

# **EITG05 – Digital Communications**

#### Week 2, Lecture 2

#### Bandwidth of Transmitted Signals





### **Fourier transform**

$$\begin{aligned} X(f) &= \mathcal{F}\{x(t)\} = \int_{-\infty}^{\infty} x(t) \ e^{-j2\pi f t} \ dt \\ &= X_{Re}(f) + j \ X_{Im}(f) \\ &= |X(f)| \ e^{j \varphi(f)} \end{aligned}$$

$$\begin{aligned} x(t) &= \mathcal{F}^{-1}\{X(f)\} = \int_{-\infty}^{\infty} X(f) \ e^{+j2\pi f t} \ df \\ &= \int_{-\infty}^{\infty} |X(f)| \ e^{+j(2\pi f t + \varphi(f))} \ df \end{aligned}$$

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### Week 2, Lecture 2

Chapter 2: Model of a Digital Communication System

> 2.5 The bandwidth of the transmitted signal

2.5.5 R(f): *M*-ary PAM signals 2.5.6 R(f): *M*-ary QAM signals 2.5.7 R(f): OFDM-type of signals

2.5.8 R(f): M-ary FSK signals

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Exercises: 2.21a,b, 2.22, 2.23, 2.25, 2.29



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## Some useful Fourier transform properties

g(at)	$\leftrightarrow$	$\frac{1}{ a } \ G(f/a)$	$g^*(T-t)$	$\leftrightarrow$	$G^*(f)e^{-j2\pi fT}$
g(-t)	$\leftrightarrow$	G(-f)	$\delta(t)$	$\leftrightarrow$	1
G(t)	$\leftrightarrow$	g(-f)	1(dc)	$\leftrightarrow$	$\delta(f)$
$g(t-t_0)$	$\leftrightarrow$	$G(f)e^{-j2\pi ft_0}$	$e^{j2\pi f_c t}$	$\leftrightarrow$	$\delta(f-f_c)$
$g(t)e^{j2\pi f_c t}$	$\leftrightarrow$	$G(f - f_c)$	$\cos(2\pi f_c t)$	$\leftrightarrow$	$\frac{1}{2} \left( \delta(f+f_c) + \delta(f-f_c) \right)$
$\frac{d}{dt} g(t)$	$\leftrightarrow$	$j2\pi f \ G(f)$	$\sin(2\pi f_c t)$	$\leftrightarrow$	$\frac{j}{2} \left( \delta(f+f_c) - \delta(f-f_c) \right)$
$g^*(t)$	$\leftrightarrow$	$G^*(-f)$	$\alpha e^{-\pi \alpha^2 t^2}$	$\leftrightarrow$	$e^{-\pi f^2/\alpha^2}$

#### $\rightarrow$ full list in Appendix C of the compendium



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### Some useful Fourier transform properties

• Consider two signals x(t) and y(t) and their Fourier transforms

 $x(t) \longleftrightarrow X(f)$ ,  $y(t) \longleftrightarrow Y(f)$ 

• Recall the convolution operation z(t) = x(t) \* y(t):



Filtering:

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$$x(t) * y(t) \longleftrightarrow X(f) \cdot Y(f)$$

Multiplication:

$$x(t) \cdot y(t) \longleftrightarrow X(f) * Y(f)$$

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### Spectrum of time-limited signals

- Consider some time-limited signal  $s_T(t)$  of duration T, with  $s_T(t) = 0$  for t < 0 and t > T
- Assume that within the interval  $0 \le t \le T$ , the signal  $s_T(t)$  is equal to some signal s(t), i.e.,

$$s_T(t) = s(t) \cdot g_{rec}(t) ,$$

where  $g_{rec}(t)$  is the rectangular pulse of amplitude A = 1

Taking the Fourier transform on both sides we get

$$S_T(f) = S(f) * G_{rec}(f) = S(f) * AT \frac{\sin(\pi f T)}{\pi f T} e^{-j\pi f T}$$

• Since  $G_{rec}(f)$  is unlimited along the frequency axis, this is the case for  $S_T(f)$  as well (convolution increases length)

Time-limited signals can never be strictly band-limited



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Some definitions of bandwidth

