

DSP-CIS

Part-I : Introduction

Chapter-3: Acoustic Modem Project

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Chapter-3: Acoustic Modem Project

- **Introduction**

Overview & Target

- **Work Plan**

Week-1

Week-2: Channel modeling & evaluation

Week 3-4: OFDM modulation

Week 5-6

Week 7-8



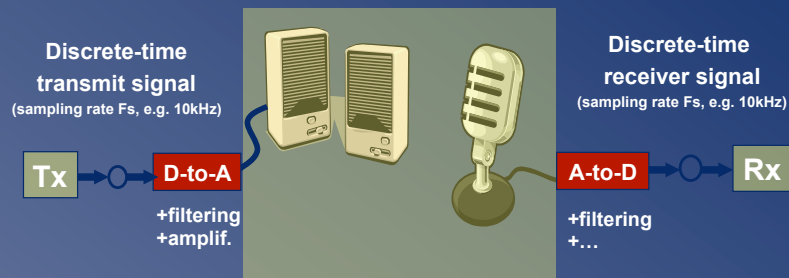
Introduction/Overview



- Digital communication over an acoustic channel (from loudspeaker to microphone)
- FFT/IFFT-based modulation format : OFDM (as in ADSL/VDSL, WiFi, DAB, DVB...)
- Channel estimation, equalization, etc...

Introduction

- Digital communications over an acoustic channel:



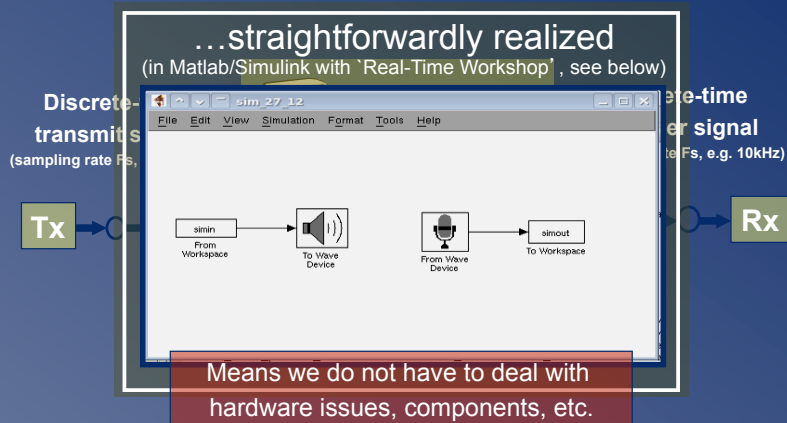
Introduction

- Digital communications over an acoustic channel:



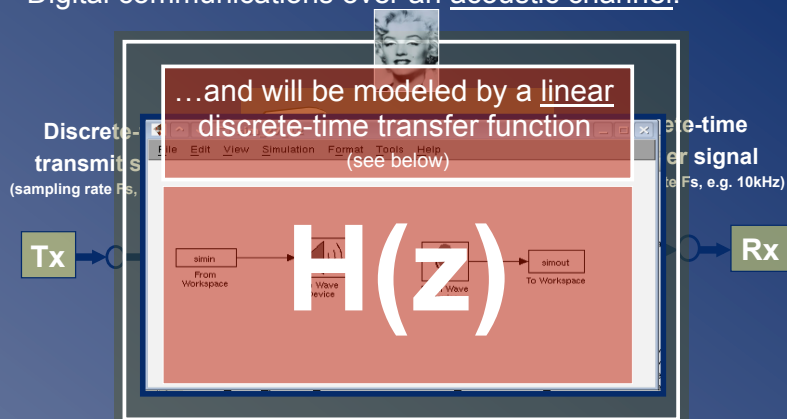
Introduction

- Digital communications over an acoustic channel:



Introduction

- Digital communications over an acoustic channel:



Introduction

- Digital communications over an acoustic channel:



Introduction

- Will use OFDM as a modulation format

Orthogonal frequency-division multiplexing From Wikipedia, the free encyclopedia

Orthogonal frequency-division multiplexing (OFDM), essentially identical to (...) **discrete multi-tone modulation (DMT)**, is a frequency-division multiplexing (FDM) scheme used as a digital multi-carrier modulation method. A large number of closely-spaced orthogonal sub-carriers are used to carry data. The data is divided into several parallel data streams or channels, one for each sub-carrier. Each sub-carrier is modulated with a conventional modulation scheme (such as quadrature amplitude modulation or phase-shift keying) at a low symbol rate, maintaining total data rates similar to conventional single-carrier modulation schemes in the same bandwidth. OFDM has developed into a popular scheme for wideband digital communication, whether wireless or over copper wires, used in applications such as digital television and audio broadcasting, wireless networking and broadband internet access.

