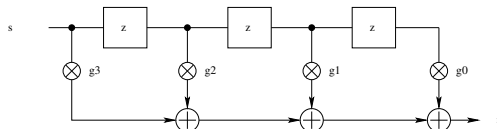


# OFDM and DMT: System Model

EIT 140, tom<AT>eit.lth.se

# Review: LTI channel and dispersion

- Time-dispersive LTI channel



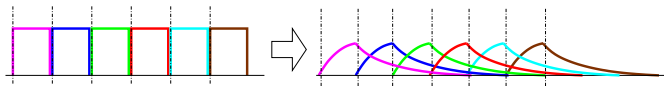
- Dispersion**  $M$  = length of impulse response  $- 1$  = order of channel filter
- LTI channel performs linear convolution

$$r(n) = \sum_k s(k)h(n - k)$$

- Length- $N$  input  $s(n), n = 0, \dots, N - 1$  yields length- $(N + M)$  output  $r(n), n = 0, \dots, N + M - 1$
- Input block is **"smeared out" (dispersed)** over  $M$  additional samples

# Review: Dispersion can cause unacceptable ISI/ICI

- High datarate  $\rightarrow$  high symbolrate  $\rightarrow$  severe ISI

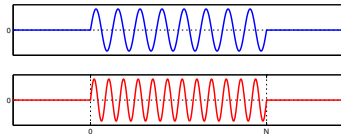


high data rate

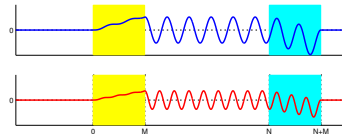
**severe ISI!**

- Dispersion causes transients: severe ICI  $\longleftrightarrow$  loss of orthogonality

*channel input*

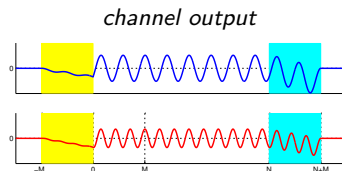
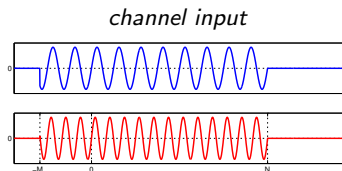


*channel output*



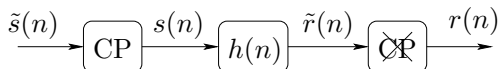
# Review: Ideas of multicarrier transmission

- Use several, mutually orthogonal, longer, sinusoidal transmit waveforms
  - “several”: compensate for the “longer” in terms of data rate
  - “mutually orthogonal”: avoid ICI
  - “sinusoidal”: **eigenfunctions of LTI systems**, efficient implementation (FFT)
- Cyclic extension avoids ISI and ICI in time-dispersive channels



- System design limits
  - $T_{MC} \gg \tau_{\max}$
  - $T_{MC} \ll \min\{T_{\text{coh}}, \text{latency limit, carrier-spacing limit}\}$

## CP: linear convolution $\longrightarrow$ circular convolution



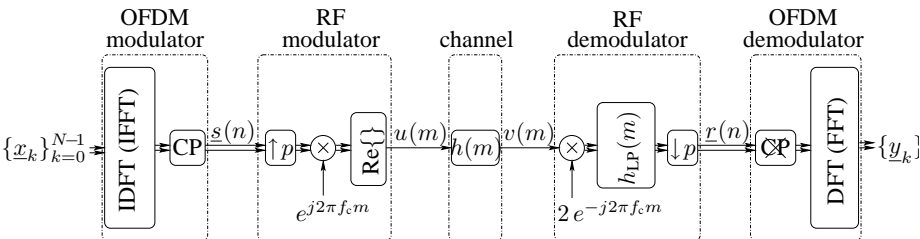
- CP-add block:  $s(n) = \begin{cases} \tilde{s}(n), & n = 0, \dots, N-1 \\ \tilde{s}(N+n), & n = -L, \dots, -1 \end{cases}$
- Channel: **linear convolution**  $\tilde{r}(n) = \sum_k s(k)h(n-k)$
- CP-remove block:  $r(n) = \tilde{r}(n), n = 0, \dots, N-1$

For  $L \geq M$ : CP-add + channel + CP-remove yield **circular convolution**

$$r(n) = \tilde{r}(n) = \sum_k \tilde{s}((n-k) \bmod N) h(k), \quad n = 0, \dots, N-1$$

and consequently  $R[k] = \tilde{S}[k]H[k], k = 0, \dots, N-1$  holds.

