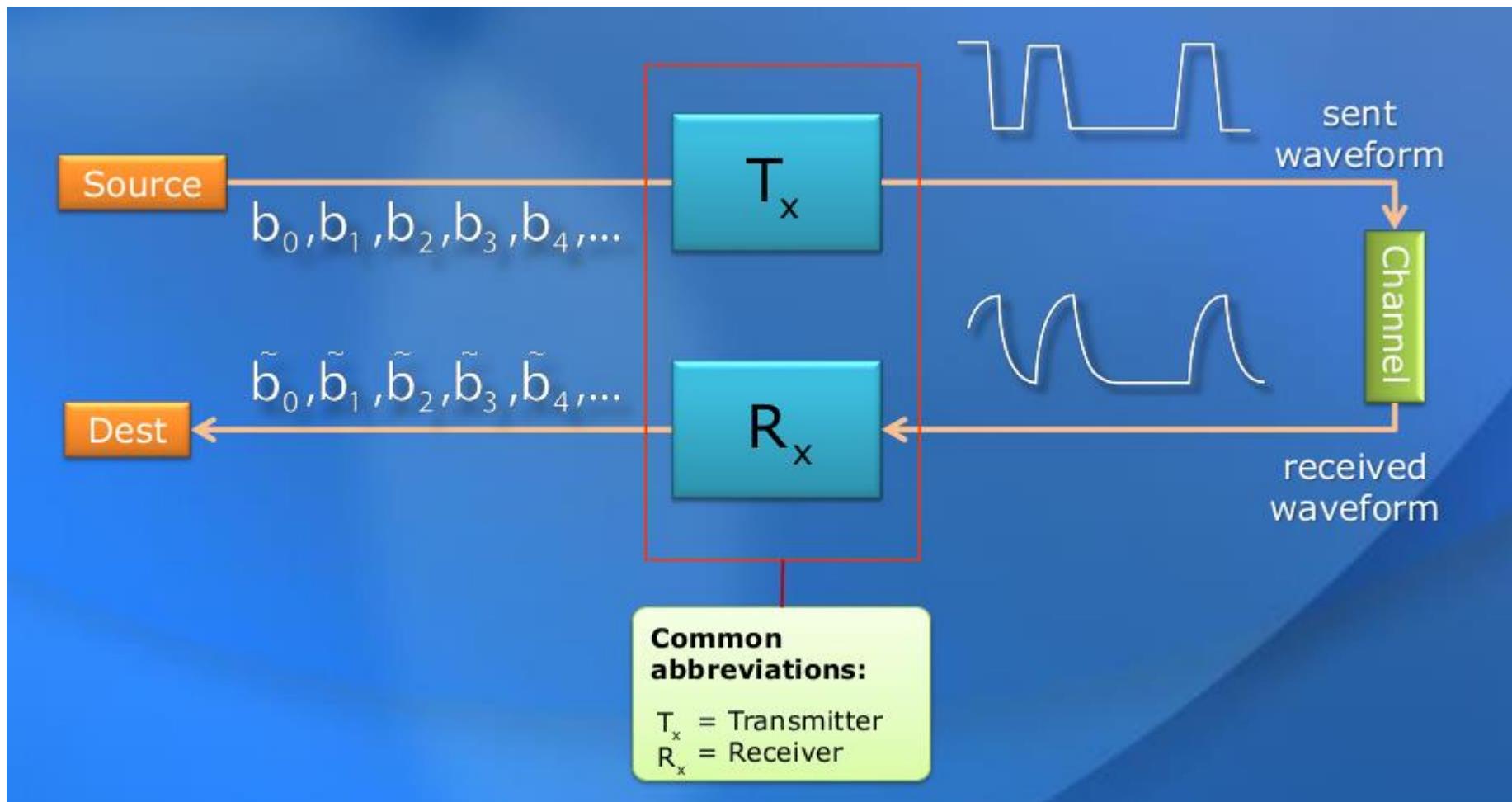
# Blok šema osnovnog digitalnog komunikacionog sistema



# Ekvivalentni načini predstavljanja bita

<b>Verbal</b>	"Encoding of the bit sequence 1,0,1,0,0,0,1 at 4 samples per bit"
<b>Graph</b>	
<b>List, table or vector of values</b>	$n = [0 \ 1 \ 2 \ 3 \ 4 \ 5 \ 6 \ 7 \ 8 \ 9 \ 10 \ 11 \ 12 \ 13 \ 14 \ 15 \ 16 \ 17 \dots]$ $x(n) = [1 \ 1 \ 1 \ 1 \ 0 \ 0 \ 0 \ 0 \ 1 \ 1 \ 1 \ 1 \ 0 \ 0 \ 0 \ 0 \ 0 \ 0 \dots]$
<b>Sum of unit step functions</b>	$x(n) = u(n) - u(n-4) + u(n-8) - u(n-12) + u(n-24) - u(n-28)$

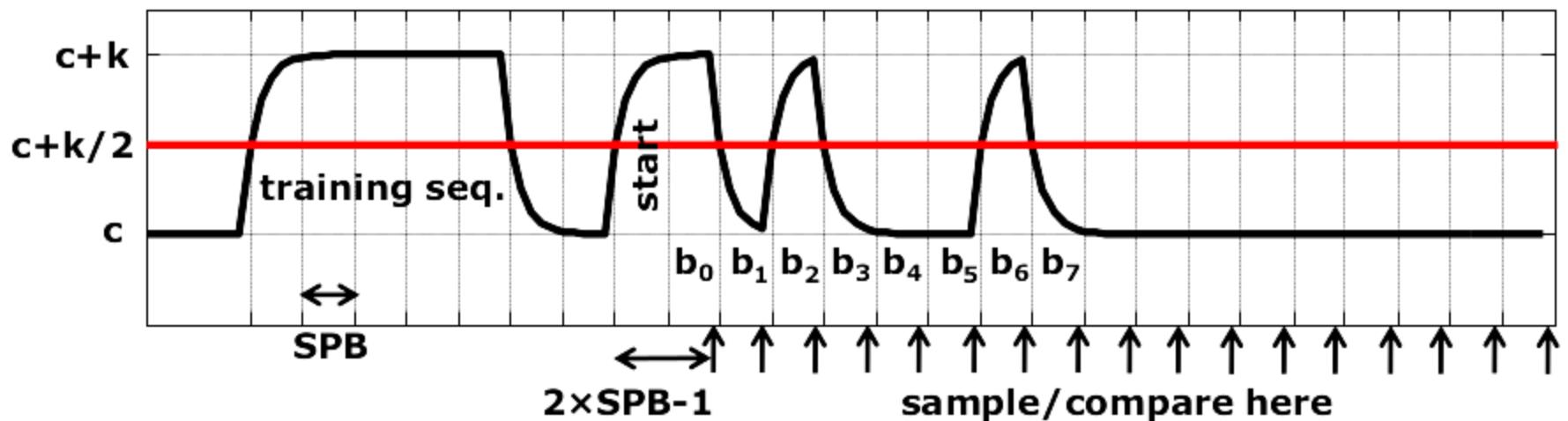
# Prenos talasnog oblika kroz kanal u komunikacionom sistemu