OFDM/DMT is used in ADSL/VDSL, WiFi, DAB, DVB ...
OFDM heavily relies on DSP functionalities (FFT/IFFT, ...)

Introduction

Target:

Design efficient OFDM based modem (Tx/Rx)
for transmission over acoustic channel



Specifications:

Data rate (e.g. 1kbits/sec), bit error rate (e.g. 0.5%),
channel tracking speed, synchronisation, ...

Work Plan

8 Weeks:

- Week 0: Introduction Matlab/Simulink
- Week 1: Audio playback, recording and analysis
- Week 2: Acoustic channel measurement & modeling
deliverable
- Week 3-4: OFDM transmitter/receiver design
deliverable
- Week 5-6: OFDM over acoustic channel
deliverable
- Week 7-8: OFDM with adaptive equalization
deliverable

Week 0 / Introduction to Matlab & Simulink

MATLAB
The Language of Technical Computing

=CRUCIAL PREREQUISITE

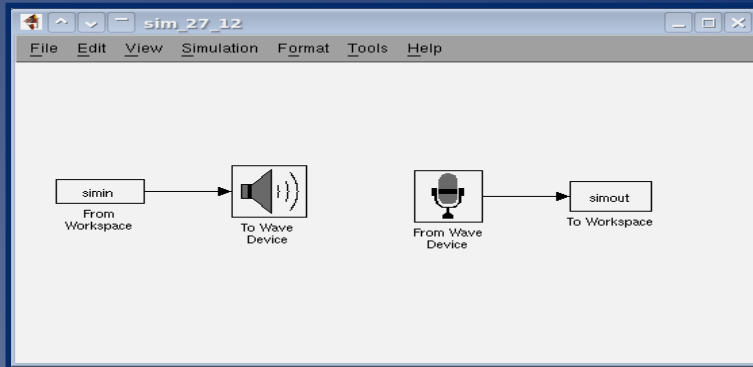
7

Matlab tutorial provided..

Self-test = exercise 6 (IF 'failure', THEN 'brush up your Matlab skills!')

Week 1 / Audio playback, recording and analysis

Will provide basic Simulink scheme... ('Real-Time Workshop')



Time-frequency analysis of recorded signals

Week 2 / Channel Modeling & Evaluation

Transmission channel consist of

- Tx 'front end' : Digital-to-Analog conv./filtering/amplification
- Loudspeaker (ps: cheap loudspeakers mostly have a non-linear characteristic ☹)
- Acoustic channel
- Microphone
- Rx 'front end' : filtering/Analog-to-Digital conv.

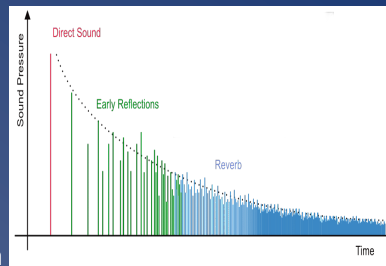


Week 2 / Channel Modeling & Evaluation

Acoustic channel ('room acoustics'):

Acoustic path between loudspeaker and microphone is represented by the **acoustic impulse response** (which can be recorded/measured)

- First there is a dead time
- Then come the direct path impulse and some early reflections, which depend on the geometry of the room
- Finally there is an exponentially decaying tail called reverberation, corresponding to multiple reflections on walls, objects,...



Week 2 / Channel Modeling & Evaluation

Complete transmission channel will be modeled by a discrete-time (FIR 'finite impulse response') transfer function

$$H(z) = h_0 + h_1 \cdot z^{-1} + h_2 \cdot z^{-2} + \dots + h_L \cdot z^{-L}$$

- Pragmatic & good-enough approximation
- Model order L depends on sampling rate (e.g. L=100...1000...)



