

List of references in Signal Processing

- 1) Charles W. Therrien, "Discrete Random Signals and Statistical Signal Processing", Prentice Hall, 1992
- 2) J.G. Proakis and D. Manolakis, "Digital Signal Processing", MacMillan, second edition, 1992
- 3) A. Oppenheim, R. Schaffer, "Discrete-time Signal Processing", Prentice Hall 1989, 1999
- 4) J. G. Proakis et al., "Advanced Digital Signal Processing", McMillan, 1992
- 5) M. H. Hayes, "Statistical Digital Signal Processing and Modeling", John Wiley & Sons, Inc, 1996.
- 6) S. Haykin, "Adaptive Filter Theory", Prentice Hall, Third Edition, 1996
- 7) C.L. Nikias and A. Petropulu, "Higher-Order Spectra Analysis", Prentice Hall, 1993
- 8) S. Haykin, Editor, "Blind Deconvolution", Prentice Hall, 1994
- 9) C.L. Nikias and M. Shao, "Signal Processing with α -stable Distributions and applications", New York: John Wiley & Sons, Inc. 1995.



THE SAMPLING THEOREM

Question: How often must I sample an analog signal in order not to lose information from it.

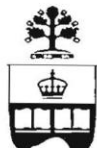
Answer: If the highest frequency contained in a signal $x_a(t)$ is Ω_0 and the signal is uniformly sampled at a rate $\Omega_s \geq 2\Omega_0$, then $x_a(t)$ can be exactly recovered from its sample values using the interpolation function

$$g(t) = \frac{\sin[\Omega_s t/2]}{\Omega_s t/2}$$

and then, $x_a(t) = \sum_{k=-\infty}^{\infty} x_a(kT) g(t - kT)$, where $\{x_a(kT)\}$ are the samples of $x_a(t)$ taken with rate $T = 2\pi/\Omega_s$ sec.

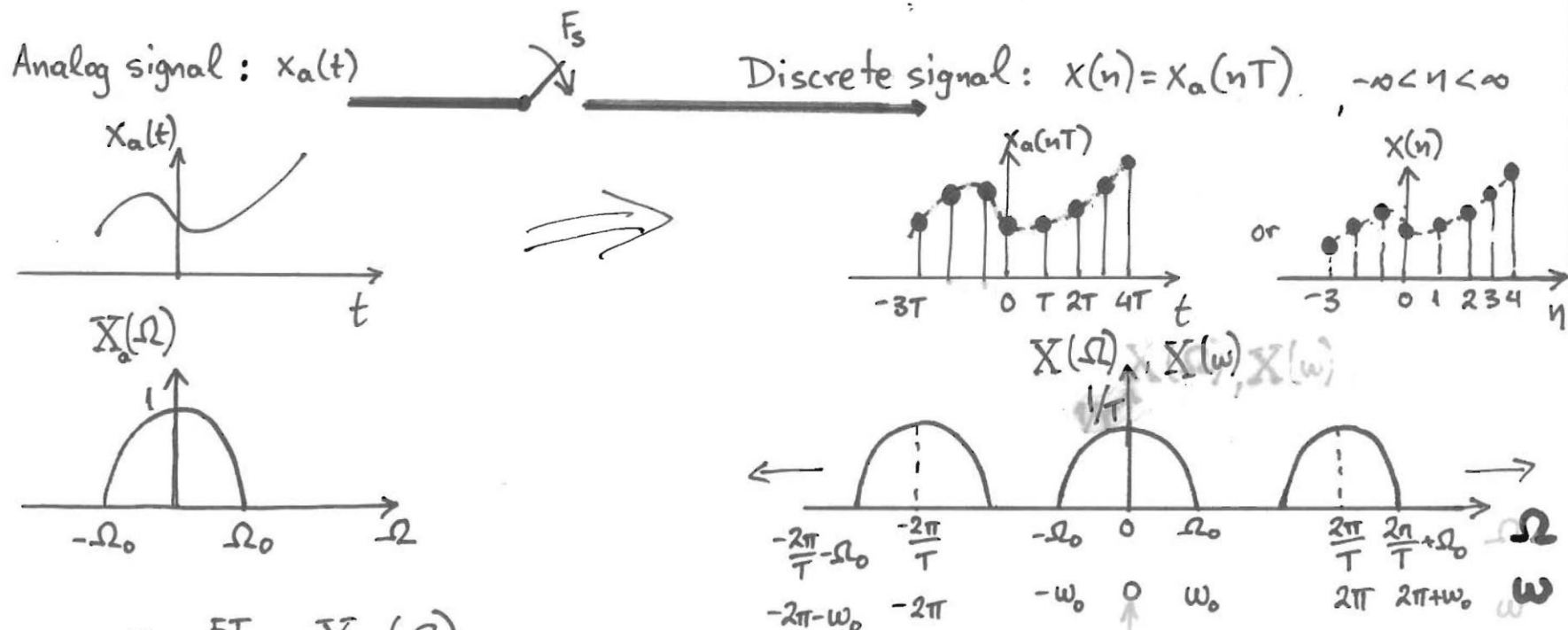
Comments: - "Aliasing" that is spectral overlapping is caused by inadequate sampling

- Aliasing is not reversible
- Assuming no aliasing has occurred $x_a(t)$ can be reconstructed via ideal LP filtering
- In practice: always choose a rate $\Omega > 2\Omega_0$; there is a need for antialiasing filtering; there are no "ideal" LP filters.