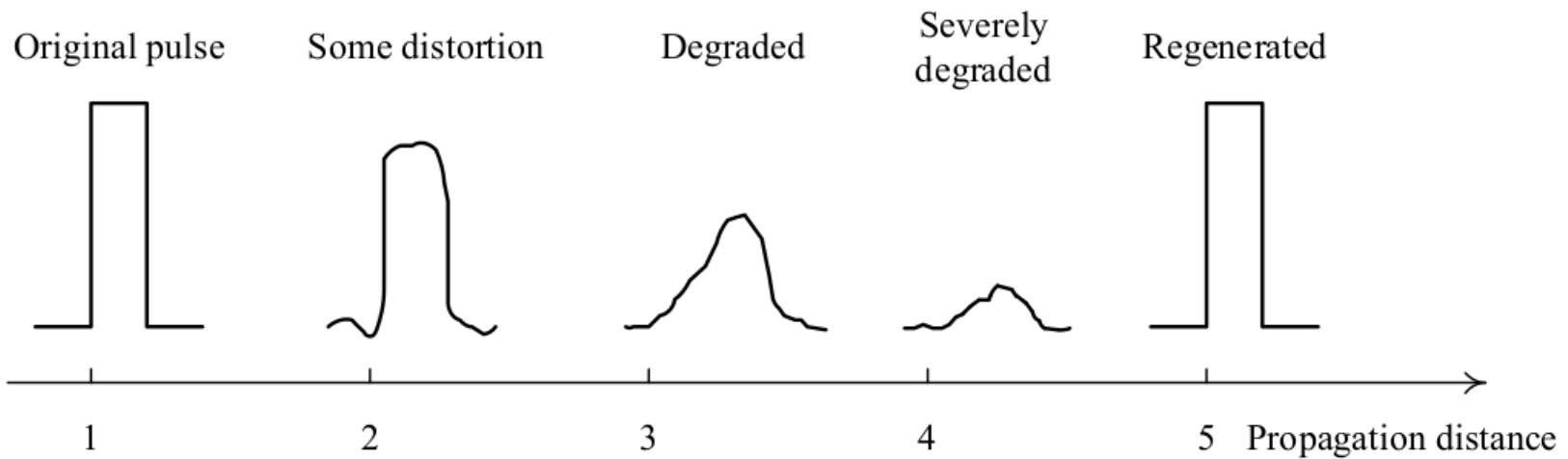
# Primer uticaja kanala za prenos na diskretni talasni oblik bita



# Primer jednostavnog protokola - prijemna strana



# Degradacija i regeneracija digitalnog signala



# Kodovanje reči: THINK korišćenjem 6-bitnog ASCII koda

Message (text):

"THINK"

Character coding  
(6-bit ASCII):

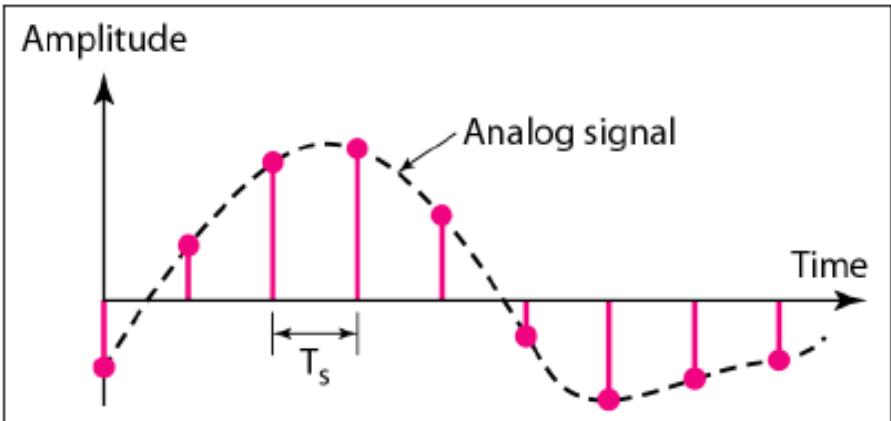
T	H	I	N	K
0 0 1 0 1 0	0 0 0 1 0 0	1 0 0 1 0 0	0 1 1 1 0 0	1 1 0 1 1 0
↓   ↓   ↓   ↓   ↓   ↓	↓   ↓   ↓   ↓   ↓   ↓	↓   ↓   ↓   ↓   ↓   ↓	↓   ↓   ↓   ↓   ↓   ↓	↓   ↓   ↓   ↓   ↓   ↓

8-ary digits  
(symbols):

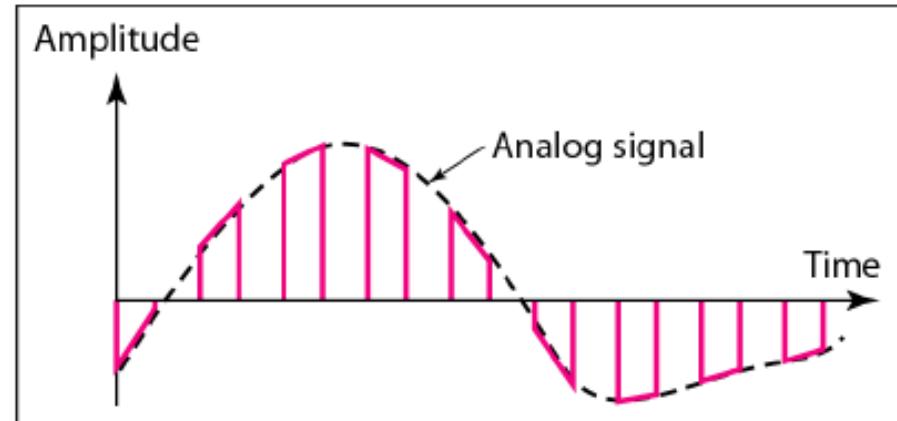
1 2 0 4 4 4 3 4 6 4

8-ary waveforms:  $s_1(t)$   $s_2(t)$   $s_0(t)$   $s_4(t)$   $s_4(t)$   $s_4(t)$   $s_3(t)$   $s_4(t)$   $s_6(t)$   $s_4(t)$