# Orthogonal Frequency Division Multiplex (OFDM) system



$$\underline{s}(n) = \text{IDFT}_N \{\underline{x}_k\}_{k=0}^{N-1}$$

$$\underline{s}(n) = \underline{s}(n+N),$$

$$\underline{s}(m) = \{\underline{s}(n)\}_{\uparrow p},$$

$$u(m) = \text{Re} \left\{ \underline{s}(m) e^{j2\pi f_c m} \right\}$$

$$v(m) = u(m) * h(m)$$

$$\underline{r}(m) = \left( v(m) 2e^{-j2\pi f_c m} \right) * h_{LP}(m)$$

$$\underline{r}(n) = \{\underline{r}(m)\}_{\downarrow p},$$

$$\underline{y}_k = \text{DFT}_N \{\underline{r}(n)\}_{n=0}^{N-1}$$

$$n = 0, \dots, N-1$$

$$n = -L, \dots, -1$$

$$m = -Lp, \dots, Np-1$$

$$n = 0, \dots, N-1$$

$$k = 0, \dots, N-1$$

# Receive signal components (passband)

Real-valued passband receive signal components<sup>1</sup>:

$$\begin{aligned}v_k^{(i)}(m) &= \frac{1}{\sqrt{N}} \cos\left(2\pi \left(f_c + \frac{k}{Np}\right) m\right), \\v_k^{(q)}(m) &= -\frac{1}{\sqrt{N}} \sin\left(2\pi \left(f_c + \frac{k}{Np}\right) m\right),\end{aligned}\quad m = 0, \dots, Np - 1$$

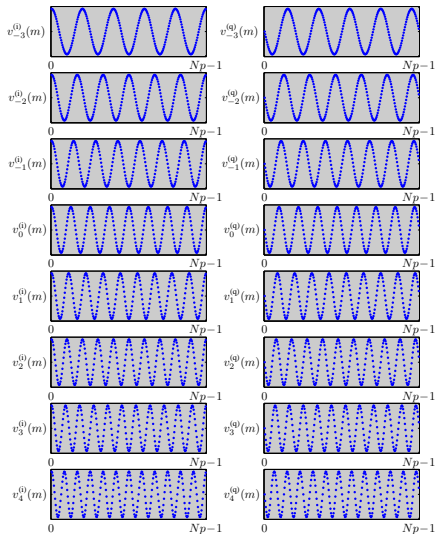
for  $k = -N/2 + 1, \dots, N/2$ .

- $f_c$  is a positive *carrier frequency*
- $p$  is the *oversampling factor* and sufficiently large to ensure that  $f_c + \frac{k}{pN} < \frac{1}{2}$ ,  $k = -N/2 + 1, \dots, N/2$  holds for all the  $N$  normalised frequencies

---

<sup>1</sup>Throughout the lecture, we assume that  $N$  is even.

# Receive signal components (passband) in time-domain



- $v_k^{(i)}(m)$  (cosine signals, left column) and  $v_k^{(q)}(m)$  (sine signals, right column); parameters:  $N = 8, p = 32, f_c = 1/32, k = -3, \dots, 4$
- There are  $2N$  mutually orthogonal real-valued length- $pN$  discrete-time sinusoidal waveforms with an integer number of periods



# Transmit signal components (passband)

Similarly, we consider the transmit signals of length  $(N + L)p$  samples

$$\begin{aligned}u_k^{(i)}(m) &= \frac{1}{\sqrt{N}} \cos\left(2\pi \left(f_c + \frac{k}{Np}\right) m\right), \\u_k^{(q)}(m) &= -\frac{1}{\sqrt{N}} \sin\left(2\pi \left(f_c + \frac{k}{Np}\right) m\right),\end{aligned}\quad m = -Lp, \dots, Np - 1$$

where  $k = -N/2 + 1, \dots, N/2$ .

- Cyclic prefix of length  $pL$  (passband)

# RF modulator

We write the passband transmit signal as

$$u(m) = \frac{1}{\sqrt{N}} \sum_{k=-N/2+1}^{N/2} \left( x_{[k]_N}^{(i)} \cos(2\pi \left( f_c + \frac{k}{Np} \right) m) - x_{[k]_N}^{(q)} \sin(2\pi \left( f_c + \frac{k}{Np} \right) m) \right)$$
$$= \operatorname{Re} \left\{ \underbrace{\frac{1}{\sqrt{N}} \sum_{k=-N/2+1}^{N/2} \underline{x}_{[k]_N} e^{j2\pi \frac{k}{Np} m}}_{\underline{s}(m)} e^{j2\pi f_c m} \right\},$$

where  $[k]_N$  denotes the modulo- $N$  operation applied to  $k$ .

