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# Signal Processing

## Collection of Formulas

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# 1 Basic Relationships and Concepts

## 1.1 Standard Signals

### 1.1.1 Continuous Functions

$T$  is the period time for a periodic function  $f(t)$ , where  $t$  [s] is the time variable  
Angular frequency  $\Omega = \frac{2\pi}{T} = 2\pi F$  [rad/s],  $F$  [Hz] is the frequency.

a) Impulse Function 
$$\delta(t) = \begin{cases} +\infty & t = 0 \\ 0 & t \neq 0 \end{cases}$$

$$\int_{-\infty}^{\infty} \delta(t) dt = 1$$

$$x(t)\delta(t-a) = x(t-a)\delta(t)$$

$$\delta\left(\frac{1}{a}\right) = a\delta(t)$$

$$\int_{-\infty}^{\infty} x(t)\delta(t-a)dt = x(a)$$

b) Step Function

$$u(t) = \begin{cases} 1 & t \geq 0 \\ 0 & t < 0 \end{cases}$$

c) Ramp Function

$$r(t) = \begin{cases} t & t \geq 0 \\ 0 & t < 0 \end{cases}$$

d) Rectangular Pulse

$$p(t) = \begin{cases} 1 & |t| < \frac{1}{2} \\ 0 & |t| > \frac{1}{2} \end{cases}$$

e) Sinc Function

$$\text{sinc } t = \frac{\sin \pi t}{\pi t}$$

f) Periodic Function

$$f(t) = f(t+T) \text{ where } T \neq 0$$

g) Periodic Sinc Function

$$\text{diric}(t, T) = \frac{\sin\left(\frac{Tt}{2}\right)}{T \sin\left(\frac{t}{2}\right)}$$

h) Complex Periodic Signal

$$e^{st} = e^{\sigma t} e^{j\Omega t}$$

i) Complex undamped per. signal

$$e^{j\Omega t} = \cos \Omega t + j \sin \Omega t$$

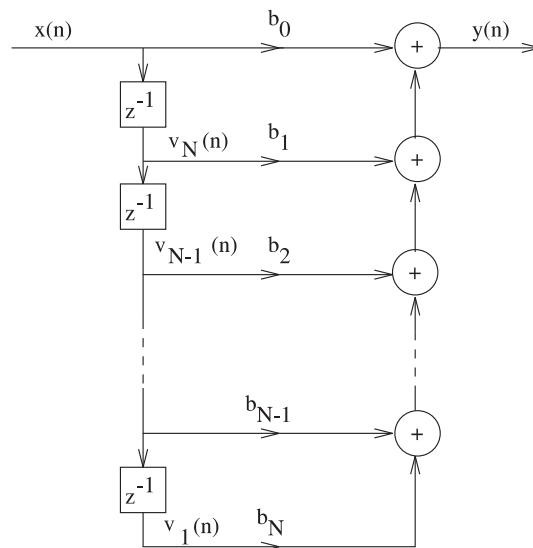
### 1.1.2 Discrete Time Functions

$N$  is the period time (integer) for a periodic function  $f(n)$ , where  $n$  is a discrete time index (integer). Normalized angular frequency  $\omega = \frac{2\pi}{N} = 2\pi f$  [rad],  $f$  is normalized frequency.

- a) Unit Pulse  $\delta(n) = \begin{cases} 1 & n = 0 \\ 0 & n \neq 0 \end{cases}$
- b) Unit Step  $u(n) = \begin{cases} 1 & n \geq 0 \\ 0 & n < 0 \end{cases}$
- c) Ramp  $r(n) = \begin{cases} n & n \geq 0 \\ 0 & n < 0 \end{cases}$
- d) Rectangular Pulse  $p(n) = \begin{cases} 1 & |n| < n_1 \\ 0 & |n| > -n_1 \end{cases}$
- e) Sinc Function  $\text{sind}_N(n) = \frac{\sin\left(\frac{Nn}{2}\right)}{N \sin\left(\frac{n}{2}\right)}$
- f) Periodic Signal  $f(n) = f(n + N)$  where  $N \neq 0$
- g) Complex periodic signal  $e^{sn} = e^{\sigma n} e^{j\omega n}$
- h) Complex undamped per. signal  $e^{j\omega n} = \cos \omega n + j \sin \omega n$

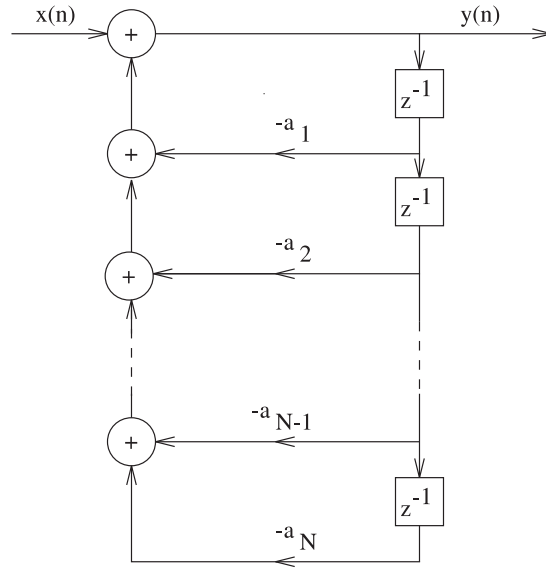
## 1.2 SISO systems (Single input, single output) Discrete Time Systems