Main-lobe definition:

 $W_{lobe}$  is defined by the width of the main-lobe of R(f)This is how we have defined bandwidth in previous examples

- In baseband we use the one-sided width, while in bandpass applications the two-sided width is used (positive frequencies)
- Percentage definition:

 $W_{99}$  is defined according to the location of 99% of the power

 $\blacktriangleright$  For bandpass signals  $W_{99}$  is found as the value that satisfies

$$\int_{f_c - W_{99}/2}^{f_c + W_{99}/2} R(f) \, df = 0.99 \, \int_0^\infty R(f) \, df$$

- Other percentages can be used as well:  $W_{90}$ ,  $W_{99,9}$
- Nyquist bandwidth

Assuming an ideal pulse with finite bandwidth (see Chapter 6)

$$W_{nyq} = \frac{R_s}{2} \, [\text{Hz}]$$



## Some definitions of bandwidth

Pulse shape	$W_{lobe}$	% power	$W_{90}$	$W_{99}$	$W_{99.9}$	Asymptotic
		in $W_{lobe}$				decay
rec	2/T	90.3	1.70/T	20.6/T	204/T	$f^{-2}$
tri	4/T	99.7	1.70/T	2.60/T	6.24/T	$f^{-4}$
hcs	3/T	99.5	1.56/T	2.36/T	5.48/T	$f^{-4}$
rc	4/T	99.95	1.90/T	2.82/T	3.46/T	$f^{-6}$
Nyquist	$R_s$	100	$0.9R_s$	$0.99R_s$	$0.999R_{s}$	ideal

Table 2.1: Double-sided bandwidth results for power spectral densities according to (2.212). The  $g_{rec}(t)$ ,  $g_{tri}(t)$ ,  $g_{hcs}(t)$  and  $g_{rc}(t)$  pulse shapes are defined in Appendix D, and T denotes the duration of the pulse. The Nyquist pulse shape is not limited in time and it is defined in (D.49) with parameters  $\beta = 0$  and  $\mathcal{T} = T_s.$ 

- This table is useful for PAM, PSK, and QAM constellations
- Except bandwidth W, the asymptotic decay is also relevant

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### From last lecture: general R(f)

• The power spectral density R(f) can be divided into a continuous part  $R_c(f)$  and a discrete part  $R_d(f)$ 

$$R(f) = R_c(f) + R_d(f)$$

The general expression for the continuous part is

$$R_{c}(f) = \frac{1}{T_{s}} \sum_{n=0}^{M-1} P_{n} |S_{n}(f) - A(f)|^{2}$$
$$= \left(\frac{1}{T_{s}} \sum_{n=0}^{M-1} P_{n} |S_{n}(f)|^{2}\right) - \frac{|A(f)|^{2}}{T_{s}}$$

For the discrete part we have

$$R_d(f) = \frac{|A(f)|^2}{T_s^2} \sum_{n=-\infty}^{\infty} \delta(f - n/T_s)$$



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### Example 2.28

Assume the bit rate  $R_b = 9600$  [bps], M-ary PAM transmission and that  $m_A = 0$ . Determine the (baseband) bandwidth W, defined as the one-sided width of the mainlobe of the power spectral density R(f), if M = 2, M = 4 and M = 8, respectively. Furthermore, assume a rectangular pulse shape with amplitude  $A_{q}$ , and duration  $T = T_{s}$ . Calculate also the bandwidth efficiency  $\rho$ .

- ▶ What is W for a given pulse shape and different M?
- Using  $T = T_s$ ,  $m_A = 0$  and  $g(t) = g_{rec}(t)$ , we have

$$R(f) = \frac{\sigma_A^2}{T_s} |G_{rec}(f)|^2$$

For the given pulse we get  $W = 1/T_s$ , where  $T_s = k T_h$ 

 $\Rightarrow M = 2 \Rightarrow W = 9600$ [Hz]  $k = 2 \Rightarrow M = 4 \Rightarrow W = 4800$ [Hz]  $k = 3 \Rightarrow M = 8 \Rightarrow W = 3200$ [Hz]