ANALOG TO DIGITAL CONVERSION (A/D)

● Ideal Sampling (uniform) : take one sample every T seconds (sampling period $F_s = \frac{1}{T}$ cycles/sec)



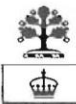
$$x_a(t) \xleftrightarrow{\text{F.T.}} X_a(\Omega)$$

$$x_a(nT) \xleftrightarrow{\text{DTFT}} X(\Omega) = \frac{1}{T} \sum_{k=-\infty}^{\infty} X_a(\Omega + k \frac{2\pi}{T})$$

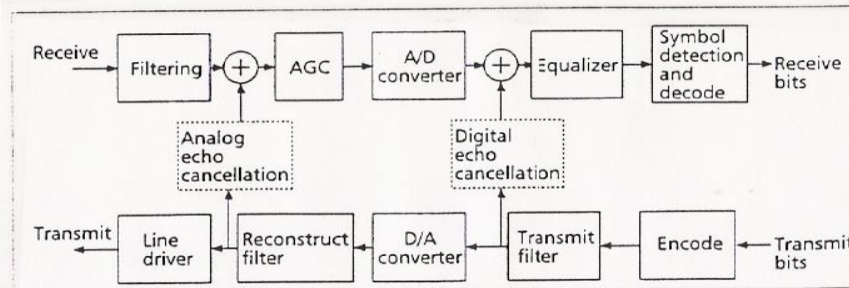
$$x(n) \xleftrightarrow{\text{DTFT}} X(\omega) = \frac{1}{T} \sum_{k=-\infty}^{\infty} X_a(\omega + k 2\pi)$$

$$\omega = \Omega T, \quad f = \frac{F}{F_s}$$

ω : Normalized frequency in rads/sample
 Ω : Real frequency in rads/sec



ANALOG TO DIGITAL CONVERSION (A/D)



■ Figure A generic digital communication transceiver.

* Analog and digital signal processors/units are used simultaneously.

* Since 1990 the system design approach places more emphasis on digital processing

* Multiple A/D, D/A conversions may be required

Type	Number of bits	Signal bandwidth	Sampling rate	Latency
Flash	6-8	$< f_s/2$	150-300 ms/s	> 1 clock cycle
Two-step	$\approx 10-12$	$< f_s/2$	10-100 ms/s	> 2 clock cycles
Pipeline	10-15	$< f_s/2$	1-100 ms/s	$> \#$ of stages
Algorithmic	12-14	$< f_s/2$	10 ks/s to < 1 ms/s	$> \#$ of bits
Succ. approx.	10-14	$< f_s/2$	100 ks/s to 5 ms/s	$> \#$ of bits
Lowpass $\Sigma-\Delta$	14-20	20-1000 kHz	2-50 MHz	N/A
Bandpass $\Sigma-\Delta$	8-15	30-1250 kHz	2-200 MHz	N/A

■ Table 1. Typical state-of-the-art performance of CMOS- and BiCMOS-based A/D converters.

Nyquist rate converters: quantize input samples every $1/f_s$ sec, ($f_s > \text{Nyquist rate}$); generate one digital sample from a single analog sample (good performance in the 10-12 bit/sample range)

Noise shaping converters: oversample analog signal; generate one digital sample by weighting many input analog samples; (Good performance in the 13-15 bit range)

* Digital converters for signals with bandwidths greater than 100 MHz are commonly used.



"Data Converters for Communication Systems"
From: IEEE Communications Magazine, October 1998, page 113

University of Toronto
Department of Electrical and Computer Engineering

THE CONCEPT OF FREQUENCY FOR DISCRETE TIME SIGNALS

⑩ $x(n) = e^{j\omega n}$, $n=0, \pm 1, \pm 2, \dots$

- ω : rads/sample
- periodic only if $f = \frac{\omega}{2\pi}$ is rational that is $f = \frac{k}{N}$ where k, N co-prime integers then $x(n) = x(n + lN)$

- Let $\omega_1 = \omega_2 + 2k\pi$. Then $e^{j\omega_1 n} = e^{j\omega_2 n}$
Thus distinct discrete exponentials (complex) can be obtained only in intervals of 2π
i.e., $-\pi \leq \omega \leq \pi$ or $0 \leq \omega \leq 2\pi$
 $-\frac{1}{2} \leq f \leq \frac{1}{2}$ or $0 \leq f \leq 1$

- The highest rate of oscillation for a discrete complex exponential is achieved for $\omega = \pi$ or $f = 1/2$

⑪ $x_k(n) = e^{jK \frac{2\pi}{N} n}$, $n=0, \pm 1, \pm 2, \dots$ N: integer

- periodic in both k and n with period N samples that is

$$x_{k+N}(n+N) = x_k(n)$$

- Thus, there are only N distinct periodic exponentials (complex) of period N

- Highest oscillation rate for $k = \frac{N}{2}$

- Important relation.

$$\frac{1}{N} \sum_{n=0}^{N-1} e^{jK \frac{2\pi}{N} n} = \begin{cases} 1 & \text{for } k=0, \pm N, \pm 2N, \dots \\ 0 & \text{otherwise.} \end{cases}$$

DTFT, DTFS, DFT

Given a discrete signal $x(n)$, $n=0, \pm 1, \pm 2, \dots$ form $\tilde{x}(n) = \sum_{l=-\infty}^{\infty} x(n+lN)$

① Discrete time Fourier transform pair (DTFT)

$$\left[\begin{array}{l} X(\omega) = \sum_{n=-\infty}^{\infty} x(n) e^{-j\omega n} \\ x(n) = \frac{1}{2\pi} \int_{-\pi}^{\pi} X(\omega) e^{j\omega n} d\omega \end{array} \right] \begin{array}{l} x(n) \text{ discrete non-} \\ \text{periodic} \\ X(\omega) \text{ continuous} \\ \text{periodic } (2\pi) \end{array}$$