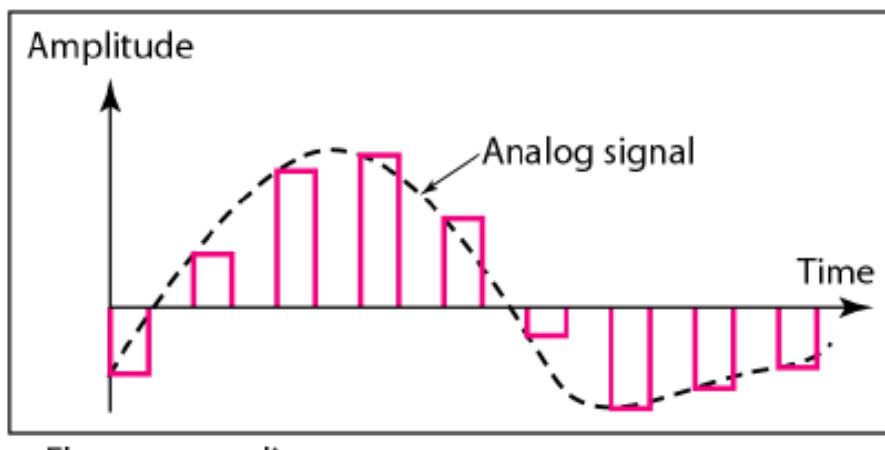
# Idealno, Flat - Top odabiranje i prirodno odabiranje