- This operation corresponds to a cos/sin modulator structure
- Perfect interpolation by factor  $p$  of  $\underline{s}(n)$  yields  $\underline{s}(m)$

- Note that  $\underline{s}(n) = \frac{1}{\sqrt{N}} \sum_{k=-N/2+1}^{N/2} \underline{x}_{[k]_N} e^{j2\pi \frac{k}{N} n} = \frac{1}{\sqrt{N}} \sum_{k=0}^{N-1} \underline{x}_k e^{j2\pi \frac{k}{N} n}$

# Baseband transmit signal processing (OFDM modulator)

$$\begin{aligned}\underline{s}(n) &= \frac{1}{\sqrt{N}} \sum_{k=-N/2+1}^{N/2} \underline{x}_{[k]_N} e^{j2\pi \frac{k}{N} n} \\&= \frac{1}{\sqrt{N}} \left( \sum_{k=0}^{N/2} \underline{x}_k e^{j2\pi \frac{k}{N} n} + \sum_{k=-N/2+1}^{-1} \underline{x}_{\underbrace{(N+k)}_{\triangleq i, \rightarrow k=i-N}} e^{j2\pi \frac{k}{N} n} \right) \\&= \frac{1}{\sqrt{N}} \left( \sum_{k=0}^{N/2} \underline{x}_k e^{j2\pi \frac{k}{N} n} + \sum_{i=N/2+1}^{N-1} \underline{x}_i \underbrace{e^{j2\pi \frac{i-N}{N} n}}_{e^{j2\pi \frac{i}{N} n} e^{-j2\pi \frac{N}{N} n} = e^{j2\pi \frac{i}{N} n}} \right) \\&= \frac{1}{\sqrt{N}} \left( \sum_{k=0}^{N/2} \underline{x}_k e^{j2\pi \frac{k}{N} n} + \sum_{k=N/2+1}^{N-1} \underline{x}_k e^{j2\pi \frac{k}{N} n} \right) \\&= \frac{1}{\sqrt{N}} \sum_{k=0}^{N-1} \underline{x}_k e^{j2\pi \frac{k}{N} n}, \quad \text{which is the IDFT of } \underline{x}_k\end{aligned}$$

# Channel output

The channel coefficients are given by

$$\underline{H}_{[k]_N} = A_{[k]_N} + jB_{[k]_N} = \sum_{m=0}^{(M+1)p-1} h(m) e^{-j2\pi(f_c + \frac{k}{pN})m}, \quad k = -N/2 + 1, \dots, N/2.$$

The passband transmit signal  $u(m)$  yields the channel output

$$v(m) = \frac{1}{\sqrt{N}} \sum_{k=-N/2+1}^{N/2} \left( (A_{[k]_N} x_{[k]_N}^{(i)} - B_{[k]_N} x_{[k]_N}^{(q)}) \cos\left(2\pi \left(f_c + \frac{k}{pN}\right) m\right) - \right. \\ \left. (B_{[k]_N} x_{[k]_N}^{(i)} + A_{[k]_N} x_{[k]_N}^{(q)}) \sin\left(2\pi \left(f_c + \frac{k}{pN}\right) m\right) \right)$$

# RF demodulator

Receiver performs the **cosine-sine demodulation**, **low pass filtering** and **downsampling** by factor  $p$  of the receive signal, which yields

$$\underline{r}(n) = \frac{1}{\sqrt{N}} \sum_{k=-\frac{N}{2}+1}^{\frac{N}{2}} \left( (A_{[k]_N} x_{[k]_N}^{(i)} - B_{[k]_N} x_{[k]_N}^{(q)}) + j(B_{[k]_N} x_{[k]_N}^{(i)} + A_{[k]_N} x_{[k]_N}^{(q)}) \right) e^{j2\pi \frac{k}{N} n},$$

which can be written as

$$\underline{r}(n) = \frac{1}{\sqrt{N}} \sum_{k=0}^{N-1} \left( (A_k x_k^{(i)} - B_k x_k^{(q)}) + j(B_k x_k^{(i)} + A_k x_k^{(q)}) \right) e^{j2\pi \frac{k}{N} n} \quad (1)$$

We can interpret (1) as receive multiplex using the receive signal components

$$\underline{r}_k(n) = \frac{1}{\sqrt{N}} e^{j2\pi \frac{kn}{N}}, \quad n = 0, \dots, N-1$$