• Bandwidth efficiency:  $\rho = R_b/W = k T_b/T_b = k$ 



### R(f): *M*-ary PAM signals

► With *M*-ary PAM signaling we have

$$s_{\ell} = A_{\ell} g(t) , \quad \ell = 0, 1, \dots, M-1$$

Then

$$S_{\ell}(f) = A_{\ell} G(f)$$
, and  $A(f) = \sum_{\ell=0}^{M-1} P_{\ell} A_{\ell} G(f)$ 

With this we obtain the simplified expression

$$R(f) = \frac{\sigma_A^2}{T_s} |G(f)|^2 + \frac{m_A^2}{T_s^2} |G(f)|^2 \sum_{n = -\infty}^{\infty} \delta(f - n/T_s)$$

where  $m_A$  denotes the mean and  $\sigma_A^2 = \overline{E}_s / E_g - m_A^2$  the variance of the amplitudes  $A_{\ell}$ 

• Assuming zero average amplitude  $m_A = 0$  and using  $\overline{P} = \sigma_A^2 E_g R_s$ this reduces to

$$R(f) = R_c(f) = \frac{\sigma_A^2}{T_s} |G(f)|^2 = \frac{\overline{P}}{E_g} |G(f)|^2$$

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### What does bandwidth efficiency tell us?

In the previous example we had a bandwidth efficiency of

$$\rho = \frac{R_b}{W} = k$$

#### Saving bandwidth

- ▶ The previous example showed that the bandwidth W can be reduced by increasing M
- $\blacktriangleright$   $T = T_s = kT_h$  increases with M
- $\blacktriangleright$   $W = 1/T = R_h/k$  decreases accordingly

#### Improving bit rate

- Assume instead that the bandwidth W is fixed in the same example, i.e., the symbol duration  $T_s = T$  is fixed
- Then  $R_b = k W$  increases with M
- Assume for example W = 1 MHz:
  - $R_b = 1$  Mbps if M = 2 (k = 1)
  - $R_b = 10 \text{ Mbps if } M = 1024 \ (k = 10)$



## R(f): *M*-ary **QAM** signals

With M-ary QAM signaling the signal alternatives are

$$s_{\ell}(t) = A_{\ell} g(t) \cos(2\pi f_c t) - B_{\ell} g(t) \sin(2\pi f_c t), \quad \ell = 0, 1, \dots, M-1$$

Then the Fourier transform becomes

$$S_{\ell}(f) = A_{\ell} \frac{G(f+f_c) + G(f-f_c)}{2} - j B_{\ell} \frac{G(f+f_c) - G(f-f_c)}{2}$$
$$= (A_{\ell} - jB_{\ell}) \frac{G(f+f_c)}{2} + (A_{\ell} + jB_{\ell}) \frac{G(f-f_c)}{2}$$

• Assuming a zero average signal a(t) = 0 and  $f_c T > 1$  this simplifies to

$$R(f) = R_c(f) = \overline{P} \frac{|G(f+f_c)|^2 + |G(f-f_c)|^2}{2E_g}$$



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## Example

Bandwidth consumption for BPSK, QPSK and 16-QAM assuming equal  $R_b$  and  $f_c = 100 R_b$ 



Figure 2.20: The power spectral density for binary QAM (BPSK, widest mainlobe), 4-ary QAM (QPSK), and 16-ary QAM (smallest mainlobe). The figure shows  $10 \log_{10}(R_b R(f)/\bar{P})$  [dB] in the frequency interval  $98R_b \leq f \leq 102R_b$ . The carrier frequency is  $f_c = 100R_b$  [Hz], and a  $T_s = kT_b \log g_{hcs}(t)$  pulse is assumed. See also (2.227) and (2.230).