a. Ideal sampling

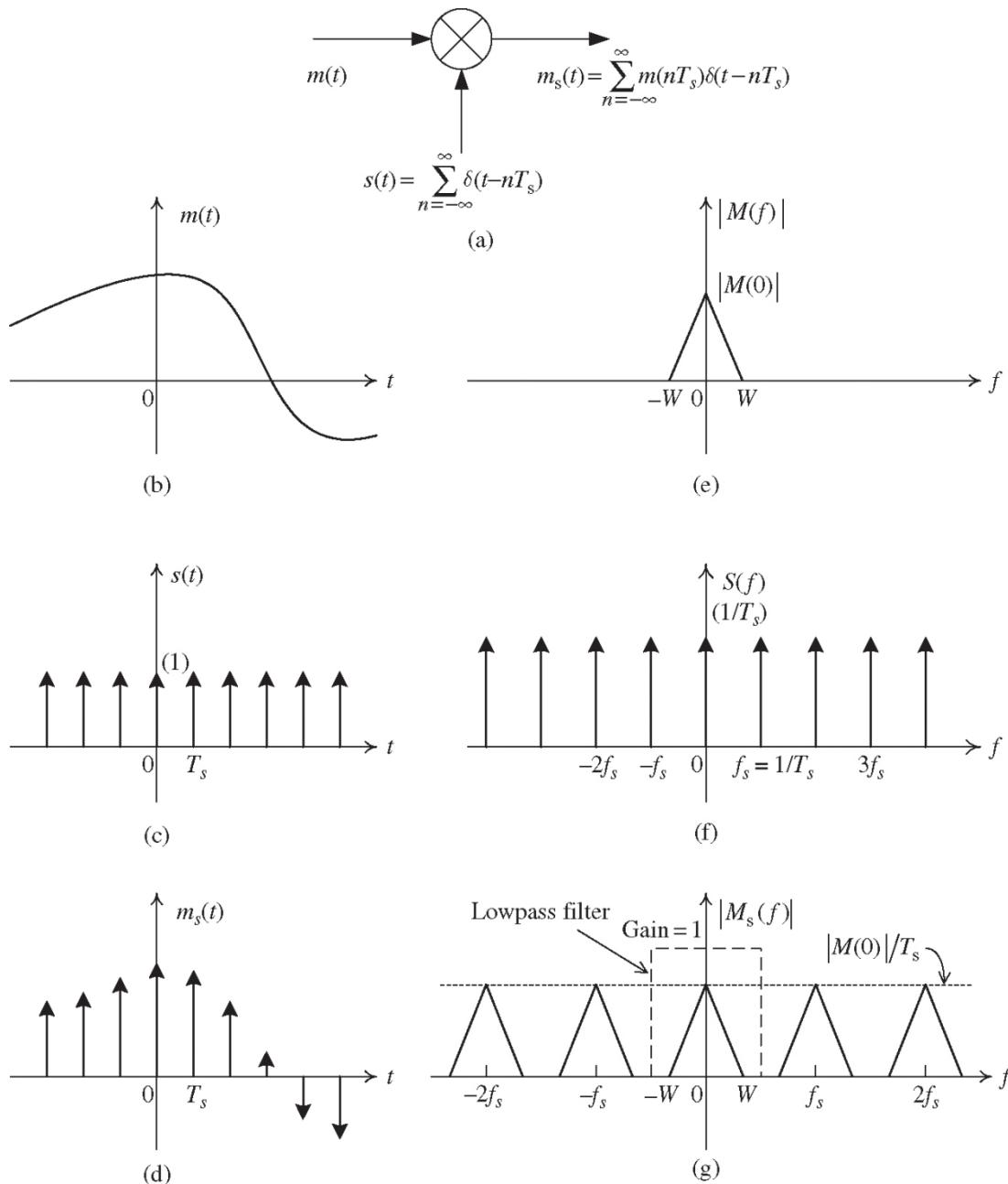


b. Natural sampling

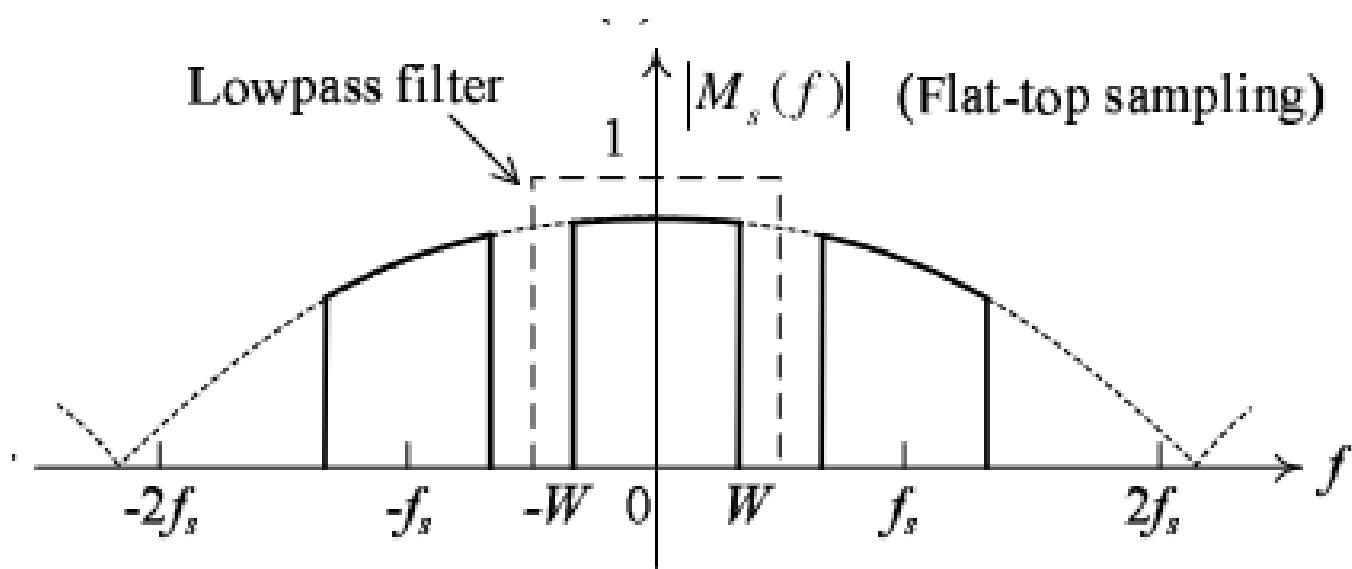


c. Flat-top sampling

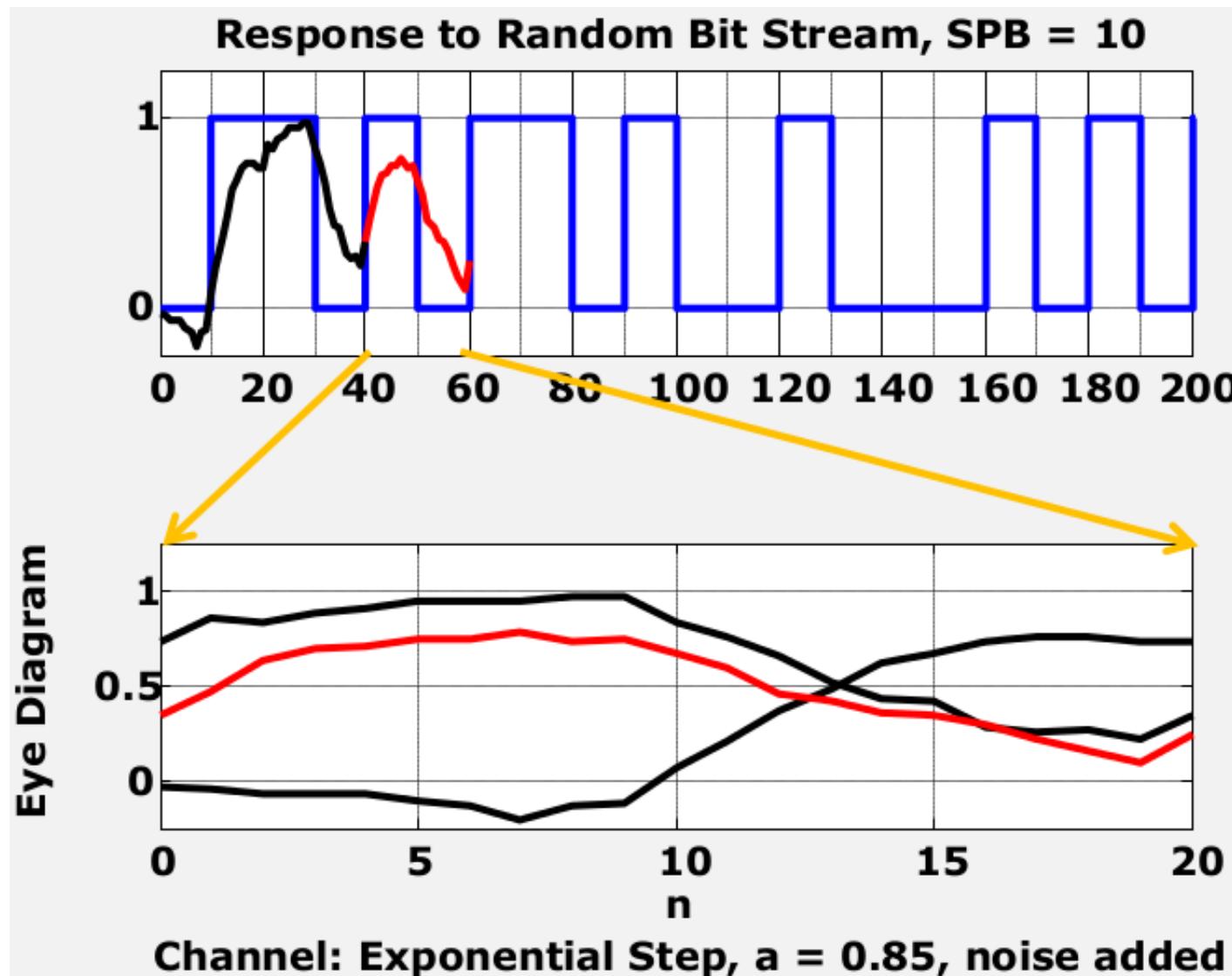
# Ilustracija idealnog odabiranja - vremenski i frekvenčijski domen