# Baseband receive signal processing (OFDM demodulator)

Finally, we pass the complex-valued receive multiplex

$$\underline{r}(n) = \frac{1}{\sqrt{N}} \sum_{k=0}^{N-1} \left( (A_k x_k^{(i)} - B_k x_k^{(q)}) + j(B_k x_k^{(i)} + A_k x_k^{(q)}) \right) e^{+j2\pi \frac{k}{N} n}$$

through a bank of correlators, which yields

$$\langle \underline{r}, \underline{r}_k \rangle = \frac{1}{\sqrt{N}} \sum_{n=0}^{N-1} \underline{r}(n) e^{-j2\pi \frac{k}{N} n} = (A_k x_k^{(i)} - B_k x_k^{(q)}) + j(B_k x_k^{(i)} + A_k x_k^{(q)})$$

where  $k = 0, \dots, N-1$ . The operation performed by these correlators is equivalent to the scaled  $N$ -point **Discrete Fourier Transform (DFT)** of the complex-valued baseband receive multiplex  $\underline{r}(n)$

$$\underline{y}_k = \frac{1}{\sqrt{N}} \sum_{n=0}^{N-1} \underline{r}(n) e^{-j2\pi \frac{k}{N} n}.$$

# Complex baseband representation

We now introduce a model of a fictitious system which has the same *behaviour* as the real-world, implementable passband system above. The received *complex* signal can be written as

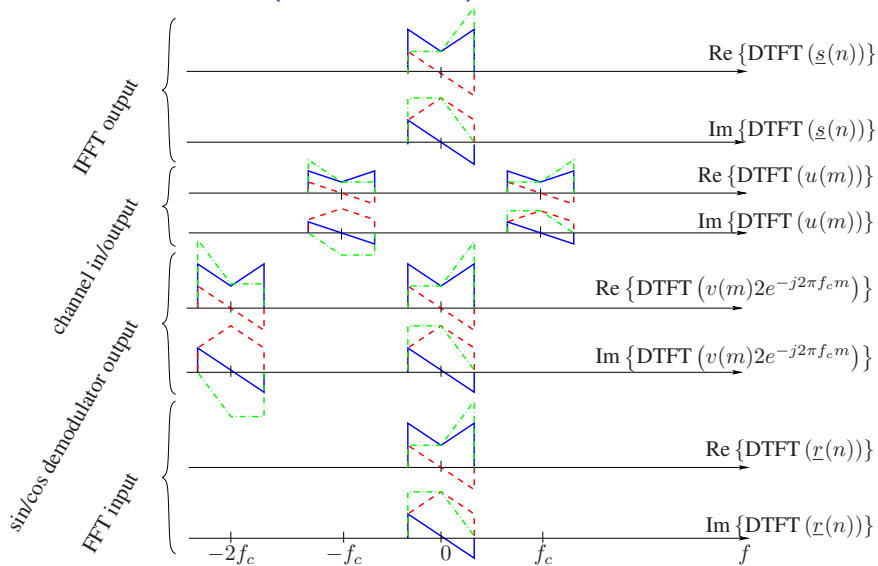
$$\underline{r}(n) = \underline{h}(n) * \underline{s}(n),$$

where the *complex* channel impulse response  $\underline{h}(n)$  is given by

$$\underline{h}(n) = \text{IDFT}_N \{ \underline{H}_k \} = \sum_{k=0}^{N-1} \underline{H}_k e^{j2\pi \frac{kn}{N}}.$$

- We model the (real-valued) passband signals as (complex-valued) baseband signals
- In essence, we omit RF modulator and RF demodulator and transform the passband channel into an equivalent baseband channel

# OFDM spectra (flat channel)



solid blue: DTFT of  $\text{Re}\{x\}$ . dashed red: DTFT of  $\text{Im}\{x\}$ . dashed-dotted green: DTFT of  $x = \text{Re}\{x\} + j\text{Im}\{x\}$ .



# Power spectral density (PSD)

- Average measure for spectral allocation of transmit power
- Stationary signals:  $\mathcal{PSD}(f) = \mathcal{F}_m(r(m))$ 
  - PSD is Fourier transform of transmit signal's autocorrelation function  $r(m)$
- Cyclostationary signals:  $\mathcal{PSD}(f) = \mathcal{F}_m\left(\frac{1}{N'} \sum_n r(m, n)\right)$ 
  - PSD is Fourier transform of time-averaged transmit signal's autocorrelation function  $r(m, n)$ . ( $r(m, n) = r(m, n + N')$ )
- For uncorrelated data (over both time and subchannels):

$$\mathcal{PSD}(f) = \sum_{k \in \text{set of used subchannels}} P_k |\text{DTFT of } k\text{th basis function}|^2$$

where  $P_k$  is the power on the  $k$ th subchannel.