② Discrete time Fourier Series pair (DTFS or DFS)

$$\left[\begin{array}{l} \tilde{X}(k) = \sum_{n=0}^{N-1} \tilde{x}(n) e^{-j2\pi \frac{kn}{N}} \\ \tilde{x}(n) = \frac{1}{N} \sum_{k=0}^{N-1} \tilde{X}(k) e^{j2\pi \frac{kn}{N}} \end{array} \right] \begin{array}{l} \tilde{x}(n), \tilde{X}(k) \\ \text{discrete and} \\ \text{periodic with} \\ \text{period } N \text{ samples} \end{array}$$

for $n, k = 0, 1, \dots, N-1$ or over $\langle N \rangle$

- N is known a-priori or can be chosen arbitrarily
- Sufficient condition for existence $\sum_{n=-\infty}^{\infty} |x(n)|^2 < \infty$



N-DFT (Discrete Fourier Transform)

Given a discrete signal $x(n)$, $n=0, \pm 1, \pm 2, \dots$ with DTFT $X(\omega)$

1) Form the signal $\tilde{x}(n) = \sum_{l=-\infty}^{\infty} x(n+lN)$, N : chosen arbitrarily

2) Form the signal $\tilde{X}(k) = X(\omega) \Big|_{\omega = \frac{2\pi k}{N}}$, $k=0, \pm 1, \pm 2, \dots$

Then,

N-DFT pair

$$\left[\begin{array}{l} X(k) = \sum_{n \in \langle N \rangle} \tilde{x}(n) e^{-j \frac{2\pi k n}{N}} \\ x(n) = \frac{1}{N} \sum_{k \in \langle N \rangle} \tilde{X}(k) e^{j \frac{2\pi k n}{N}} \end{array} \right] \xrightarrow{x(n) \text{ of length } \leq N} \left[\begin{array}{l} \tilde{X}(k) = \sum_{n=0}^{N-1} x(n) e^{-j \frac{2\pi k n}{N}} \\ x(n) = \frac{1}{N} \sum_{k=0}^{N-1} \tilde{X}(k) e^{j \frac{2\pi k n}{N}} \end{array} \right]$$

for $n, k = 0, 1, \dots, N-1$ or any period $\langle N \rangle$

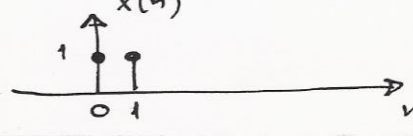
● The N-DFT maps one period of $\sum_{l=-\infty}^{\infty} x(n+lN)$ to one period of $\tilde{X}(\omega = \frac{2\pi k}{N})$

● If we do not take sufficient number of (samples) in the DTFT domain, then

"time aliasing" occurs in time domain (dual of sampling theorem)



Ex. Let



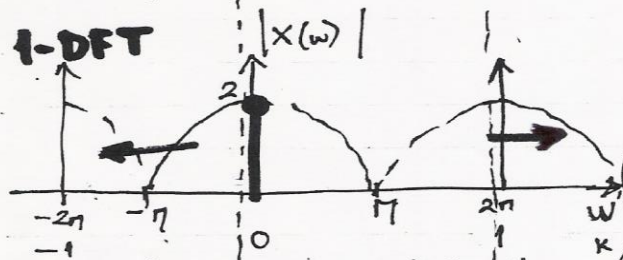
$$X_+(w) = 2 \cos \frac{w}{2} e^{-jw/2} = 1 + e^{-jw}$$

Evaluate: 1-DFT, 2-DFT, 3-DFT, 8-DFT

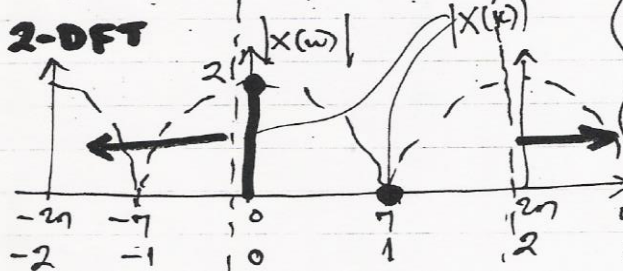
A) Working in frequency domain.