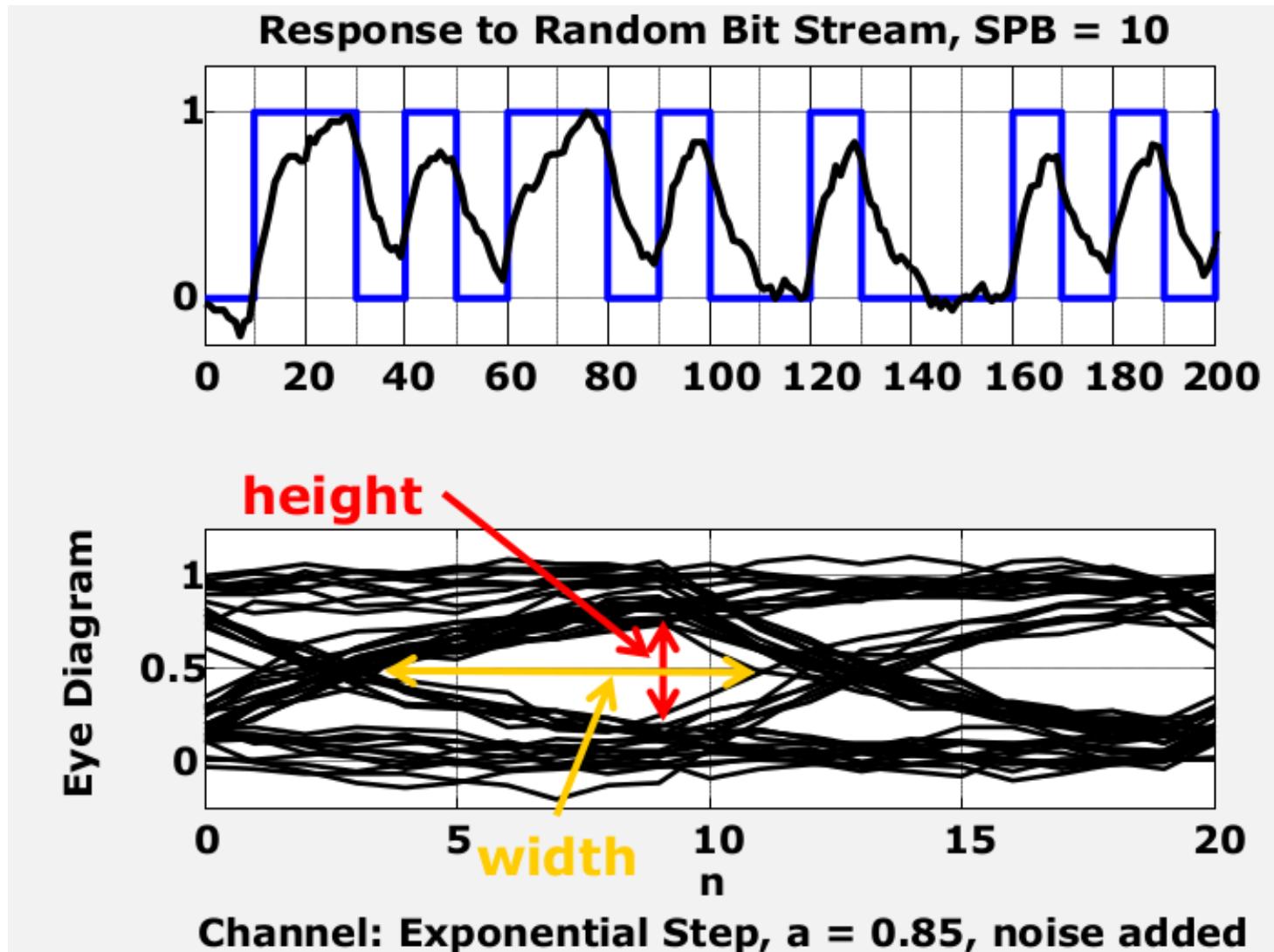
# Frekvencijski domen signala koji je flat-top odabiran



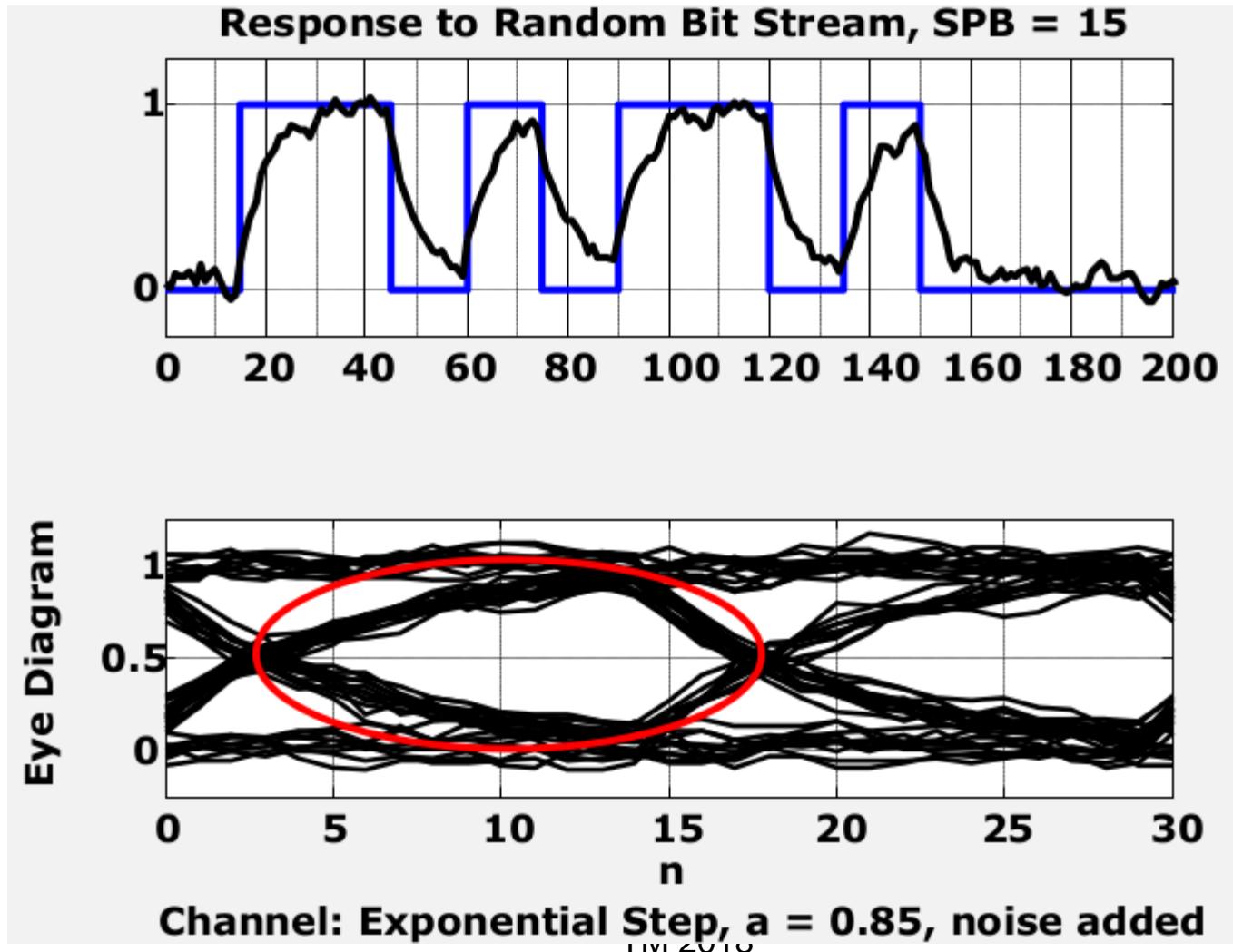
# Konstrukcija dijagrama oka - rezultat nakon 60 odbiraka, SPB = 10