### 1.2.1 FIR Filters



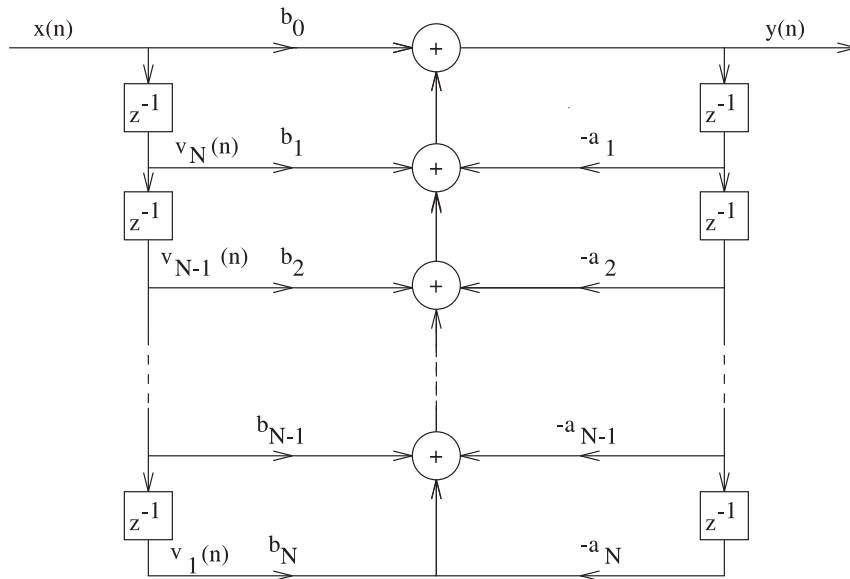
$$y(n) = \sum_{k=0}^M b_k x(n-k)$$

### 1.2.2 IIR Filters



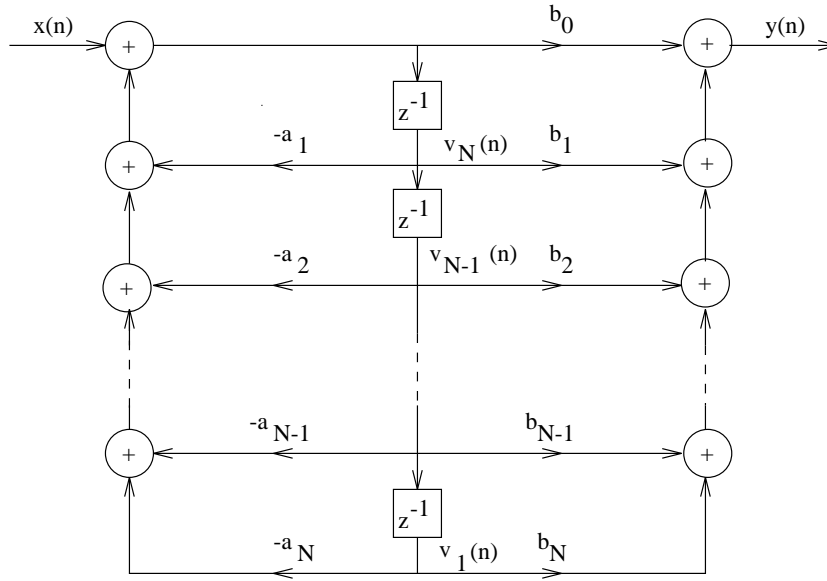
$$y(n) = - \sum_{k=1}^N a_k y(n - k)$$

### 1.2.3 IIR Canonical form - Direct Form I



$$y(n) = - \sum_{k=1}^N a_k y(n - k) + \sum_{k=0}^M b_k x(n - k)$$

### 1.2.4 IIR Canonical form - Direkt Form II



$$y(n) = - \sum_{k=1}^N a_k y(n-k) + \sum_{k=0}^M b_k x(n-k)$$

## 1.3 Some Methods of Calculation

### 1.3.1 Convolution

$$y(n) = h * x = \sum_{k=-\infty}^{\infty} h(k)x(n-k) = \sum_{k=-\infty}^{\infty} h(n-k)x(k)$$

### 1.3.2 State-Space Model of Difference Equation

If

$$y(n) = - \sum_{k=1}^N a_k y(n-k) + \sum_{k=0}^M b_k x(n-k)$$

the State-Space Model is given by

$$\begin{cases} \mathbf{v}(n+1) = \mathbf{F}\mathbf{v}(n) + \mathbf{q} \cdot x(n) \\ y(n) = \mathbf{g}^T \mathbf{v}(n) + d \cdot x(n) \end{cases}$$

where

$$\mathbf{F} = \begin{pmatrix} 0 & 1 & \dots & 0 & 0 \\ \vdots & \vdots & & \vdots & \vdots \\ 0 & 0 & & 0 & 1 \\ -a_k & -a_{k-1} & \dots & -a_2 & -a_1 \end{pmatrix} ; \quad \mathbf{q} = \begin{pmatrix} 0 \\ 0 \\ \vdots \\ 1 \end{pmatrix}$$

$$\mathbf{g}^T = (b_k, \dots, b_2, b_1) - b_0(a_k, \dots, a_2, a_1) ; \quad d = b_0$$