$$|X(k)| = X(w = \frac{2\pi k}{N}), \quad k=0, \dots, N-1$$

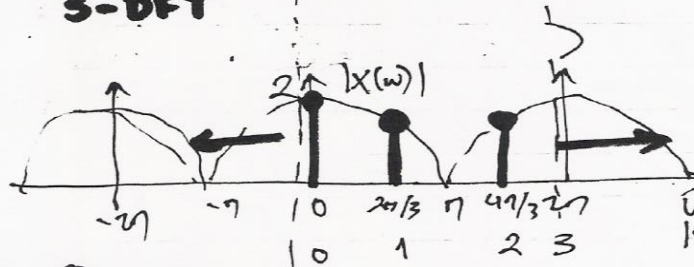
1-DFT



2-DFT



3-DFT



8-DFT

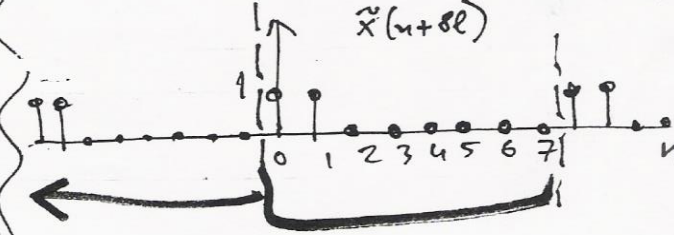
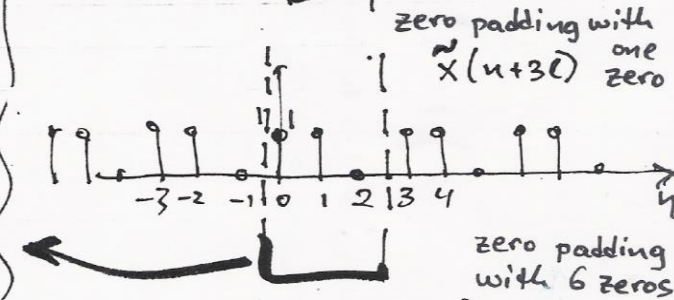
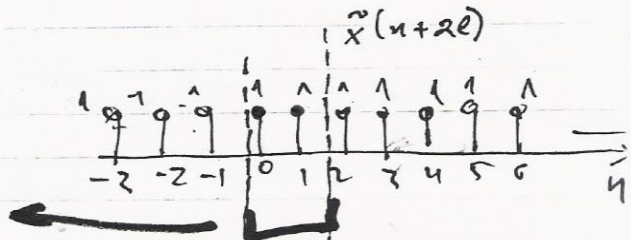
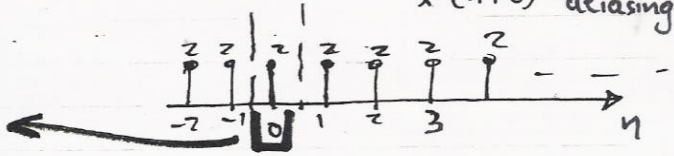


B) Time domain $\xrightarrow{N\text{-DFT}}$ Frequency domain

$$\tilde{X}(n) = \sum_{l=-\infty}^{\infty} X(n+lN)$$

$$X(k) = \sum_{n=0}^{N-1} \tilde{X}(n) e^{-j\frac{2\pi kn}{N}}$$

$\tilde{X}(n+l)$: time aliasing



ZERO PADDING - DIGITAL INTERPOLATION

Given $x(n)$ of length less or equal to N and its N -DFT $X(k)$, $k=0,1,\dots,N-1$

- ① Zero-padding increases the effective period of the N -DFT by introducing zeros in the "stop-band" of either the signal or the discrete Fourier transform
- ② By zero padding $x(n)$ with $K \cdot N$ zeros and taking $(K+1)N$ -DFT, then K new values are interpolated between (any) two values of the N -DFT
- ③ By zero padding $X(k)$ with $K \cdot N$ zeros (between the samples $k=N/2, N/2+1$) and taking $(K+1)N$ -IDFT and multiplying by $(K+1)$, K new values are interpolated between any two values of $x(n)$.