PS: will use shorthand notation here, i.e. h_k, x_k, y_k , instead of $h[k], x[k], y[k]$

Week 2 / Channel Modeling & Evaluation

When a discrete-time (Tx) signal x_k is sent over a channel...

$$H(z) = h_0 + h_1 z^{-1} + h_2 z^{-2} + \dots + h_L z^{-L}$$

..then channel output signal (=Rx input signal) y_k is

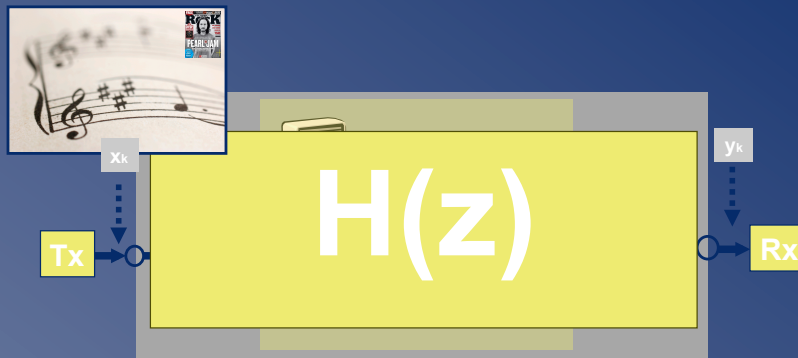
$$\begin{bmatrix} y_k \\ y_{k+1} \\ y_{k+2} \\ y_{k+3} \\ y_{k+4} \\ \vdots \\ y_{k+K} \end{bmatrix} = \begin{bmatrix} x_k & x_{k-1} & x_{k-2} & \dots & x_{k-L} \\ x_{k+1} & x_k & x_{k-1} & \dots & x_{k+1-L} \\ x_{k+2} & x_{k+1} & x_k & \dots & x_{k+2-L} \\ x_{k+3} & x_{k+2} & x_{k+1} & \dots & x_{k+3-L} \\ x_{k+4} & x_{k+3} & x_{k+2} & \dots & x_{k+4-L} \\ \vdots & \vdots & \vdots & \ddots & \vdots \\ x_{k+K} & x_{k+K-1} & x_{k+K-2} & \dots & x_{k+K-L} \end{bmatrix} \cdot \begin{bmatrix} h_0 \\ h_1 \\ h_2 \\ \vdots \\ h_L \end{bmatrix}$$

= 'convolution'

Week 2 / Channel Modeling & Evaluation

Can now run parameter estimation experiment:

1. Transmit your favorite signal x_k
2. Record corresponding signal y_k



Week 2 / Channel Modeling & Evaluation

3. Least squares estimation

$$\min_{h_0, h_1, h_2, \dots, h_L} \left\| \begin{bmatrix} y_k \\ y_{k+1} \\ y_{k+2} \\ y_{k+3} \\ y_{k+4} \\ \vdots \\ y_{k+K} \end{bmatrix} - \begin{bmatrix} x_k & x_{k-1} & x_{k-2} & \dots & x_{k-L} \\ x_{k+1} & x_k & x_{k-1} & \dots & x_{k+1-L} \\ x_{k+2} & x_{k+1} & x_k & \dots & x_{k+2-L} \\ x_{k+3} & x_{k+2} & x_{k+1} & \dots & x_{k+3-L} \\ x_{k+4} & x_{k+3} & x_{k+2} & \dots & x_{k+4-L} \\ \vdots & \vdots & \vdots & \ddots & \vdots \\ x_{k+K} & x_{k+K-1} & x_{k+K-2} & \dots & x_{k+K-L} \end{bmatrix} \begin{bmatrix} h_0 \\ h_1 \\ h_2 \\ \vdots \\ h_L \end{bmatrix} \right\|_2^2$$

(i.e. one line of Matlab code ☺)

Carl Friedrich Gauss (1777 – 1855)

Week 2 / Channel Modeling & Evaluation

Estimated transmission channel can then be analysed...

- Frequency response
- Information theoretic capacity

$$C(\text{bits/sec}) = \int_{f_{\min}}^{f_{\max}} \log_2 \left(1 + \frac{S(f)}{N(f)} \right) df$$

ps: noise spectrum?

Claude Shannon 1916-2001

Week 3-4 / OFDM modulation

OFDM – Orthogonal Frequency Division Multiplexing

a.k.a.

DMT – Discrete Multitone Modulation

Basic idea is to (QAM-)modulate (many) different carriers with low-rate bit streams. The modulated carriers are summed and then transmitted.