### 1.3.3 State-Space Equation

a) Direct Solution

$$y(n) = \mathbf{g}^T \cdot \mathbf{F}^n \mathbf{v}(0) + \sum_{k=0}^{n-1} \mathbf{g}^T \cdot \mathbf{F}^{n-1-k} \mathbf{q} x(k) u(n-1) + dx(n)$$

b) Impulse Function

$$h(n) = \mathbf{g}^T \cdot \mathbf{F}^{n-1} \mathbf{q} u(n-1) + d\delta(n)$$

c) System Function

$$\mathcal{H}(z) = \mathbf{g}^T [z\mathbf{I} - \mathbf{F}]^{-1} \mathbf{q} + d$$

### 1.3.4 System Function

$$\mathcal{H}(z) = \frac{\mathcal{Y}(z)}{\mathcal{X}(z)} = \frac{b_0 + b_1 z^{-1} + \dots + b_M z^{-M}}{1 + a_1 z^{-1} + \dots + a_N z^{-N}}$$



## 1.4 Analogue Sinusoidal Signal through Linear, Causal Filter

### 1.4.1 Complex, Non-causal Input Signal

$$x(t) = e^{j\Omega_0 t} = (\cos(\Omega_0 t) + j \sin(\Omega_0 t)) \quad -\infty < t < \infty$$

$$y(t) = \int_{\tau=0}^{\infty} h(\tau)x(t-\tau)d\tau = \int_{\tau=0}^{\infty} h(\tau)e^{j\Omega_0(t-\tau)}d\tau = \underbrace{H(s)|_{s=j\Omega_0}}_{\text{stationary}} e^{j\Omega_0 t}$$

### 1.4.2 Complex, Causal Input Signal

$$x(t) = e^{j\Omega_0 t}u(t) = (\cos(\Omega_0 t) + j \sin(\Omega_0 t)) u(t); \quad X(s) = \frac{1}{s - j\Omega_0}$$

$$Y(s) = H(s)X(s) = \frac{T(s)}{N(s)} \frac{1}{s - j\Omega_0} = \underbrace{\frac{T_1(s)}{N(s)}}_{\text{transient}} + \underbrace{H(s)|_{s=j\Omega_0} \frac{1}{s - j\Omega_0}}_{\text{stationary}}$$

$$y(t) = \text{transient} + \underbrace{H(s)|_{s=j\Omega_0} e^{j\Omega_0 t}}_{\text{stationary}}$$

### 1.4.3 Real, Non-causal Input Signal

$$x(t) = \text{Re}\{e^{j\Omega_0 t}\} = \cos(\Omega_0 t) \quad -\infty < t < \infty$$

$$\begin{aligned} y(t) &= \int_{\tau=0}^{\infty} h(\tau)x(t-\tau)d\tau = \int_{\tau=0}^{\infty} h(\tau)\frac{1}{2}(e^{j\Omega_0(t-\tau)} + e^{-j\Omega_0(t-\tau)})d\tau = \\ &= \underbrace{|H(s)|_{s=j\Omega_0} \cos(\Omega_0 t + \arg\{H(s)|_{s=j\Omega_0}\})}_{\text{stationary}} \end{aligned}$$

### 1.4.4 Real, Causal Input Signal

$$x(t) = \text{Re}\{e^{j\Omega_0 t}\} u(t) = \cos(\Omega_0 t) u(t); \quad X(s) = \frac{s}{s^2 + \Omega_0^2}$$

$$Y(s) = H(s)X(s) = \frac{T(s)}{N(s)} \frac{s}{s^2 + \Omega_0^2} = \underbrace{\frac{T_1(s)}{N(s)}}_{\text{transient}} + \underbrace{\frac{C_1 s + C_0}{s^2 + \Omega_0^2}}_{\text{stationary}}$$

$$H(s)|_{s=j\Omega_0} = A e^{j\theta}; \quad C_1 = A \cos(\theta); \quad C_0 = -A\Omega_0 \sin \theta$$

$$\begin{aligned}
y(t) &= \text{transient} + \underbrace{C_1 \cos(\Omega_0 t) + \frac{C_0}{\Omega_0} \sin(\Omega_0 t)}_{\text{stationary}} = \\
&= \text{transient} + \underbrace{|H(s)|_{s=j\Omega_0} \cos(\Omega_0 t + \text{arg}\{H(s)|_{s=j\Omega_0}\})}_{\text{stationary}}
\end{aligned}$$