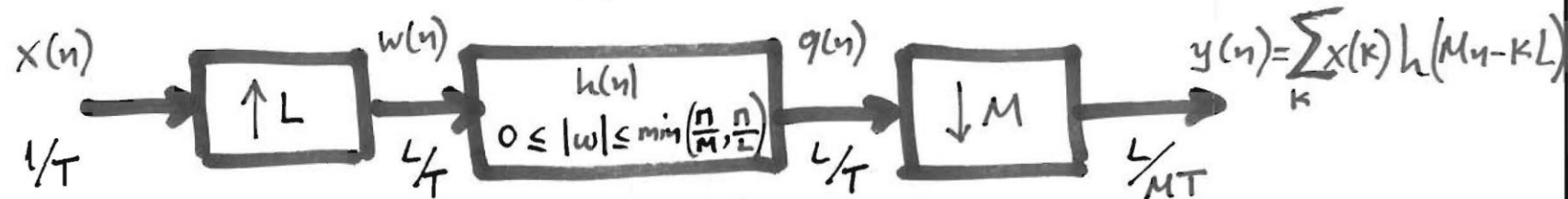
i.e

INTERPOLATION OF
 L values in
 $x(n)$



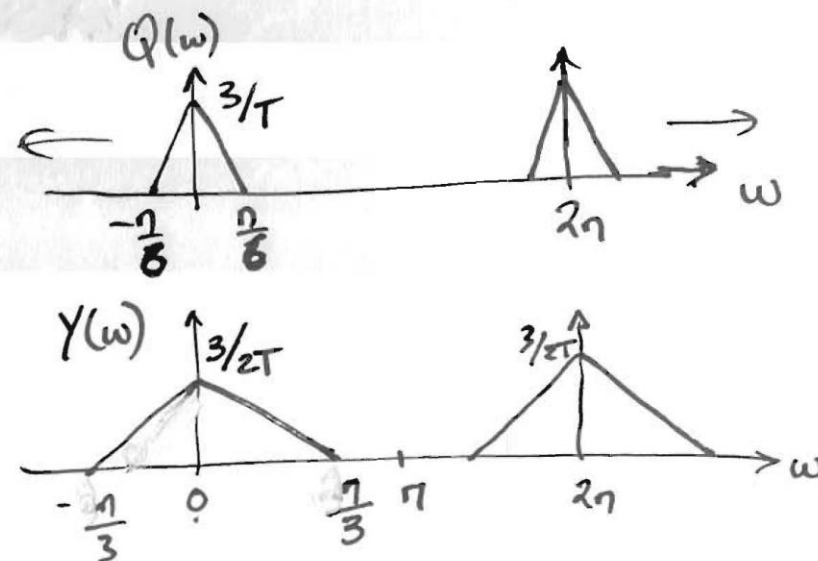
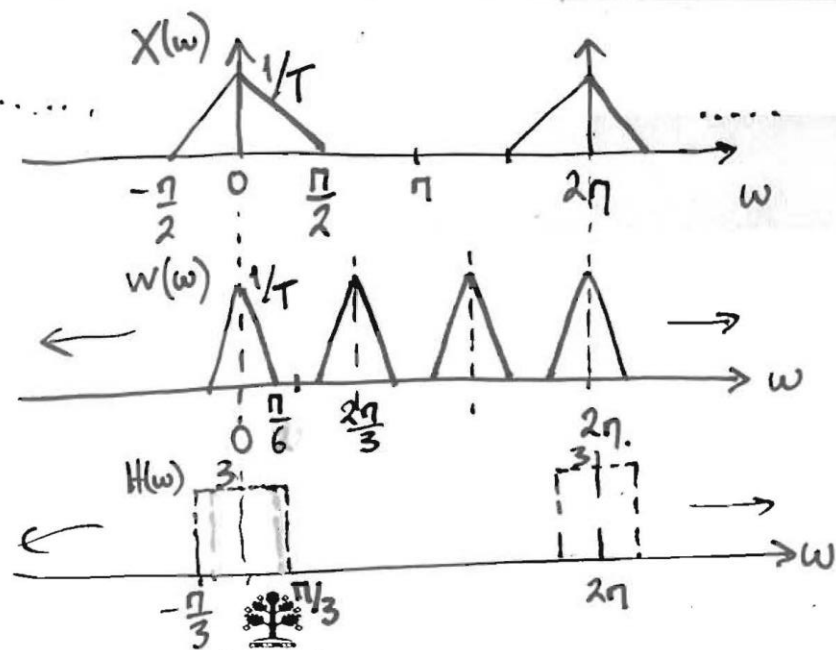
MULTIRATE SYSTEMS (3)

① Change of sampling rate by $\frac{L}{M}$: non-integer



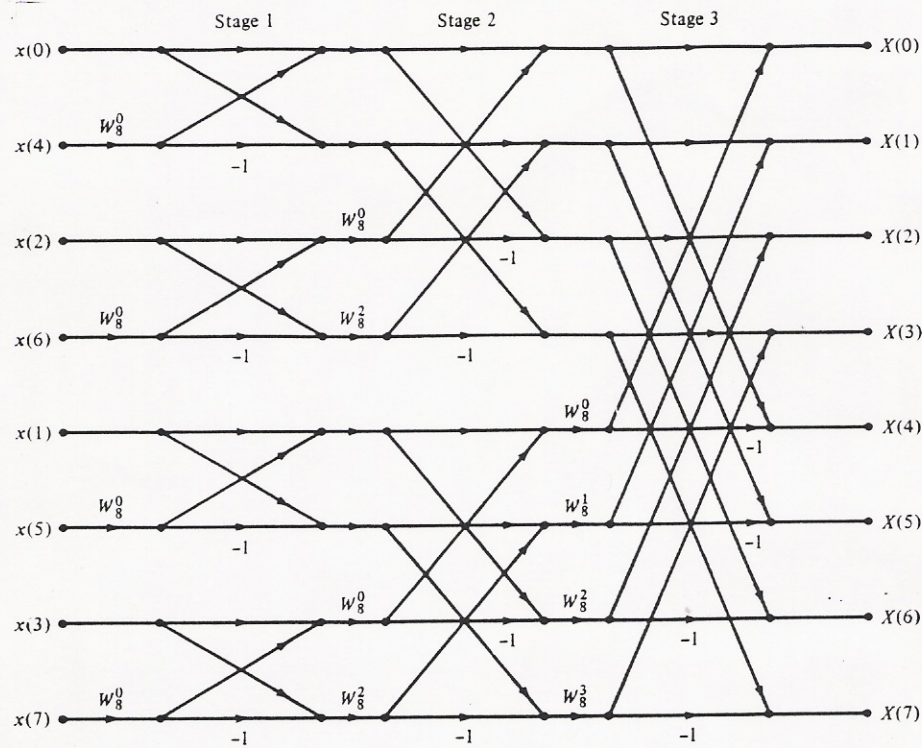
$$Y(\omega) = \frac{1}{M} \sum_{i=0}^{M-1} H\left(\frac{\omega - 2\pi i}{M}\right) X\left(\frac{\omega - 2\pi i}{M} L\right) \xrightarrow{\text{no-aliasing}} \frac{1}{M} H\left(\frac{\omega}{M}\right) X\left(\frac{\omega L}{M}\right)$$

Ex. $L=3, M=2$, no-aliasing



Radix 2 ($N=2^9$) Decimation in time FFT

Input
in bit
reversed
order



Output
in
normal
order

FIGURE Eight-point decimation-in-time FFT algorithm.

Decimation in time FFT is based on the calculation of successively smaller DFT's by decomposing the signal into successively smaller sequences

In general for $N=2^9$ - FFT. (Decimation in time).