A high-rate bit stream is thus carried by dividing it into hundreds of low-rate streams.

Modulation/demodulation is performed by FFT/IFFT (see below)

Now 14 pages of (simple) maths/theory...

OFDM Modulation

1/14

Consider the modulation of a complex exponential carrier (with period N)

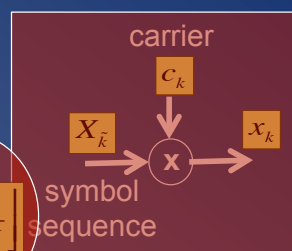
$$c_k = (e^{j2\pi/N})^k \text{ for } k = 0, 1, \dots$$

by a 'symbol sequence' (see p.27)

$$X_{\tilde{k}} \text{ for } \tilde{k} = 0, 1, \dots$$

defined as

$$x_k = c_k \cdot X_{\tilde{k}} \text{ for } k = 0, 1, \dots \text{ and } \tilde{k} = \left\lfloor \frac{k}{N} \right\rfloor$$



(i.e. "1 symbol per N samples of the carrier")

- PS: remember that modulation of sines and cosines is similar/related to modulation of complex exponentials (see also p.26, 2nd 'PS')

OFDM Modulation

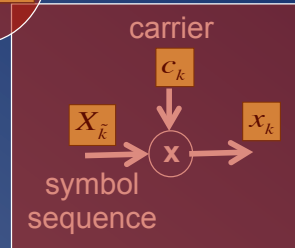
2/14

$$x_k = c_k \cdot X_{\tilde{k}} \text{ for } k = 0, 1, \dots \text{ and } \tilde{k} = \left\lfloor \frac{k}{N} \right\rfloor$$

This corresponds to (for $k=0, N, 2N, \dots$):

$$\begin{bmatrix} x_k \\ x_{k+1} \\ x_{k+2} \\ x_{k+3} \\ x_{k+4} \\ \vdots \\ x_{k+N-1} \end{bmatrix} = \begin{bmatrix} (e^{j2\pi/N})^0 \\ (e^{j2\pi/N})^1 \\ (e^{j2\pi/N})^2 \\ (e^{j2\pi/N})^3 \\ (e^{j2\pi/N})^4 \\ \vdots \\ (e^{j2\pi/N})^{N-1} \end{bmatrix} \cdot X_{\tilde{k}}$$

time-domain signal segment



OFDM Modulation

3/14

Now consider the modulation of N such complex exponential carriers

$$c_k^{(n)} = (e^{j2\pi n/N})^k \text{ for } k = 0, 1, \dots \text{ and } n = 0, 1, \dots, N-1$$

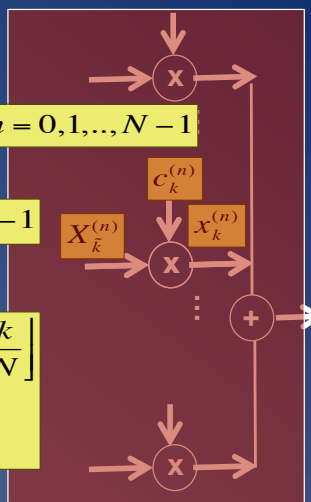
by 'symbol sequences'

$$X_{\tilde{k}}^{(n)} \text{ for } \tilde{k} = 0, 1, \dots \text{ and } n = 0, 1, \dots, N-1$$

defined as

$$x_k^{(n)} = c_k^{(n)} \cdot X_{\tilde{k}}^{(n)} \text{ for } k = 0, 1, \dots \text{ and } \tilde{k} = \left\lfloor \frac{k}{N} \right\rfloor$$

$$x_k = \sum_{n=0}^{N-1} x_k^{(n)}$$



OFDM Modulation

4/14

This corresponds to (for $k=0,N,2N,..$):