## 1.5 Discrete Time Sinusoidal Signal through Linear, Causal Filter

### 1.5.1 Complex, Non-Causal Input Signal

$$x(n) = e^{j\omega_0 n} = (\cos(\omega_0 n) + j \sin(\omega_0 n)) \quad -\infty < n < \infty$$

$$y(n) = \sum_{k=0}^{\infty} h(k)x(n-k) = \sum_{k=0}^{\infty} h(k)e^{j\omega_0(n-k)} = \underbrace{H(z)|_{z=e^{j\omega_0}}}_{\text{stationary}} e^{j\omega_0 n}$$

### 1.5.2 Complex, Causal Input Signal

$$x(n) = e^{j\omega_0 n} u(n) = (\cos(\omega_0 n) + j \sin(\omega_0 n)) u(n); \quad X(z) = \frac{1}{1 - e^{j\omega_0} z^{-1}}$$

$$Y(z) = H(z)X(z) = \frac{T(z)}{N(z)} \frac{1}{1 - e^{j\omega_0} z^{-1}} = \underbrace{\frac{T_1(z)}{N(z)}}_{\text{transient}} + \underbrace{H(z)|_{z=e^{j\omega_0}} \frac{1}{1 - e^{j\omega_0} z^{-1}}}_{\text{stationary}}$$

$$y(n) = \text{transient} + \underbrace{H(z)|_{z=e^{j\omega_0}} e^{j\omega_0 n}}_{\text{stationary}}$$

### 1.5.3 Real, Non-Causal Input Signal

$$x(n) = \text{Re}\{e^{j\omega_0 n}\} = \cos(\omega_0 n) \quad -\infty < n < \infty$$

$$\begin{aligned} y(n) &= \sum_{k=0}^{\infty} h(k)x(n-k) = \sum_{k=0}^{\infty} h(k) \frac{1}{2} (e^{j\omega_0(n-k)} + e^{-j\omega_0(n-k)}) = \\ &= \underbrace{|H(z)|_{z=e^{j\omega_0}} \cos(\omega_0 n + \arg\{H(z)|_{z=e^{j\omega_0}}\})}_{\text{stationary}} \end{aligned}$$

### 1.5.4 Real, Causal Input Signal

$$x(n) = \text{Re}\{e^{j\omega_0 n}\} u(n) = \cos(\omega_0 n) u(n); \quad X(z) = \frac{1 - \cos \omega_0 z^{-1}}{1 - 2 \cos \omega_0 z^{-1} + z^{-2}}$$

$$Y(z) = H(z)X(z) = \frac{T(z)}{N(z)} \frac{1 - \cos \omega_0 z^{-1}}{1 - 2 \cos \omega_0 z^{-1} + z^{-2}} = \underbrace{\frac{T_1(z)}{N(z)}}_{\text{transient}} + \underbrace{\frac{C_0 + C_1 z^{-1}}{1 - 2 \cos \omega_0 z^{-1} + z^{-2}}}_{\text{stationary}}$$

$$H(z)|_{z=e^{j\omega_0}} = A e^{j\theta}; \quad C_0 = A \cos(\theta); \quad C_1 = -A(\sin \omega_0 \sin \theta + \cos \omega_0 \cos \theta)$$

$$\begin{aligned}
 y(n) &= \text{transient} + \underbrace{C_0 \cos(\omega_0 n) + \frac{C_1 + C_0 \cos(\omega_0)}{\sin(\omega_0)} \sin(\omega_0 n)}_{\text{stationary}} = \\
 &= \text{transient} + \underbrace{|H(z)|_{z=e^{j\omega_0}} \cos(\omega_0 n + \arg\{H(z)|_{z=e^{j\omega_0}}\})}_{\text{stationary}}
 \end{aligned}$$

## 1.6 Fourier Series Expansion

For table see Appendix

### 1.6.1 Continuous Time

A periodic function with period  $T_0$ , i.e.  $f(t) = f(t - T_0)$ , can be expressed in the form of a series expansion according to

$$f(t) = \sum_{k=-\infty}^{\infty} c_k e^{j2\pi k F_0 t}$$

where

$$c_k = \frac{1}{T_0} \int_{T_0} f(t) e^{-j2\pi k F_0 t} dt ; F_0 = \frac{1}{T_0}$$

If  $f(t)$  is real this can also be expressed as

$$\begin{aligned} f(t) &= c_0 + 2 \sum_{k=1}^{\infty} |c_k| \cos(2\pi k F_0 t + \theta_k) = \\ &= a_0 + \sum_{k=1}^{\infty} a_k \cos 2\pi k F_0 t - b_k \sin 2\pi k F_0 t \end{aligned}$$

where

$$\begin{aligned} a_0 &= c_0 = \frac{1}{T_0} \int_{T_0} f(t) dt \\ a_k &= 2|c_k| \cos \theta_k = \frac{2}{T_0} \int_{T_0} f(t) \cos(2\pi k F_0 t) dt \\ b_k &= 2|c_k| \sin \theta_k = \frac{-2}{T_0} \int_{T_0} f(t) \sin(2\pi k F_0 t) dt \end{aligned}$$