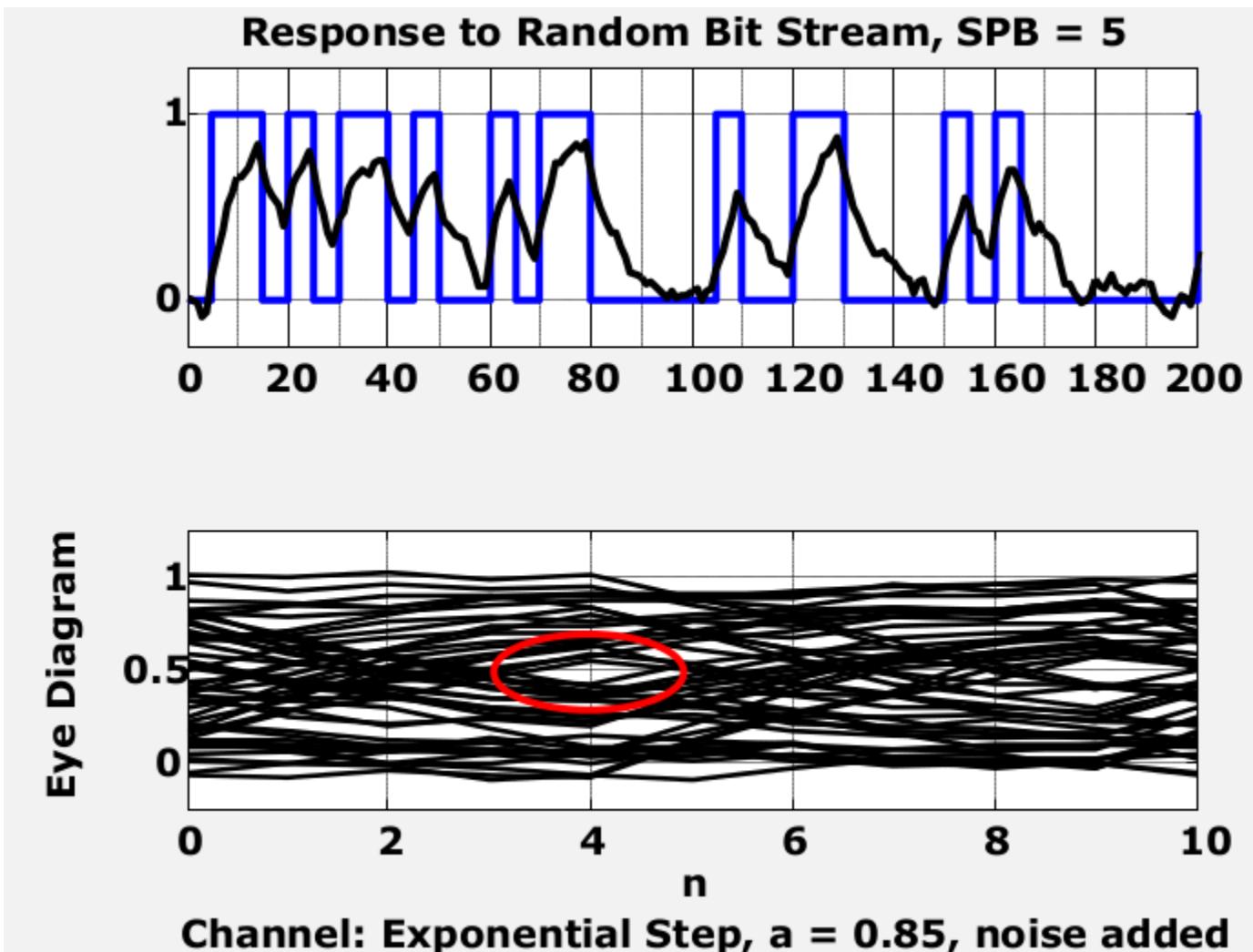
# Konstrukcija dijagrama oka - rezultat nakon 200 odbiraka, SPB = 10