## R(f): *M*-ary **QAM** signals

▶ Remember that *M*-ary QAM signals contain *M*-ary PSK and *M*-ary bandpass PAM signals as special cases:

> **BP-PAM:**  $B_{\ell} = 0$ **PSK:**  $A_{\ell} = \cos(v_{\ell})$ ,  $B_{\ell} = \sin(v_{\ell})$

- ▶  $\Rightarrow$  our results for R(f) of *M*-ary QAM signals include these cases
- For symmetric constellations, such that a(t) = 0, the simplified version applies
- The bandwidth *W* is determined by  $|G(f-f_c)|^2$  and hence the two-sided main-lobe of  $|G(f)|^2$

 $\Rightarrow$  if the same pulse g(t) is used then *M*-ary QAM, *M*-ary bandpass PAM and *M*-ary PSK have the same bandwidth *W* 



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## R(f): *M*-ary FSK signals

▶ With *M*-ary frequency shift keying (FSK) signaling the signal alternatives are

$$s_\ell(t) = A \cos(2\pi f_\ell t + \mathbf{v}), \quad 0 \le t \le T_s$$

• Choosing  $v = -\pi/2$  this can be written as

 $s_{\ell}(t) = g_{rec}(t) \sin(2\pi f_{\ell} t)$ , with  $T = T_s$ ,

since  $s_{\ell}(t) = 0$  outside the symbol interval

► The Fourier transform is then

$$S_{\ell}(f) = j \frac{G_{rec}(f+f_c) - G_{rec}(f-f_c)}{2}$$

• The exact power spectral density R(f) can now be computed by the general formula (2.202)-(2.204)



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## R(f): *M*-ary FSK signals

- Let us find an approximate expression for the FSK bandwidth W
- Assume that

 $f_{\ell} = f_0 + \ell f_{\Delta}, \quad \ell = 0, \dots, M - 1$ 

▶ Then the bandwidth *W* can be approximated by

$$W \approx R_s + f_{M-1} - f_0 + R_s = (M-1)f_{\Delta} + 2R_s$$

- Consider now orthogonal FSK with  $f_{\Delta} = I \cdot R_s/2$  for some I > 0
- ► The bandwidth efficiency is then

$a - \frac{R_b}{\sim} \sim$	$R_b$	$\_$ $R_b$	_	$\log_2 M$
$p - \frac{1}{W} \sim$	$\overline{(M-1)f_{\Delta}+2R_s}$	$-\frac{((M-1)I/2+2)R_s}{((M-1)I/2+2)R_s}$	_	(M-1)I/2+2

#### **Observe:** the bandwidth efficiency of orthogonal M-ary FSK gets

small if *M* is large

Last week we saw: *M*-ary FSK has good energy and Euclidean distance properties  $\Rightarrow$  trade-off



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## *R*(*f*): **OFDM-type signals**

An OFDM symbol (signal alternative) x(t) can be modeled as a superposition of N orthogonal QAM signals, each carrying k<sub>n</sub> bits, that are transmitted at different frequencies (sub-carriers)

$$x(t) = \sum_{n=0}^{N-1} s_{n,QAM}(t)$$

Assuming each QAM signal has zero mean and that the different carriers have independent bit streams we get

$$R(f) = R_c(f) = R_s E\{|X(f)|^2\} = \sum_{n=0}^{N-1} R_n(f)$$

Using our previous results for QAM in each sub-carrier we get

$$R(f) = R_c(f) = \sum_{n=0}^{N-1} \overline{P} \; \frac{|G(f+f_c)|^2 + |G(f-f_c)|^2}{2E_g}$$



### Example 2.36

Assume that orthogonal M-ary FSK is used to communicate digital information in the frequency band 1.1  $\leq f \leq$  1.2 [MHz].

For each M below, find the largest bit rate that can be used (use bandwidth approximations):

 $i) \ M = 2 \qquad ii) \ M = 4 \qquad iii) \ M = 8 \qquad iv) \ M = 16 \qquad v) \ M = 32$ 

Which of the M-values above give a higher bit rate than the  $M=2\ case?$ 

Solution:

It is given that  $W_{M-FSK} = 100$  [kHz]. From (2.245), the largest bit rate is obtained with I = 1:

$$R_b \approx 10^5 \cdot \frac{\log_2(M)}{(M-1)/2 + 2}$$



From this table it is seen that M = 4, 8, 16 give a higher bit rate than M = 2.