1) $q = \log_2 N$ stages, $N/2$ butterflies / stage

2) m -th stage, 2^{m-1} different butterflies $\left\{ \begin{array}{l} \text{multiply lower branches of each} \\ \text{butterfly } W_N^{lN/2^m}, l=0, \dots, 2^{m-1} \end{array} \right.$

3) # of \otimes_c : $1/\text{butterfly} \times \frac{N}{2} \text{ butts/stage} \times \log_2 N \text{ stages} = \frac{N}{2} \log_2 N$

of \oplus_c : $2/\text{"} \times \frac{N}{2} \text{"} \times \text{"} \text{"} \text{"} = N \log_2 N$

of storage locations: $2/\text{butterfly} \times \frac{N}{2} \text{ butts.} = (\text{in place computation}) = N$

4) $2^{\log_2 N - 1}$ exponentials stored in look up table

5) $G(N) = \frac{\text{\# of FFF } \otimes}{\text{\# of DFT } \otimes} = \frac{1}{2N} \log_2 N \ll 1$. For ex. $G(8) = \frac{3}{16}$

$G(4096) \approx 0.0015$

Linear Algebra Relations

Note Title

02/09/2005

→ Vector $\underline{x} = \begin{bmatrix} x_1 \\ \vdots \\ x_n \end{bmatrix}$ ($n \times 1$)
real or complex

→ Transpose: $\underline{x}^T = [x_1, \dots, x_n]$ ($1 \times n$)

→ Hermitian transpose: $\underline{x}^H = (\underline{x}^T)^* = [x_1^*, \dots, x_n^*]$

→ Magnitude of a vector (norm or distance)

Euclidean L_2 norm: $\|\underline{x}\|_2 = \sqrt{\sum_{i=1}^n |x_i|^2}$

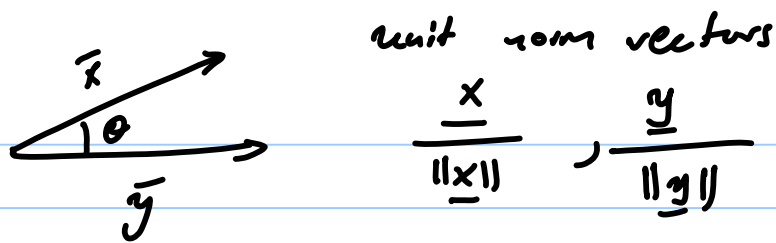
L_1 norm: $\|\underline{x}\|_1 = \sum_{i=1}^n |x_i|$

L_∞ norm: $\|\underline{x}\|_\infty = \max_i |x_i|$

→ distance between vectors $d(\underline{x}, \underline{y}) = \|\underline{x} - \underline{y}\|$

→ Inner product: $\langle \underline{x}, \underline{y} \rangle = \underline{x}^H \cdot \underline{y} = \sum_{i=1}^n x_i^* y_i$

for Euclidean space: $\langle \underline{x}, \underline{y} \rangle = \|\underline{x}\| \|\underline{y}\| \cos \theta$



→ $\underline{x}, \underline{y}$: orthogonal , $\langle \underline{x}, \underline{y} \rangle = 0$

→ $\underline{x}, \underline{y}$: linearly independent if

$$a_1 \underline{x} + a_2 \underline{y} = 0 \rightarrow a_1 = a_2 = 0$$

(in this case $\underline{x}, \underline{y}$ can be seen as the generating vectors of a 2-D space where each vector can be obtained by linear combination of \underline{x} and \underline{y}).

Ex: Let $x[n] \xrightarrow{\text{FIR, } h[n]} y[n]$

$$y[n] = \sum_{i=0}^{N-1} h[i] x[n-i] = \underline{h}^T \underline{x}[n]$$

where: $\underline{h}^T = [h_0, \dots, h_{N-1}]$, $\underline{x}[n] = [x[n], \dots, x[n-N+1]]$

Now, given a matrix \underline{A} : $n \times m$

→ Square matrix \underline{A} : $n \times n$ elements

→ symmetric square matrix: $\underline{A} = \underline{A}^T$

→ Hermitian square matrix: $\underline{A} = \underline{A}^H$

$$\text{Also: } (\underline{A} + \underline{B})^H = \underline{A}^H + \underline{B}^H$$

$$(\underline{A}^H)^H = \underline{A}$$

$$(\underline{A}\underline{B})^H = \underline{B}^H \underline{A}^H$$

→ Rank of a matrix: $\rho(\underline{A}) \leq \min(n, m)$
is the number of linearly independent columns.

$$\text{Note: } \rho(\underline{A}) = \rho(\underline{A}^H \underline{A}) = \rho(\underline{A} \underline{A}^H)$$

→ Inverse \underline{A}^{-1} of a square matrix \underline{A} : $\underline{A}^{-1} \underline{A} = \underline{I}$

where $\underline{I} = \begin{bmatrix} 1 & & 0 \\ & \ddots & \\ 0 & & 1 \end{bmatrix}$, $n \times n$ identity matrix

→ Determinant of a matrix \underline{A} ($n \times n$)

$$\det(\underline{A}) = \sum_{i=1}^n (-1)^{i+j} a_{ij} \det(\underline{A}_{ij})$$

where \underline{A}_{ij} is the $(n-1) \times (n-1)$ matrix obtained by deleting the i -th row and j -th column of \underline{A}

→ An $n \times n$ matrix \underline{A} is invertible iff $\det(\underline{A}) \neq 0$

Also, $\det(\underline{AB}) = \det(\underline{A}) \det(\underline{B})$

$$\det(\underline{A}^T) = \det(\underline{A})$$

$$\det(a \underline{A}) = a^n \det(\underline{A})$$

$$\det(\underline{A}^{-1}) = 1/\det(\underline{A})$$

→ trace: $\text{tr}(\underline{A}) = \sum_{i=1}^n a_{ii}$

→ Linear equations: $\underline{A} \underline{x} = \underline{b} \Rightarrow \underline{x} = \underline{A}^{-1} \underline{b}$

(if $\underline{A}: n \times m$
 $n < m$ $\underline{x} = \underline{A}^H (\underline{A} \underline{A}^H)^{-1} \underline{b}$: minimum norm solution)

* The matrix $\underline{A}^H (\underline{A} \underline{A}^H)^{-1}$ is called the "pseudoinverse"

* Triangular matrix (upper or lower)

$$\det(\underline{A}) = \prod_{i=1}^n a_{ii}$$

* Symmetric matrix $\underline{A} = \underline{A}^T$

* Toeplitz matrix: All elements along each of the diagonals have the same value

}

etc.