The power is given by (Parseval's Relation)

$$P = \frac{1}{T_0} \int_{T_0} |f(t)|^2 dt = \sum_{k=-\infty}^{\infty} |c_k|^2$$

In addition, for real signals

$$P = c_0^2 + 2 \sum_{k=1}^{\infty} |c_k|^2 = a_0^2 + \frac{1}{2} \sum_{k=1}^{\infty} (a_k^2 + b_k^2)$$

### 1.6.2 Discrete Time

A periodic function with the period  $N$ , i.e.  $f(n) = f(n - N)$ , can be expressed as a series expansion according to

$$f(n) = \sum_{k=0}^{N-1} c_k e^{j2\pi k n/N}$$

where

$$c_k = \frac{1}{N} \sum_{n=0}^{N-1} f(n) e^{-j2\pi k n/N}, \quad k = 0, \dots, N-1$$

The series expansion is often written using DTFS (Discrete-Time Fourier Series). If  $f(n)$  is real this can also be expressed as

$$\begin{aligned} f(n) &= c_0 + 2 \sum_{k=1}^L |c_k| \cos\left(2\pi \frac{kn}{N} + \theta_k\right) = \\ &= a_0 + \sum_{k=1}^L \left( a_k \cos\left(2\pi \frac{kn}{N}\right) - b_k \sin\left(2\pi \frac{kn}{N}\right) \right) \end{aligned}$$

where

$$\begin{aligned} a_0 &= c_0 \\ a_k &= 2|c_k| \cos(\theta_k) \\ b_k &= 2|c_k| \sin(\theta_k) \\ L &= \begin{cases} \frac{N}{2} & \text{if } N \text{ even} \\ \frac{N-1}{2} & \text{if } N \text{ odd} \end{cases} \end{aligned}$$

The power is given by

$$P = \frac{1}{N} \sum_{n=0}^{N-1} |f(n)|^2 = \sum_{k=0}^{N-1} |c_k|^2$$

and the energy over one period is given by

$$E_N = \sum_{n=0}^{N-1} |f(n)|^2 = N \sum_{k=0}^{N-1} |c_k|^2$$

## 1.7 Fourier Transformer

Continuous time signal:

$$\begin{cases} X_a(F) &= \int_{-\infty}^{\infty} x_a(t) e^{-j2\pi Ft} dt \\ x_a(t) &= \int_{-\infty}^{\infty} X_a(F) e^{j2\pi Ft} dF \end{cases}$$

Discrete time signal:

$$\begin{cases} X(f) &= \sum_{n=-\infty}^{\infty} x(n) e^{-j2\pi fn} \\ x(n) &= \int_{-1/2}^{1/2} X(f) e^{j2\pi fn} df \end{cases}$$

## 1.8 Discrete Fourier Transform (DFT)

### 1.8.1 Definition

$$X_k = DFT(x_n) = \sum_{n=0}^{N-1} x_n e^{-j2\pi nk/N} \quad k = 0, 1, \dots, N-1 \quad \text{Transform}$$

$$x_n = IDFT(X_k) = \frac{1}{N} \sum_{k=0}^{N-1} X_k e^{j2\pi nk/N} \quad n = 0, 1, \dots, N-1 \quad \text{Inversion}$$

Note:

$$\sum_{n=0}^{N-1} e^{j2\pi \frac{k-k_0}{N} \cdot n} = N \cdot \delta(k - k_0, (\text{modulo } N))$$

### 1.8.2 Circular Convolution

$$x_n \circledast y_n = \sum_{\ell=0}^{N-1} x_\ell y_{n-\ell} \xleftrightarrow{\text{DFT}} X_k Y_k \quad \text{Circular convolution}$$

where  $\circledast$  denotes circular convolution. This means that the sequences  $x_n$  and  $y_n$  should be repeated periodically before the summation. I.e. outside the interval  $n = 0, 1, \dots, N-1$ ,  $x_{n-\ell N} = x_n$  and  $y_{n-\ell N} = y_n$  ( $\ell = \text{integer}$ ) In other words, the index is calculated modulo  $N$ . Circular convolution is also denoted  $x(n) * y(n)$ .

### 1.8.3 Non-Circular Convolution using DFT

If  $x(n) = 0$  for  $n \notin [0, L-1]$  and  $y(n) = 0$  for  $n \notin [0, M-1]$  then  $x * y = 0$  for  $n \notin [0, N-1]$  where  $N \geq L + M - 1$ .