$$\begin{bmatrix} x_k \\ x_{k+1} \\ x_{k+2} \\ x_{k+3} \\ \vdots \\ x_{k+N-1} \end{bmatrix} = \begin{bmatrix} (e^{j\frac{2\pi}{N}})^0 & (e^{j\frac{2\pi}{N}})^0 & (e^{j\frac{2\pi}{N}})^0 & (e^{j\frac{2\pi}{N}})^0 & (e^{j\frac{2\pi}{N}})^0 & \dots & (e^{j\frac{2\pi}{N}})^0 \\ (e^{j\frac{2\pi}{N}})^0 & (e^{j\frac{2\pi}{N}})^1 & (e^{j\frac{2\pi}{N}})^2 & (e^{j\frac{2\pi}{N}})^3 & (e^{j\frac{2\pi}{N}})^4 & \dots & (e^{j\frac{2\pi}{N}})^{N-1} \\ (e^{j\frac{2\pi}{N}})^0 & (e^{j\frac{2\pi}{N}})^2 & (e^{j\frac{2\pi}{N}})^4 & (e^{j\frac{2\pi}{N}})^6 & (e^{j\frac{2\pi}{N}})^8 & \dots & (e^{j\frac{2\pi}{N}})^{2(N-1)} \\ (e^{j\frac{2\pi}{N}})^0 & (e^{j\frac{2\pi}{N}})^3 & (e^{j\frac{2\pi}{N}})^6 & (e^{j\frac{2\pi}{N}})^9 & (e^{j\frac{2\pi}{N}})^{12} & \dots & (e^{j\frac{2\pi}{N}})^{3(N-1)} \\ \vdots & \vdots & \vdots & \vdots & \vdots & \ddots & \vdots \\ (e^{j\frac{2\pi}{N}})^0 & (e^{j\frac{2\pi}{N}})^{N-1} & (e^{j\frac{2\pi}{N}})^{2(N-1)} & (e^{j\frac{2\pi}{N}})^{3(N-1)} & (e^{j\frac{2\pi}{N}})^{4(N-1)} & \dots & (e^{j\frac{2\pi}{N}})^{(N-1)^2} \end{bmatrix} \begin{bmatrix} X_k^{(0)} \\ X_k^{(1)} \\ X_k^{(2)} \\ X_k^{(3)} \\ \vdots \\ X_k^{(N-1)} \end{bmatrix}$$

= (N) * IDFT -matrix

..and so can be realized by means of an N-point 'Inverse Discrete Fourier Transform' (IDFT) !!!

OFDM Modulation

5/14

- PS: Note that $X_k^{(0)}$ modulates a DC signal (hence often set to zero)
- PS: To ensure time-domain signal is real-valued, have to choose

$$X_k^{(N-1)} = (X_k^{(1)})^* , X_k^{(N-2)} = (X_k^{(2)})^* , \dots$$

- **PS: The IDFT matrix is a cool matrix:**
 - For any chosen dimension N, an IDFT matrix can be constructed as given on the previous slide.
 - Its inverse is the DFT matrix (symbol 'F').
- DFT and IDFT matrices are unitary (up to a scalar), i.e.

$$F = (\text{IDFT - matrix})^{-1} = N \cdot (\text{IDFT - matrix})^H$$

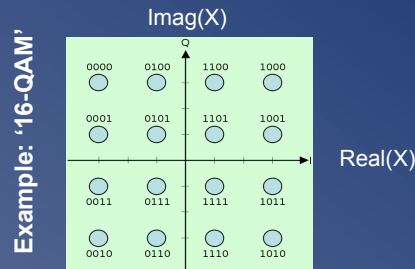
- The structure of the IDFT matrix allows for a cheap (complexity $N \cdot \log N$ instead of $N \cdot N$) algorithm to compute the matrix-vector product on the previous slide (=IFFT =inverse fast Fourier transform)

OFDM Modulation

6/14

So this will be the basic modulation operation at the Tx :

- The X 's are (QAM-symbols) defined by the input bit stream



- The time-domain signal segments $x_k, x_{k+1}, x_{k+2}, \dots, x_{k+N-1}$ are obtained by IDFT/IFFT and then transmitted over the channel, one after the other. At the Rx, demodulation is done with an inverse operation (i.e. DFT/FFT=fast Fourier transform, see also p.33).

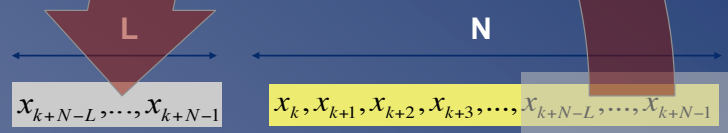
OFDM Modulation

7/14

Sounds simple, but forgot one thing: channel $H(z)$!!

OFDM has an ingenious way of dealing with the channel effect, namely through the insertion of a so-called 'cyclic prefix' at the Tx :

If the channel is FIR with order L (see p.16), then per segment, instead of transmitting N samples, $N+L$ samples are transmitted (assuming $L \ll N$) where the last L samples are copied and put up front...