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## R(f): OFDM-type signals

Illustration of  $R_n(f)$  contributed by three neighboring sub-carriers:



• Assuming  $f_n = f_0 + n/(T_s - \Delta_h)$  we can estimate the bandwidth as

$$W \approx (N+1) f_\Delta = rac{N+1}{1 - \Delta_h/T_s} R_s \approx N \cdot R_s \;, \quad N \gg 1 \;, \; \Delta_h \ll T_s$$

The bandwidth efficiency is then approximated by

$$ho = rac{R_b}{W} = rac{R_s}{W} \sum_{k=0}^{N-1} k_n pprox rac{1}{N} \sum_{k=0}^{N-1} k_n \; [ ext{bps/Hz}]$$



## **Example:** R(f) for OFDM



- ► N = 16 sub-carriers
- ►  $T = T_s = 0.1 \text{ [ms]}$
- $f_{\Delta} = R_s / 0.95 = 10.53 \, [\text{kHz}]$

• 
$$W \approx \frac{17}{0.95} R_s = 179 \text{ [kHz]}$$



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## What about filtering away the side-lobes?

- ► Let us use a spectral rectangular pulse  $X_{srec}(f)$  of amplitude A = 1and width  $f_{\Delta}$  to strictly limit the bandwidth
- Similar to the time-limited case we can write

$$S_{f_{\Delta}}(f) = S(f) \cdot X_{srec}(f)$$

► Taking the inverse Fourier transform on both sides we get

$$s_{f_{\Delta}}(t) = s(t) * x_{srec}(t) = s(t) * Af_0 \frac{\sin(\pi f_0 t)}{\pi f_0 t}$$

► Since x<sub>srec</sub>(t) is unlimited along the time axis, this is the case for the filtered signal s<sub>f<sub>λ</sub></sub>(t) as well

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• The signal  $x_{srec}(t)$  defines the ideal Nyquist pulse

As a consequence of filtering, the transmitted symbols will overlap in time domain  $\Rightarrow$  inter-symbol-interference (ISI)



## Example 2.35

#### ADSL: uses plain telephone cable (twisted pair, copper)



In ADSL, a coded OFDM technique is used. The level of the power spectral density in the downstream is roughly -73 dB. As a basic example, let us here assume that the OFDM symbol rate in the downlink is 4000 [symbol/s], and that the subchannel carrier spacing is 5 kHz. Furthermore, it is here also assumed that uncoded 16-ary QAM is used in each subchannel (assumes a very "good" communication link).

For the ADSL downlink above, determine the bit rate in each subchannel, the total bit rate, and the bandwidth efficiency.



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# Nyquist Pulse



Figure 6.6: a) Ideal Nyquist spectrum; b) Ideal Nyquist pulse.

$$\begin{aligned} x_{nc}(t) &= x_0 \frac{\sin(\pi R_{nyq} t)}{\pi R_{nyq} t} , -\infty \le t \le \infty \end{aligned} (6.39) \\ X_{nc}(f) &= \begin{cases} x_0/R_{nyq} , & |f| \le R_{nyq}/2 \\ 0 , & |f| > R_{nyq}/2 \end{cases} (6.40)$$

#### The Nyquist pulse and the effect of ISI will be studied in Chapter 6



### How can we further improve $\rho$ ?



- MIMO: multiple-input multiple output
- ▶ transmission over multiple antennas in the same frequency band
- challenge: the individual wireless channels interfere
- 5G world record 2016: (team from Lund involved) spectral efficiency of 145.6 bps/Hz with 128 antennas

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### **Example: discrete frequencies in** R(f)

- Assume M = 2
- Let  $s_0(t) = 0$  and  $s_1(t) = 5$  with a pulse duration  $T = T_b/2$
- With this the average signal becomes

$$a(t) = \frac{s_0(t) + s_1(t)}{2} = 2.5$$
,  $0 \le t \le T$ 

▶ We can then write (within the pulse duration *T*)

$$s_0(t) = -2.5 + a(t)$$
,  $s_1(t) = +2.5 + a(t)$ 

#### Observe:

- **1.** this method is a waste of signal energy since a(t) does not carry any information
- repetition of *a*(*t*) in every symbol interval creates some periodic signal component in the time domain, which leads to discrete frequencies in the frequency domain

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