The convolution can be calculated as

$$x * y = \begin{cases} x \circledast y = IDFT(X_k Y_k) & n = 0, 1, \dots, N-1 \\ 0 & \text{Else} \end{cases}$$

where

$$\begin{aligned} X_k &= DFT(x(n)) \\ Y_k &= DFT(y(n)) \end{aligned}$$

### 1.8.4 Relation to the Fourier Transform $X(f)$ :

$$X(k/N) = X_k = DFT(x(n)) \text{ if } x(n) = 0 \text{ for } n \notin [0, N-1]$$

$$X(k/N) = X_k = DFT(x_p(n)) \text{ generally } x(n) \text{ where } x_p(n) = \sum_{\ell=-\infty}^{\infty} x(n - \ell N)$$

### 1.8.5 Relation to Fourier Series

$$X\left(\frac{k}{N}\right) = X_k = DFT(x(n)) = N \cdot c_k$$

if

$$x(n) = x_p(n), \quad 0 \leq n \leq N-1$$

where

$$x_p(n) = \sum_{k=0}^{N-1} c_k e^{j2\pi \frac{nk}{N}} \quad -\infty < n < \infty$$

and

$$c_k = \frac{1}{N} \sum_{n=0}^{N-1} x_p(n) e^{-j2\pi \frac{nk}{N}} \quad k = 0, 1, \dots, N-1$$

### 1.8.6 Parseval's Theorem

$$\sum_{n=0}^{N-1} x(n)y^*(n) = \frac{1}{N} \sum_{k=0}^{N-1} X_k Y^*(k)$$



## 1.9 Some Window Functions and their Fourier Transforms

- i) The window functions are centered around the origin (odd filter length  $M$ ), i.e. the functions are not equal to zero only for  $-(M-1)/2 \leq n \leq (M-1)/2$

Rectangular window:

$$w_{\text{rect}}(n) = 1$$

$$W_{\text{rect}}(f) = M \cdot \frac{\sin(\pi f M)}{M \sin(\pi f)}$$

Bartlett window (Triangular window):

$$w(n) = 1 - \frac{|n|}{(M-1)/2}$$

$$W(f) = \frac{M}{2} \left( \frac{\sin \frac{\pi f M}{2}}{\frac{M}{2} \sin(\pi f)} \right)^2 \approx \frac{2}{M} W_{\text{rect}}^2 \left( \frac{f}{2} \right) \text{ if } f \text{ is small}$$

Hanning Window:

$$w(n) = 0.5 + 0.5 \cos \left( \frac{2\pi n}{M-1} \right)$$

$$W(f) = 0.5 W_{\text{rect}}(f) +$$

$$+ 0.25 W_{\text{rect}} \left( f - \frac{1}{M-1} \right) +$$

$$+ 0.25 W_{\text{rect}} \left( f + \frac{1}{M-1} \right)$$

Hamming Window:

$$w(n) = 0.54 + 0.46 \cos \left( \frac{2\pi n}{M-1} \right)$$

$$W(f) = 0.54 W_{\text{rect}}(f) +$$

$$+ 0.23 W_{\text{rect}} \left( f - \frac{1}{M-1} \right) +$$

$$+ 0.23 W_{\text{rect}} \left( f + \frac{1}{M-1} \right)$$

Blackman window:

$$w(n) = 0.42 + 0.5 \cos \frac{2\pi n}{M-1} + 0.08 \cos \frac{4\pi n}{M-1}$$

$$W(f) = 0.42 W_{\text{rect}}(f) +$$

$$+ 0.25 W_{\text{rect}} \left( f - \frac{1}{M-1} \right) +$$

$$\begin{aligned}
&+0.25 W_{\text{rect}} \left( f + \frac{1}{M-1} \right) + \\
&+0.04 W_{\text{rect}} \left( f - \frac{2}{M-1} \right) + \\
&+0.04 W_{\text{rect}} \left( f + \frac{2}{M-1} \right)
\end{aligned}$$

Kaiser window:

$$w(n) = \frac{I_0 \left( \beta \sqrt{1 - [2n/(M-1)]^2} \right)}{I_0(\beta)}$$

$$I_0(x) = 1 + \sum_{k=1}^L \left[ \frac{(x/2)^k}{k!} \right]^2$$

## 2 Sampling Analogue Signals

### 2.1 Sampling and Reconstruction

#### Sampling Theorem

If  $x_a(t)$  is bandlimited, i.e.  $X_a(F) = 0$  for  $|F| \geq 1/2T$  then

$$x_a(t) = \sum_{n=-\infty}^{\infty} x(n) \frac{\sin \frac{\pi}{T} (t - nT)}{\frac{\pi}{T} (t - nT)}$$

Sampling Frequency  $F_s = 1/T$ .

#### Sampling

$$\begin{aligned} x(n) &= x_a(nT); \quad T = \frac{1}{F_s} \\ X(f) &= X\left(\frac{F}{F_s}\right) = F_s \sum_{k=-\infty}^{\infty} X_a(F - kF_s) \\ \Gamma(f) &= \Gamma\left(\frac{F}{F_s}\right) = F_s \sum_{k=-\infty}^{\infty} \Gamma_a(F - kF_s) \end{aligned}$$

Reconstruction (ideal)