- cf. `psdmc.m`

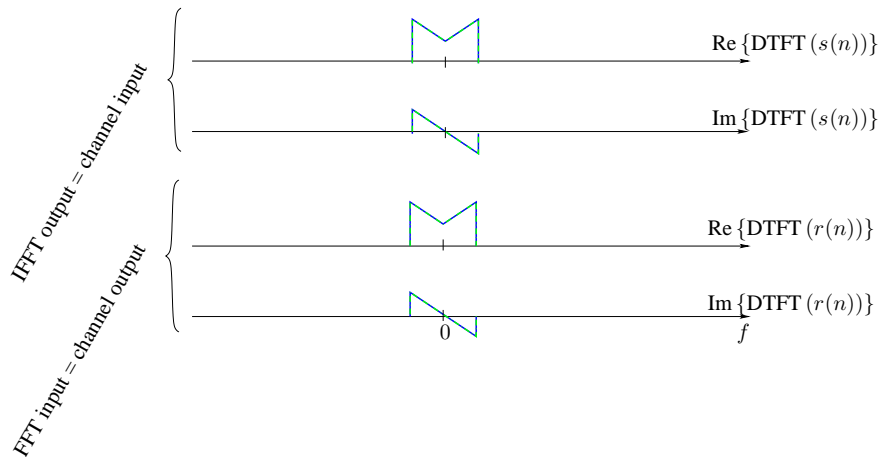
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$\mathcal{F}_\cdot(r(\cdot, \dots))$  is the Fourier transform (DTFT) of the continuous(discrete)-time signal  $r$  with respect to  $\cdot$

# Discrete Multi-Tone (DMT) spectra (flat channel)

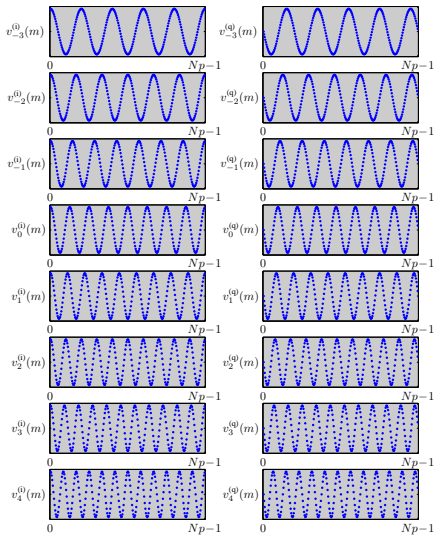
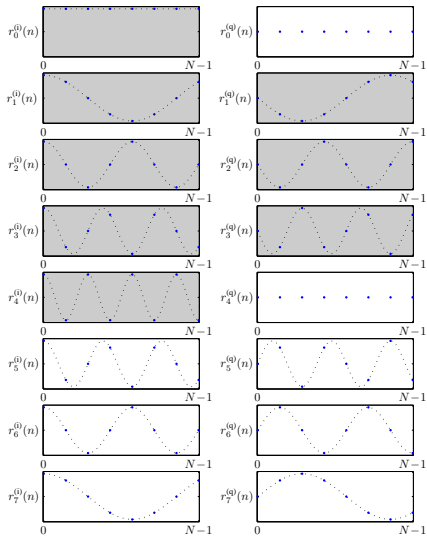
In case we have a baseband channel, we

- do not need the RF modulator/demodulator
- need a real-valued transmit multiplex

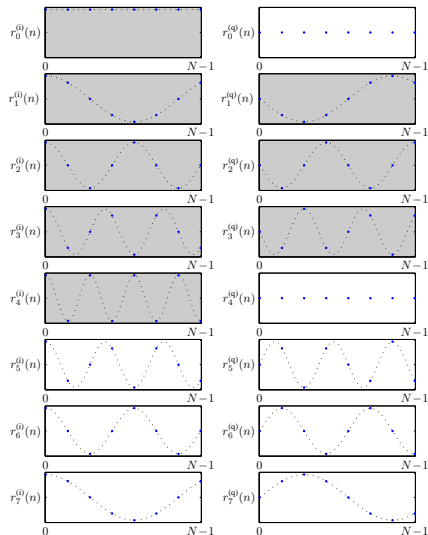


solid blue:  $\text{DTFT of } \text{Re}\{x\}$ . dashed red:  $\text{DTFT of } \text{Im}\{x\}$ . dashed-dotted green:  $\text{DTFT of } x = \text{Re}\{x\} + j\text{Im}\{x\}$ .