# Dijagram oka za SPB = 15

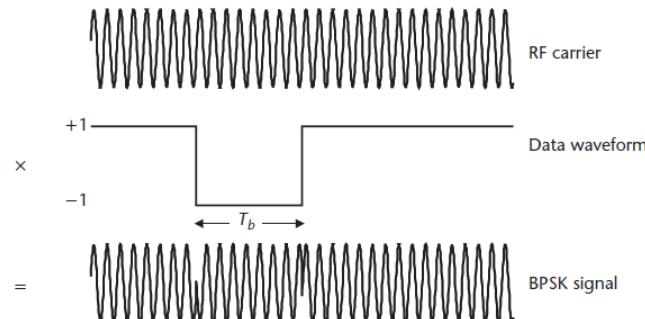


# Dijagram oka za SPB = 5

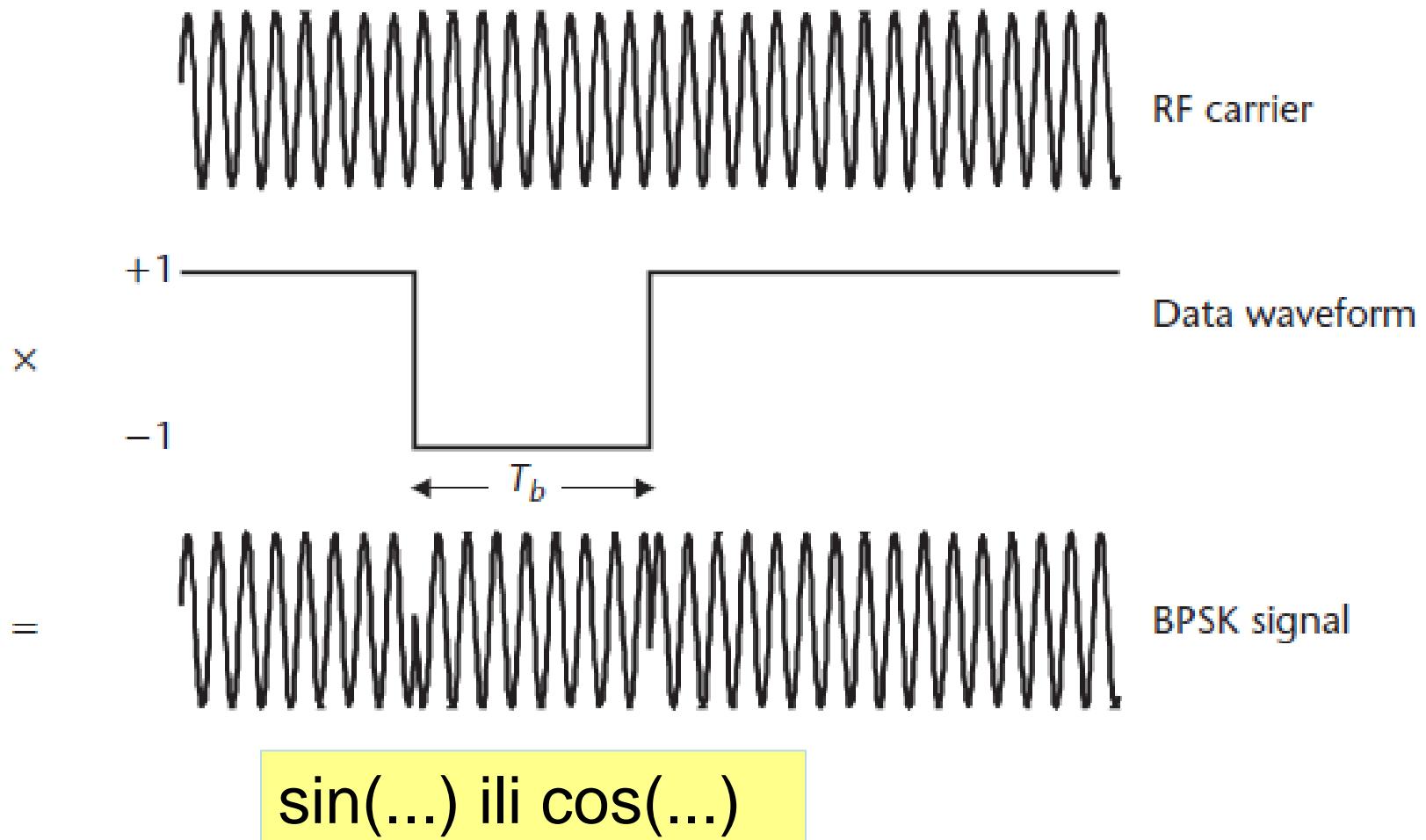


# Sistem za digitalni prenos, bežični kanal

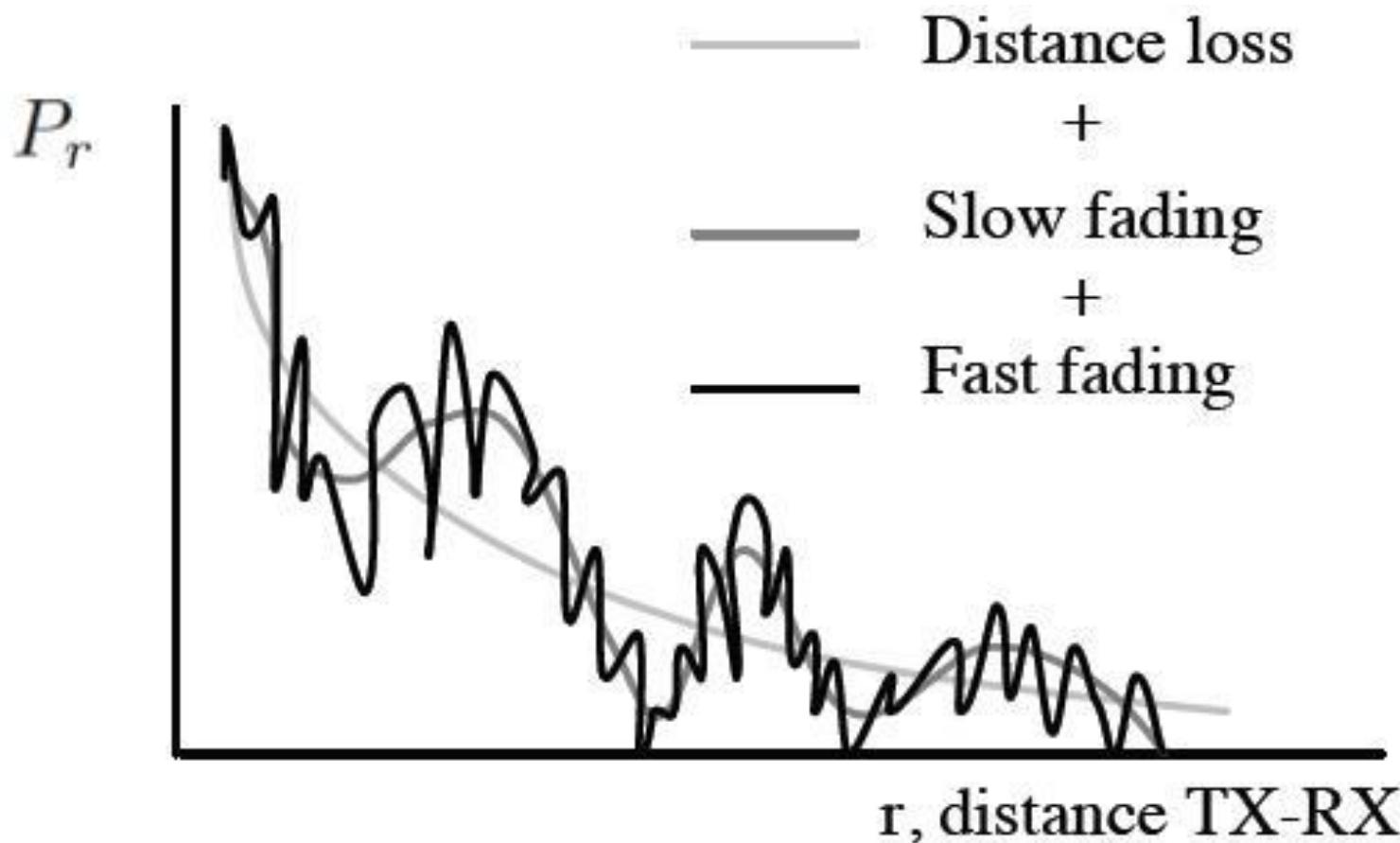
- PSK (Phase Shift Keying) modulacija, digitalna fazna modulacija
- PSK modulacija će biti objašnjena na primeru binarne PSK modulacije kada se prenose samo biti 0 ili 1
- Za prenos signala neophodan je RF nosilac, čija je frekvencija za WSN tipično jednaka  $f_c = 2,4 \text{ GHz}$   
Za prestavljanje bita 0,  $s(t)$   
signal nakon modulatora je  $\cos(2 \pi f_c t)$ ,  
dok je predstavljanje bita 1 fazno pomereno za  $\pi$ ,  
odnosno jednako  $\cos(2 \pi f_c t + \pi) = -\cos(2 \pi f_c t)$