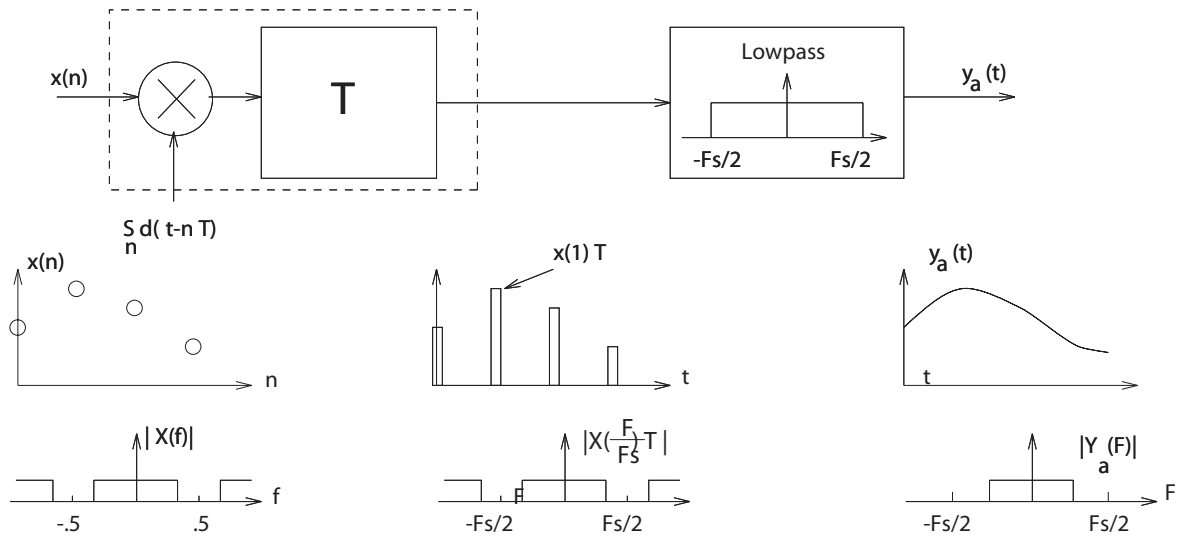
$$\begin{aligned} x_a(t) &= \sum_{n=-\infty}^{\infty} x(n) \frac{\sin \frac{\pi}{T} (t - nT)}{\frac{\pi}{T} (t - nT)} \\ X_a(F) &= \frac{1}{F_s} X\left(\frac{F}{F_s}\right) \quad |F| \leq \frac{F_s}{2} \\ \Gamma_a(F) &= \frac{1}{F_s} \Gamma\left(\frac{F}{F_s}\right) \quad |F| \leq \frac{F_s}{2} \end{aligned}$$

Reconstruction using Sample-and-Hold

$$\begin{aligned} X_a(F) &= \frac{1}{F_s} X\left(\frac{F}{F_s}\right) \cdot \frac{\sin(\pi FT)}{\pi FT} e^{-j2\pi F \frac{T}{2}} \cdot H_{LP}(F) \\ \Gamma_a(F) &= \frac{1}{F_s} \Gamma\left(\frac{F}{F_s}\right) \left| \frac{\sin(\pi FT)}{\pi FT} \right|^2 \cdot |H_{LP}(F)|^2 \end{aligned}$$

# Block Scheme describing D/A conversion

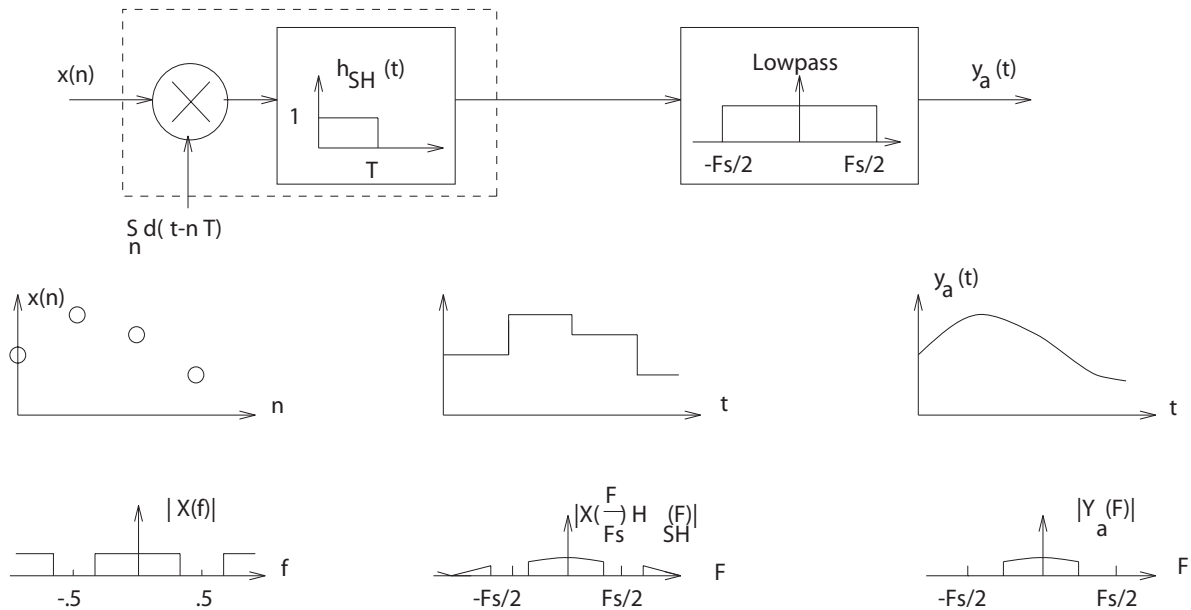
Ideal reconstruction



$$y_a(t) = \sum_{n=-\infty}^{\infty} x(n) \frac{\sin \frac{\pi}{T} (t - nT)}{\frac{\pi}{T} (t - nT)}$$

$$Y_a(F) = \frac{1}{F_s} X\left(\frac{F}{F_s}\right) \quad |F| \leq \frac{F_s}{2}$$