# Receive signal components: baseband $\leftrightarrow$ passband



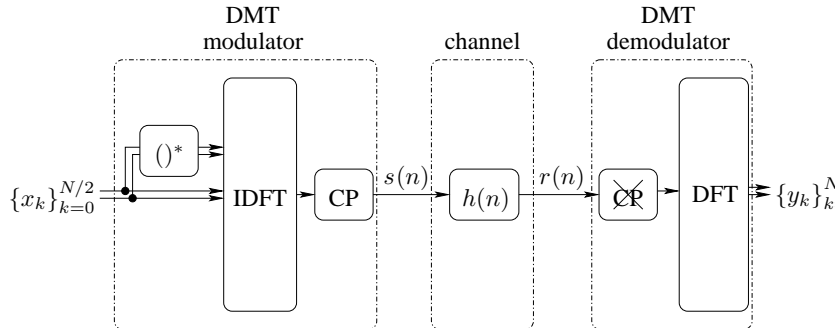
# Sinusoidal receive signal components



- There are  $N$  mutually orthogonal real-valued length- $N$  discrete-time sinusoidal waveforms with an integer number of periods ( $N = 8$  in this example).
- For  $k \in \{\lfloor N/2 \rfloor + 1, \dots, N-1\}$ , the number of oscillations per  $N$  samples decreases with increasing  $k$ .
- In fact, it is easy to verify that

$$\begin{aligned}\cos(2\pi \frac{k}{N} n) &= \cos(2\pi \frac{(N-k)}{N} n), \\ \sin(2\pi \frac{k}{N} n) &= -\sin(2\pi \frac{(N-k)}{N} n)\end{aligned}\quad (2)$$

# Discrete Multi-Tone (DMT) system



$$\begin{aligned} \underline{x}_{N-k} &= \underline{x}_k^* & k &= 0, \dots, N/2 \\ s(n) &= \text{IDFT}_N \{ \underline{x}_k \}_{k=0}^{N-1} & n &= 0, \dots, N-1 \\ s(n) &= s(n+N), & n &= -L, \dots, -1 \\ r(n) &= h(n) * s(n), & n &= 0, \dots, N-1 \\ \underline{y}_k &= \text{DFT}_N \{ r(n) \}_{n=0}^{N-1} & k &= 0, \dots, N/2 \end{aligned}$$

# DMT: Hermitian symmetry of transmit symbols

We make the following choice for the transmit symbols

$$x_k^{(i)} = x_{N-k}^{(i)} \quad \text{and} \quad x_k^{(q)} = -x_{N-k}^{(q)}, \quad (3)$$

which implies  $x_0^{(q)} = x_{N/2}^{(q)} = 0$ . (3) ensures Hermitian symmetry of the transmit symbol blocks. With (2), we obtain for  $n = -L, \dots, N-1$

$$\begin{aligned} s(n) = \frac{1}{\sqrt{N}} \sum_{k=0}^{N-1} \left( x_k^{(i)} \cos(2\pi \frac{k}{N} n) - x_k^{(q)} \sin(2\pi \frac{k}{N} n) \right) \\ + j \frac{1}{\sqrt{N}} \underbrace{\sum_{k=0}^{N-1} \left( x_k^{(i)} \sin(2\pi \frac{k}{N} n) + x_k^{(q)} \cos(2\pi \frac{k}{N} n) \right)}_{=0} = \frac{1}{\sqrt{N}} \sum_{k=0}^{N-1} \underline{x}_k e^{j2\pi \frac{k}{N} n}. \end{aligned}$$

- On the interval  $0, \dots, N-1$ , the transmit signal multiplex  $s(n)$  is generated by the scaled  $N$ -point **Inverse Discrete Fourier Transform (IDFT)** of the data symbols  $\underline{x}_k$ .

# DMT and OFDM

- DMT: **real-valued** baseband multiplex  $s(n)$ , OFDM: **complex-valued** baseband multiplex  $\underline{s}(n)$
- DMT: **wireline** communications (**lowpass** channel), OFDM: **wireless** communications (**passband** channel)
- DMT: channel **known** at the transmitter, OFDM: channel **unknown** at transmitter