OFDM Modulation

8/14

At the Rx, throw away L samples corresponding to cyclic prefix, keep the other N samples, which correspond to

$$\begin{matrix} \updownarrow N \end{matrix} \begin{bmatrix} y_k \\ y_{k+1} \\ y_{k+2} \\ y_{k+3} \\ y_{k+4} \\ \vdots \\ y_{k+N-1} \end{bmatrix} = \begin{bmatrix} h_L & h_{L-1} & \dots & h_1 & h_0 & 0 & 0 & 0 & \dots & 0 \\ 0 & h_L & \dots & h_2 & h_1 & h_0 & 0 & 0 & \dots & 0 \\ 0 & 0 & \dots & h_3 & h_2 & h_1 & h_0 & 0 & \dots & 0 \\ 0 & 0 & \dots & h_4 & h_3 & h_2 & h_1 & h_0 & \dots & 0 \\ \vdots & \vdots & \vdots & \vdots & \vdots & \vdots & \vdots & \vdots & \vdots & \vdots \\ 0 & 0 & \dots & 0 & 0 & 0 & 0 & 0 & \dots & h_0 \end{bmatrix} \begin{matrix} \text{prefix} \rightarrow \\ \begin{bmatrix} x_{k+N-L} \\ x_{k+N-L+1} \\ \vdots \\ x_{k+N-1} \\ x_k \\ x_{k+1} \\ x_{k+2} \\ x_{k+3} \\ \vdots \\ x_{k+N-1} \end{bmatrix} \end{matrix} \begin{matrix} \updownarrow N+L \end{matrix}$$

This is equivalent to ...

OFDM Modulation

9/14

$$\begin{matrix} \updownarrow N \end{matrix} \begin{bmatrix} y_k \\ y_{k+1} \\ y_{k+2} \\ y_{k+3} \\ y_{k+4} \\ \vdots \\ y_{k+N-1} \end{bmatrix} = \begin{bmatrix} h_L & h_{L-1} & \dots & h_1 & h_0 & 0 & 0 & 0 & \dots & 0 \\ 0 & h_L & \dots & h_2 & h_1 & h_0 & 0 & 0 & \dots & 0 \\ 0 & 0 & \dots & h_3 & h_2 & h_1 & h_0 & 0 & \dots & 0 \\ 0 & 0 & \dots & h_4 & h_3 & h_2 & h_1 & h_0 & \dots & 0 \\ \vdots & \vdots & \vdots & \vdots & \vdots & \vdots & \vdots & \vdots & \vdots & \vdots \\ 0 & 0 & \dots & 0 & 0 & 0 & 0 & 0 & \dots & h_0 \end{bmatrix} \begin{bmatrix} 0 & 1 & I_{L,L} \\ & & I_{N,N} \end{bmatrix} \begin{matrix} \updownarrow N \\ \begin{bmatrix} x_k \\ x_{k+1} \\ x_{k+2} \\ x_{k+3} \\ x_{k+4} \\ \vdots \\ x_{k+N-1} \end{bmatrix} \end{matrix}$$

$$\begin{bmatrix} h_0 & 0 & 0 & 0 & 0 & \dots & h_3 & h_2 & h_1 \\ h_1 & h_0 & 0 & 0 & 0 & \dots & h_4 & h_3 & h_2 \\ h_2 & h_1 & h_0 & 0 & 0 & \dots & h_5 & h_4 & h_3 \\ h_3 & h_2 & h_1 & h_0 & 0 & \dots & & & h_4 \\ h_4 & h_3 & h_2 & h_1 & h_0 & \dots & & & \\ \vdots & \vdots & \vdots & \vdots & \vdots & \vdots & \vdots & \vdots & \vdots \\ 0 & 0 & 0 & 0 & 0 & \dots & h_2 & h_1 & h_0 \end{bmatrix} \quad (*)$$

The resulting matrix (call it 'H') is an NxN 'circulant matrix' = every row is the previous row up to a 'cyclic shift'

OFDM Modulation

10/14

- PS: Cyclic prefix converts a (linear) convolution (see p.29) into a so-called 'circular convolution' (see p.30)
- **Circulant matrices** are cool matrices...
A weird property (proof by Matlab!) is that when a circulant matrix H is pre-/post-multiplied by the DFT/IDFT matrix, a diagonal matrix is always obtained: $\mathbf{F}\mathbf{H}\mathbf{F}^{-1} = [\text{diagonal matrix}]$