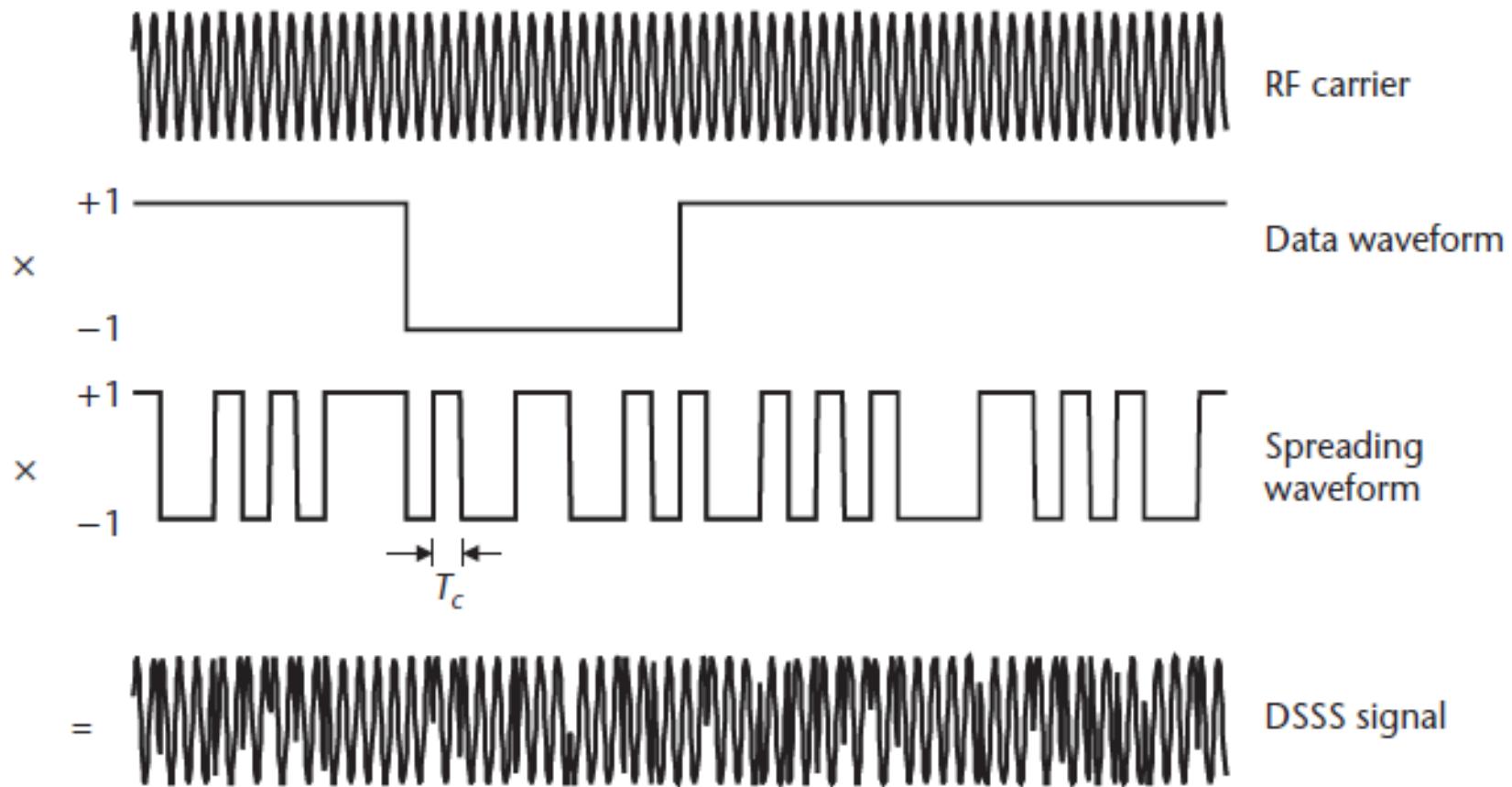
# BPSK modulacija



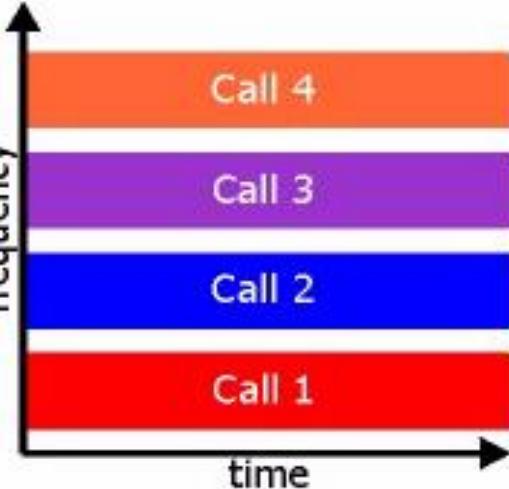
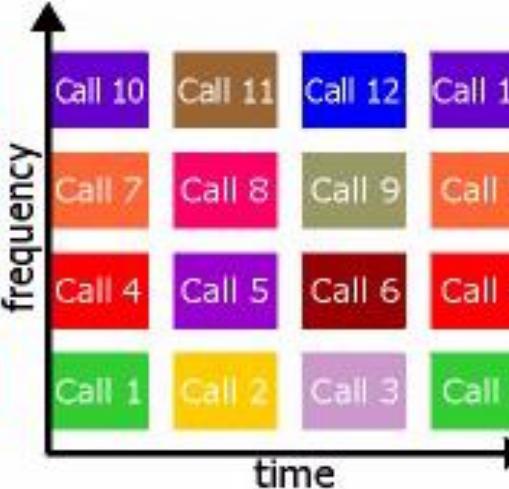
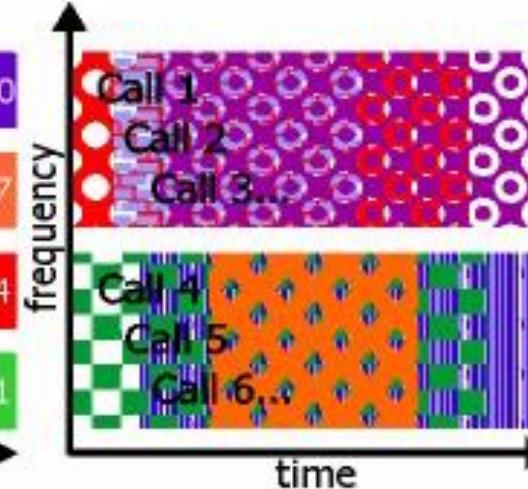
Promena snage koja stiže na prijemnik  $P_R$  sa rastojanjem, usled *shadowing* fedinga i *multi-path* propagacije



# Direct sequence spread spectrum (DSSS) je proširenje BPSK modulacije



# FDMA, TDMA, CDMA.

	FDMA	TDMA	CDMA
Time vs. Frequency			
Conversation Analogy	Everyone talks in a different room to prevent interference. Since the conversation can't be heard from another room, it can be filtered from the other by going to the other room.	Within each room, everyone takes turns talking to prevent interference. Within each room, one person is talking at once, so they must talk fast to say everything.	Everyone speaks a different language at the same time in the same room. Since each language is unique, one may be filtered from another.

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**mlutovac@viser.edu.rs**

**Ova prezentacija je nekomercijalna.**

Slajdovi mogu da sadrže materijale preuzete sa Interneta, stručne i naučne građe, koji su zaštićeni Zakonom o autorskim i srodnim pravima.

Ova prezentacija se može koristiti samo privremeno tokom usmenog izlaganja nastavnika u cilju informisanja i upućivanja studenata na dalji stručni, istraživački i naučni rad i u druge svrhe se ne sme koristiti –

Član 44 - Dozvoljeno je bez dozvole autora i bez plaćanja autorske naknade za nekomercijalne svrhe nastave:

- (1) javno izvođenje ili predstavljanje objavljenih dela u obliku neposrednog poučavanja na nastavi;
- ZAKON O AUTORSKOM I SRODNIM PRAVIMA ("Sl. glasnik RS", br. 104/2009 i 99/2011)