# Examples of OFDM/DMT systems and parameters

	HiperLAN/2	DVB-T 8k	ADSL DS	VDSL DS	DAB
multiplex <sup>2</sup> :	complex	complex	real	real	complex
$N$ :	64	8192	512	8192	2048
$L$ :	16	256	32 or <sup>3</sup> 40	640	504
bandwidth:	20 MHz	7.61 MHz	966 kHz	$\approx 8$ MHz	1.536 MHz
band:	5.15-5.35 GHz	UHF	0-1.1 MHz	0-12 MHz	III
$\frac{F_s}{N}$ :	0.3125 MHz	1.116 kHz	4.3125 kHz	4.3125 kHz	1 kHz
datarate:	6-54 Mbps	5-30 Mbps	0.5-8 Mbps	$\leq 54$ Mbps	$\leq 348$ kbps
modulation:	B/Q/8PSK 16/64QAM	4/16/64QAM	BPSK 4-2 <sup>15</sup> QAM	BPSK 4-2 <sup>15</sup> QAM	diff. QPSK
mobility:	pedestrian	vehicular	none	none	vehicular
channel:	wireless	wireless	wire	wire	wireless
reach:	$\leq 150$ m	$\approx$ km	$\leq 5$ km	$\leq 1.5$ km	$\approx$ km

III band: 174MHz - 240MHz; UHF band: 790MHz - 806MHz

Note that many standards include various modes with different parameter sets.

<sup>2</sup>complex  $\triangleq$  OFDM. real  $\triangleq$  DMT.

<sup>3</sup> $L = 32$  in case a synch-symbol is used, otherwise  $L = 40$



# Matrix representation of DMT and OFDM

- Comes in handy since OFDM/DMT is a block transmission scheme
- More compact and less painful than scalar description
- Yields another interpretation of cyclic prefixing, ISI-free and ICI-free transmission
- Notation:
  - lowercase boldface: (usually a column<sup>4</sup>) vector (e.g.  $\mathbf{a} \in \mathbb{R}^N$ )
  - uppercase boldface: matrix (e.g.  $\mathbf{A} \in \mathbb{C}^{N \times (N+L)}$ )
  - $\mathbf{A}^T$ : transpose.  $\mathbf{A}^H$ : Hermitian transpose (transposed conjugate).

---

<sup>4</sup>In channel coding, they like to use  $\mathbf{a}$  as row vector.

# Channel: linear convolution matrix

We describe the channel dispersion using a convolution matrix  $\mathbf{H}$ , where we assume  $L = M$ :

$$\begin{bmatrix} r(0) \\ r(1) \\ \vdots \\ r(N-1) \end{bmatrix} = \begin{bmatrix} \color{red}{h(L)} & \cdots & \color{red}{h(1)} & h(0) & 0 & \cdots & \cdots & \cdots & \cdots & 0 \\ 0 & \ddots & & h(1) & h(0) & \ddots & & & & \vdots \\ \vdots & & \color{red}{h(L)} & \vdots & h(1) & \ddots & & & & \vdots \\ \vdots & & & h(L) & \vdots & \ddots & \ddots & \ddots & & \vdots \\ \vdots & & & 0 & h(L) & & \ddots & \ddots & \ddots & \vdots \\ \vdots & & & \vdots & \ddots & \ddots & & \ddots & \ddots & 0 \\ 0 & \cdots & \cdots & 0 & \cdots & 0 & h(L) & \cdots & h(1) & h(0) \end{bmatrix} \begin{bmatrix} \color{red}{s(-L)} \\ \vdots \\ \color{red}{s(-1)} \\ s(0) \\ s(1) \\ \vdots \\ s(N-1) \end{bmatrix} \quad (4)$$

- Both DMT ( $\mathbf{H}$  **real-valued**) or OFDM ( $\mathbf{H}$  **complex-valued**) can be represented
- The first  $L$  received samples  $r(n)$ ,  $n = -L, \dots, -1$  are implicitly ignored

## Channel + CP: circulant convolution matrix

Exploiting the fact that  $s(n) = s(N - n)$ ,  $n = -L, \dots, -1$ , we can rewrite (4) as

$$\underbrace{\begin{bmatrix} r(0) \\ r(1) \\ \vdots \\ r(N-1) \end{bmatrix}}_{\mathbf{r}} = \underbrace{\begin{bmatrix} h(0) & 0 & \cdots & 0 & h(L) & \cdots & h(1) \\ h(1) & h(0) & \ddots & & 0 & \ddots & \vdots \\ \vdots & h(1) & \ddots & & \ddots & \ddots & h(L) \\ h(L) & \vdots & \ddots & \ddots & & & 0 \\ 0 & h(L) & & \ddots & \ddots & & \vdots \\ \vdots & \ddots & \ddots & \ddots & \ddots & \ddots & \vdots \\ 0 & \cdots & 0 & h(L) & & h(1) & h(0) \end{bmatrix}}_{\tilde{\mathbf{H}}} \underbrace{\begin{bmatrix} s(0) \\ s(1) \\ \vdots \\ s(N-1) \end{bmatrix}}_{\mathbf{s}}.$$