Hence, a circulant matrix can **always** be written as
(=eigenvalue decomposition!)

$$\mathbf{H} = \mathbf{F}^{-1} \begin{bmatrix} H_0 & 0 & \dots & 0 \\ 0 & H_1 & \dots & 0 \\ \vdots & \vdots & \dots & \vdots \\ 0 & 0 & \dots & H_{N-1} \end{bmatrix} \mathbf{F}$$

OFDM Modulation

11/14

Combine previous formulas, to obtain...

$$\mathbf{F} \begin{bmatrix} y_k \\ y_{k+1} \\ y_{k+2} \\ y_{k+3} \\ y_{k+4} \\ \vdots \\ y_{k+N-1} \end{bmatrix} = \begin{bmatrix} H_0 & 0 & \dots & 0 \\ 0 & H_1 & \dots & 0 \\ \vdots & \vdots & \dots & \vdots \\ 0 & 0 & \dots & H_{N-1} \end{bmatrix} \mathbf{F} \begin{bmatrix} x_k \\ x_{k+1} \\ x_{k+2} \\ x_{k+3} \\ x_{k+4} \\ \vdots \\ x_{k+N-1} \end{bmatrix}$$

$$\mathbf{F} \begin{bmatrix} y_k \\ y_{k+1} \\ y_{k+2} \\ y_{k+3} \\ y_{k+4} \\ \vdots \\ y_{k+N-1} \end{bmatrix} = \begin{bmatrix} H_0 & 0 & \dots & 0 \\ 0 & H_1 & \dots & 0 \\ \vdots & \vdots & \dots & \vdots \\ 0 & 0 & \dots & H_{N-1} \end{bmatrix} \mathbf{F} \begin{bmatrix} x_k \\ x_{k+1} \\ x_{k+2} \\ x_{k+3} \\ x_{k+4} \\ \vdots \\ x_{k+N-1} \end{bmatrix}$$

OFDM Modulation

12/14

In other words...

$$\begin{bmatrix} Y_{\tilde{k}}^{(0)} \\ Y_{\tilde{k}}^{(1)} \\ Y_{\tilde{k}}^{(2)} \\ Y_{\tilde{k}}^{(3)} \\ Y_{\tilde{k}}^{(4)} \\ \vdots \\ Y_{\tilde{k}}^{(N-1)} \end{bmatrix} = \mathbf{F} \begin{bmatrix} 3k \\ 3k+1 \\ 3k+2 \\ 3k+3 \\ \vdots \\ 3k+N-1 \end{bmatrix} = \begin{bmatrix} H_0 & 0 & 0 & 0 & 0 & \dots & 0 \\ 0 & H_1 & 0 & 0 & 0 & \dots & 0 \\ 0 & 0 & H_2 & 0 & 0 & \dots & 0 \\ 0 & 0 & 0 & H_3 & 0 & \dots & 0 \\ \vdots & \vdots & \vdots & \vdots & \vdots & \ddots & \vdots \\ 0 & 0 & 0 & 0 & 0 & \dots & H_{N-1} \end{bmatrix} \begin{bmatrix} X_{\tilde{k}}^{(0)} \\ X_{\tilde{k}}^{(1)} \\ X_{\tilde{k}}^{(2)} \\ X_{\tilde{k}}^{(3)} \\ X_{\tilde{k}}^{(4)} \\ \vdots \\ X_{\tilde{k}}^{(N-1)} \end{bmatrix}$$

This means that after removing the prefix part and performing a DFT in the Rx, the obtained Y's are equal to the transmitted X's, up to (scalar) channel attenuations H_n (!!)

$$\rightarrow Y_{\tilde{k}}^{(n)} = H_n \cdot X_{\tilde{k}}^{(n)}$$

OFDM Modulation

13/14

- PS: It can be shown (check first column of $\mathbf{F}\mathbf{H} = [\text{diagonal matrix}]\mathbf{F}$) that H_n is the channel frequency response evaluated at the n-th carrier !