* $\underline{A}: n \times n$ is orthogonal if $\underline{A}^T \underline{A} = \underline{I} \Rightarrow \underline{A}^{-1} = \underline{A}^T$

Thus, if $\underline{A} = [\underline{a}_1, \underline{a}_2 \dots \underline{a}_n]$ then $\underline{a}_i^T \underline{a}_j = \delta(i-j)$

* $\underline{A}: n \times n$ is unitary $\underline{A}^{-1} = \underline{A}^H$
(orthogonal & complex)

Quadratic form of a real symmetric or Hermitian matrix $A: n \times n$

$$Q_A(\underline{x}) = \underline{x}^T \underline{A} \underline{x} = \sum_{i,j=1}^n x_i a_{ij} x_j$$

$$Q_A(\underline{x}) = \underline{x}^H \underline{A} \underline{x} = \sum_{i,j=1}^n x_i^* a_{ij} x_j$$

* If $Q_A(\underline{x}) > 0$ then \underline{A} is positive definite
($\underline{A} > 0$)
for all $\underline{x} \neq 0$

If $Q_A(\underline{x}) \geq 0$ then \underline{A} is positive semidefinite

* Eigenvalues & Eigenvectors

$$\text{Solve: } \underline{A} \underline{v} = \lambda \underline{v} \Rightarrow (\underline{A} - \lambda \underline{I}) \underline{v} = 0$$

non-trivial solution: $\det(\underline{A} - \lambda \underline{I}) = 0$

Eigenvalues: $\lambda_1, \lambda_2, \dots, \lambda_n$

Eigenvectors: $\underline{v}_1, \underline{v}_2, \dots, \underline{v}_n$

* $\underline{A} : n \times n$ singular matrix $\rightarrow \underline{A}\underline{v} = 0$

and $\lambda = 0$ is an eigenvalue of \underline{A}

* Eigenvectors corr. to distinct eigenvalues are linearly independent

* The eigenvalues of a Hermitian matrix are real

* \underline{A} Hermitian and positive definite \Leftrightarrow all $\lambda_i > 0$

* $\det(\underline{A}) = \prod_{i=1}^n \lambda_i$, $\text{tr}(\underline{A}) = \sum_{i=1}^n \lambda_i$

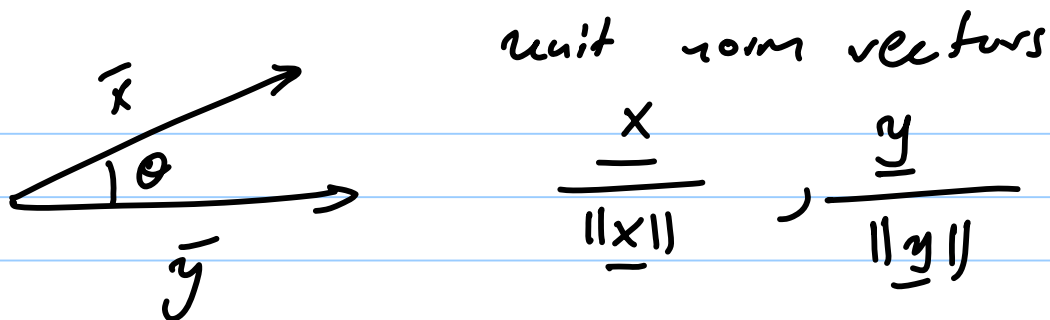
* \underline{A} Hermitian \Leftrightarrow if $\lambda_i \neq \lambda_j$ then $\langle \underline{v}_i, \underline{v}_j \rangle = 0$

* Eigenvalue decomposition:

Let $\underline{\Lambda} = \text{diag.}\{\lambda_1, \lambda_2, \dots, \lambda_n\}$

$\underline{V} = [\underline{v}_1, \underline{v}_2, \dots, \underline{v}_n]$

Then, $\underline{A} = \underline{V} \underline{\Lambda} \underline{V}^H = \sum_{i=1}^n \lambda_i \underbrace{\underline{v}_i \underline{v}_i^H}_{\text{Rank 1 matrix}}$



→ $\underline{x}, \underline{y}$: orthogonal , $\langle \underline{x}, \underline{y} \rangle = 0$

→ $\underline{x}, \underline{y}$: linearly independent if
 $a_1 \underline{x} + a_2 \underline{y} = 0 \rightarrow a_1 = a_2 = 0$

(in this case $\underline{x}, \underline{y}$ can be seen as the generating vectors of a 2-D space where each vector can be obtained by linear combination of \underline{x} and \underline{y}).