- Only possible for  $L \geq M$
- Addition of the CP and removal of the CP are implicitly included

# Receive signal processing

$$\underbrace{\begin{bmatrix} y_0 \\ y_1 \\ \vdots \\ y_{N-1} \end{bmatrix}}_{\mathbf{y}} = \frac{1}{\sqrt{N}} \underbrace{\begin{bmatrix} 1 & 1 & 1 & \dots & 1 \\ 1 & w & w^2 & \dots & w^{N-1} \\ 1 & w^2 & w^4 & \dots & w^{2(N-1)} \\ \vdots & \vdots & \vdots & & \vdots \\ 1 & w^{N-1} & w^{2(N-1)} & \dots & w^{(N-1)^2} \end{bmatrix}}_{\mathbf{R}} \underbrace{\begin{bmatrix} r(0) \\ r(1) \\ \vdots \\ r(N-1) \end{bmatrix}}_{\mathbf{r}}$$

where  $w = e^{-j2\pi\frac{1}{N}}$ .

- $\mathbf{R}$  is the normalised DFT matrix
- Rows of  $\mathbf{R}$  contain the receive signal components
- Multiplication of  $\mathbf{R}$  with  $\mathbf{r}$  corresponds to computing  $N$  correlations (one per subcarrier/subchannel)
- Can be implemented in  $\mathcal{O}(N \log N)$  using the FFT

# Transmit signal processing

$$\underbrace{\begin{bmatrix} s(0) \\ s(1) \\ \vdots \\ s(N-1) \end{bmatrix}}_{\mathbf{s}} = \frac{1}{\sqrt{N}} \underbrace{\begin{bmatrix} 1 & 1 & 1 & \dots & 1 \\ 1 & w^{-1} & w^{-2} & \dots & w^{-(N-1)} \\ 1 & w^{-2} & w^{-4} & \dots & w^{-2(N-1)} \\ \vdots & \vdots & \vdots & \ddots & \vdots \\ 1 & w^{-(N-1)} & w^{-2(N-1)} & \dots & w^{-(N-1)^2} \end{bmatrix}}_{\mathbf{R}^H} \underbrace{\begin{bmatrix} \underline{x}_0 \\ \underline{x}_1 \\ \vdots \\ \underline{x}_{N-1} \end{bmatrix}}_{\mathbf{x}}$$

- $\mathbf{R}^H$  is the normalised IDFT matrix
- The columns of  $\mathbf{R}^H$  are the transmit signal components
- $k$ th transmit symbol ( $\underline{x}_{k-1}$ ) scales the  $k$ th column of  $\mathbf{R}^H$
- Transmit multiplex is the sum of  $N$  scaled complex exponentials (scaled by the data)
- Can be implemented in  $\mathcal{O}(N \log N)$  using the IFFT

# “Diagonalisation” of the circulant channel matrix

- Note:
  - $\mathbf{R}^{-1} = \mathbf{R}^H$  ( $\mathbf{R}$  is unitary matrix)  $\rightarrow \mathbf{R}\mathbf{R}^H = \mathbf{R}^H\mathbf{R} = \mathbf{I}$
  - Any circulant matrix*  $\tilde{\mathbf{H}}$  can be written as  $\tilde{\mathbf{H}} = \mathbf{R}^H\mathbf{\Lambda}\mathbf{R}$ , where  $\mathbf{\Lambda}$  is a diagonal matrix
- Consequently, we can write

$$\mathbf{y} \triangleq \mathbf{R}\mathbf{r} = \mathbf{R}\tilde{\mathbf{H}}\mathbf{s} = \mathbf{R}\tilde{\mathbf{H}}\mathbf{R}^H\mathbf{x} = \mathbf{R} \underbrace{\tilde{\mathbf{H}}}_{\mathbf{R}^H\mathbf{\Lambda}\mathbf{R}} \mathbf{R}^H\mathbf{x} = \mathbf{\Lambda}\mathbf{x}.$$

- The DFT/IDFT matrix-pair diagonalises any circulant channel matrix
- As long as  $L \geq M$  (proper cyclic prefix), we obtain *parallel, independent* subchannels (no ISI, no ICI)

# Summary

- 1 Transmit signal processing: linear combination of transmit signal components (complex exponentials)  $\longrightarrow$  IFFT
- 2 Receive signal processing: correlation with complex exponentials  $\longrightarrow$  FFT
- 3 OFDM vs DMT
- 4 Three interpretations of cyclic prefixing
- 5 Matrix notation