$$\begin{bmatrix} H_0 \\ H_1 \\ H_2 \\ H_3 \\ \vdots \\ H_{N-1} \end{bmatrix} = \mathbf{F} \cdot (\text{1st column of } \mathbf{H}) = \mathbf{F} \cdot \begin{bmatrix} h_0 \\ h_1 \\ \vdots \\ h_L \\ 0 \\ \vdots \\ 0 \end{bmatrix} \Rightarrow H_n = H(z) \Big|_{z=e^{j2\pi n/N}}$$

(p.33 then represents 'frequency domain version' of circular convolution, i.e. 'component-wise multiplication in the frequency domain')

OFDM Modulation

13/14

- PS: It can be shown (check first column of $\mathbf{F}\mathbf{H} = [\text{diagonal matrix}]\mathbf{F}$) that H_n is the channel frequency response evaluated at the n-th carrier !

$$\begin{bmatrix} H_0 \\ H_1 \\ H_2 \\ H_3 \\ \vdots \\ H_{N-1} \end{bmatrix} = \mathbf{F} \cdot (\text{1st column of } \mathbf{H}) = \mathbf{F} \cdot \begin{bmatrix} h_0 \\ h_1 \\ \vdots \\ h_L \\ 0 \\ \vdots \\ 0 \end{bmatrix} \Rightarrow H_n = H(z)|_{z=e^{j2\pi n/N}}$$

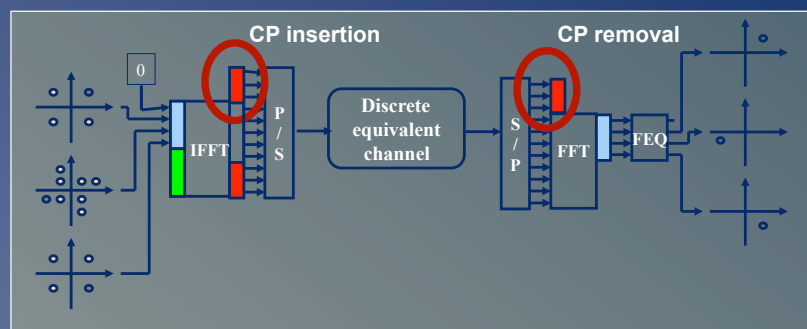
'Channel equalization' may then be performed after the DFT (=in the frequency domain), by component-wise division (divide by H_n for carrier-n). This is referred to as '1-tap FEQ' (Freq.-domain EQualization)

$$Y_{\bar{k}}^{(n)} = H_n \cdot X_{\bar{k}}^{(n)} \Rightarrow \text{estimate}\{X_{\bar{k}}^{(n)}\} = (H_n)^{-1} \cdot Y_{\bar{k}}^{(n)}$$

OFDM Modulation

14/14

- Conclusion: DMT-modulation with cyclic prefix leads to a simple (trivial) channel equalization problem (!!)



Week 3-4

Target Week 3-4:

- Study/understand OFDM scheme
Surf around, use IEEE Xplore, Wikipedia, etc.
- Simulate basic OFDM Transceiver in Matlab
First without channel dispersion & without noise, then with noise, then with channel (model from Week-2)
- Optional : Extend OFDM Tx/Rx with `bit-loading`
=Carriers with a high SNR transmit more bits/sec

Week 5-6

Target Week 5-6:

- OFDM over acoustic channel, with basic Simulink (Real-time Workshop) scheme (Week-1)
- Extend OFDM Tx/Rx with mechanism for channel estimation and/or equalizer (FEQ) initialization/updating based on transmitted training symbols



Week 7-8

Target Week 7-8:

- OFDM over acoustic channel, with decision-directed adaptive equalization (see Part III)



Important !

PS: groups of 2

- Runs over 8 weeks
 - Each week
 - 1 PC/Matlab session (supervised, 2.5hrs)
 - 2 'Homework' sessions (unsupervised, 2*2.5hrs)
- PS: Time budget = $8 \times (2.5\text{hrs} + 5\text{hrs}) = 60\text{ hrs}$
- 'Deliverables' after week 2, 4, 6, 8
 - Grading: based on deliverables, evaluated during sessions
main part=80%, optional part=20%
 - TAs: amin.hassani@kuleuven.be (English+Persian)
mohit.sharma@kuleuven.be (English+Hindi)
robbe.vanrompaey@kuleuven.be (Dutch+English+French)
jeroen.verdyck@kuleuven.be (Dutch+English+French)

BETHERE !!