Let $x[n] \xrightarrow{\text{FIR, } h[n]} y[n]$

$$y[n] = \sum_{i=0}^{N-1} h[i] x[n-i] = \underline{h}^T \underline{x}[n]$$

where: $\underline{h}^T = [h_1, \dots, h_N]$, $\underline{x}[n] = [x[n], \dots, x[n-N+1]]$

Optimization Theory / Gradient of a scalar

Note Title

02/09/2005

* Let $f(x)$ be a scalar function of a real variable x . Then,

if $f(x)$ is differentiable local & global minima satisfy the conditions

$$\frac{df(x)}{dx} = 0 \quad \text{and} \quad \frac{d^2 f(x)}{dx^2} > 0$$

if $f(x)$ convex: global minimum.



* Let $f(z)$ be a scalar function of a complex variable z . Then,

if $f(z)$ is differentiable proceed as before

However, in most practical situations $f(z)$ may not be differentiable, being a function of both z and z^* .

$$\text{Ex: } f(z) = |z|^2 = z z^*$$

Solution:

1. Express $f(z) = f(x+jy)$ in terms of real and imaginary parts. Then, minimize w.r.t. x and y
or
2. Treat z and z^* as independent variables
Then, minimize with r.t. both z and z^* .

Ex: to find global & local minima of $|z|^2$

$$\text{take } \frac{d}{dz} |z|^2 = z^* = 0, \quad \frac{d}{dz^*} |z|^2 = z = 0$$

Note: If $f(z)$ is real function of z and z^ it is sufficient to minimize w.r.t. either z or z^* only. (as in above example)

*Note: The justification of the solution 2 is basically based on the observation that the obtained results from such a treatment provide meaningful solutions !!

* Now let $f(\underline{x})$ be a scalar function of a vector \underline{x}

Then,

$$\nabla_{\underline{x}} f(\underline{x}) = \frac{d}{d\underline{x}} f(\underline{x}) = \begin{bmatrix} \frac{\partial}{\partial x_1} f(\underline{x}) \\ \vdots \\ \frac{\partial}{\partial x_n} f(\underline{x}) \end{bmatrix}, \quad \underline{x} = \begin{bmatrix} x_1 \\ x_2 \\ \vdots \\ x_n \end{bmatrix}$$

\underline{x} : minimum or maximum
(necessary condition) : $\nabla_{\underline{x}} f(\underline{x}) = 0$

\underline{x} : minimum : Hessian $\underline{H}_{\underline{x}} > 0$ (positive definite)

where: $\{\underline{H}_{\underline{x}}\}(i,j) = \frac{\partial^2 f(\underline{x})}{\partial x_i \partial x_j}$

* Similarly, if $f(\underline{z})$ is a scalar function of a complex vector \underline{z} , treat \underline{z} and \underline{z}^* as independent and proceed as before

Note: If $f(\underline{z}) = f(\underline{z}, \underline{z}^*)$ is real the stationary points (Max or min) are the solutions of the equation $\nabla_{\underline{z}^*} f(\underline{z}, \underline{z}) = 0$

Example:

Let $\underline{z} = [z_1, \dots, z_n]$ be a complex vector

\underline{R} : be a positive definite Hermitian matrix

\underline{a} : be a given complex vector

find \underline{z} that minimizes $\underline{z}^H \underline{R} \underline{z}$ s.t.c. $\underline{z}^H \underline{a} = 1$

Solution:

 (possible approach)

Use the Lagrange multiplier λ and minimize the unconstrained function:

real scalar $\rightarrow Q(\underline{z}, \lambda) = \frac{1}{2} \underline{z}^H \underline{R} \underline{z} + \lambda (1 - \underline{z}^H \underline{a})$

1st order: $\nabla_{\underline{z}^*} Q(\underline{z}, \lambda) = \underline{R} \underline{z} - \lambda \underline{a} = 0$ (1)

$$\Rightarrow \underline{z} = \lambda \underline{R}^{-1} \underline{a}$$

Also, $\frac{\partial Q(\underline{z}, \lambda)}{\partial \lambda} = 1 - \underline{z}^H \underline{a} = 0$ (2)

(1) \rightarrow (2) : $\lambda = \frac{1}{\underline{a}^H \underline{R}^{-1} \underline{a}}$ (3)

(3) \rightarrow (1) : $\underline{z} = \frac{\underline{R}^{-1} \underline{a}}{\underline{a}^H \underline{R}^{-1} \underline{a}}$ (4)

Q : scalar, $\underline{a}, \underline{b}$: real vectors, \underline{B} : real matrix $(N \times N)$

$$\begin{cases} \nabla_{\underline{a}} (\underline{b}^T \underline{a}) = \nabla_{\underline{a}} (\underline{a}^T \underline{b}) = \underline{b} \\ \nabla_{\underline{a}} (\underline{a}^T \underline{B} \underline{a}) = (\underline{B} + \underline{B}^T) \underline{a} \end{cases}$$

Q : scalar, $\underline{a}, \underline{b}$: complex vectors, \underline{B} : complex matrix

$$\begin{cases} \nabla_{\underline{a}} Q = \frac{1}{2} (\nabla_{\underline{a}_r} Q - j \nabla_{\underline{a}_i} Q) \\ \nabla_{\underline{a}^*} Q = \frac{1}{2} (\nabla_{\underline{a}_r} Q + j \nabla_{\underline{a}_i} Q) \\ \nabla_{\underline{a}} (\underline{b}^H \underline{a}) = \underline{b} \\ \nabla_{\underline{a}^*} (\underline{a}^H \underline{b}) = \underline{b} \\ \nabla_{\underline{a}} (\underline{a}^H \underline{B} \underline{a}) = \underline{B}^T \underline{a}^* \\ \nabla_{\underline{a}^*} (\underline{a}^H \underline{B} \underline{a}) = \underline{B} \underline{a} \end{cases}$$