Reconstruction using Sample-and-Hold



$$Y_a(F) = \frac{1}{F_s} X\left(\frac{F}{F_s}\right) \cdot \frac{\sin(\pi FT)}{\pi FT} e^{-j2\pi F \frac{T}{2}} \cdot H_{LP}(F)$$

## 2.2 Measures of Distortion

### 2.2.1 Folding distortion when sampling

Spectrum after anti-aliasing filter:

$$\Gamma_{in}(F)$$

Folding distortion:

$$D_A = 2 \cdot \int_{F_s - F_p}^{\infty} \Gamma_{in}(F) dF$$

Effective signal power:

$$D_s = 2 \int_0^{F_p} \Gamma_{in}(F) dF$$

where  $0 \leq F_p \leq F_s/2$

Signal distortion relationship:

$$\text{A: } SDR_A = \frac{D_s}{D_A} = \frac{\int_0^{F_p} \Gamma_{in}(F) dF}{\int_{F_s - F_p}^{\infty} \Gamma_{in}(F) dF}$$

$$\text{B: } SDR_A^0 = \min_{|F| \leq F_p} \frac{\Gamma_{in}(F)}{\Gamma_{in}(F_s - F)}$$

If the spectrum is monotonously decreasing

$$SDR_A^0 = \frac{\Gamma_{in}(F_p)}{\Gamma_{in}(F_s - F_p)}$$

### 2.2.2 Periodic Distortion when Reconstructing

Periodic Distortion:

$$D_P = 2 \cdot \int_{F_s/2}^{\infty} \Gamma_{ut}(F) dF$$

Effective signal power:

$$D_S = 2 \cdot \int_0^{F_s/2} \Gamma_{ut}(F) dF$$

Signal Distortion Relationship:

$$\text{A: } SDR_P = \frac{D_S}{D_P} = \frac{\int_0^{F_s/2} \Gamma_{ut}(F) dF}{\int_{F_s/2}^{\infty} \Gamma_{ut}(F) dF}$$

$$\text{B: } SDR_P^0 = \min_{|F| < F_s/2} \frac{\Gamma_{ut}(F)}{\Gamma_{ut}(F_s - F)}$$

A good estimate is often given by

$$SDR_P^0 = \frac{\Gamma_{ut}(F_p)}{\Gamma_{ut}(F_s - F_p)}$$

where  $F_p$  is the highest frequency component in the sampled signal.

## 2.3 Quantizing Distortion

$$D_Q \simeq \frac{\Delta^2}{12} \text{ linear quantizing, } \Delta \text{ small}$$

$$\text{SQNR} = \frac{\text{Signal Power}}{D_q}$$

Quantizing distortion for sinusoidal signal, maximum dynamical range used,  $r$  bits

$$\text{SQNR} = 1.76 + 6 \cdot r [\text{dB}]$$

Quantizing distortion, dynamical range usage expressed in top- and RMS value,  $r$  bits

$$\text{SQNR} = 6 \cdot r + 1.76 - 10^{10} \log \left( \frac{A_{peak}}{A_{RMS} \cdot \sqrt{2}} \right)^2 - 10^{10} \log \left( \frac{V}{A_{peak}} \right)^2$$

where  $[-V, V]$  is the dynamical range of the quantizer.

## 2.4 Decimation and Interpolation

Downsampling a factor  $M$

$$\downarrow M \quad y(n) = \{\dots u(0), u(M), u(2M) \dots\}$$

$$Y(f) = \frac{1}{M} \sum_{i=0}^{M-1} U\left(\frac{f-i}{M}\right)$$

Upsampling a factor  $L$

$$\uparrow L \quad w(n) = \{\dots x(0), \underbrace{0, 0, \dots}_{L-1 \text{ st}}, x(1), \underbrace{0, 0, \dots}_{L-1 \text{ st}}, x(2) \dots\}$$

$$W(f) = X(fL)$$

### 3 Analogue Filters

#### 3.1 Filter Approximations if ideal LP filters

General form of the amplitude function of the approximation

$$|H(\Omega)| = \frac{K}{\sqrt{1 + g_N \left( \left( \frac{\Omega}{\Omega_p} \right)^2 \right)}} \quad \Omega = 2\pi F$$

where

$$g_N \left( \left( \frac{\Omega}{\Omega_p} \right)^2 \right) \begin{cases} \ll 1 & \left| \frac{\Omega}{\Omega_p} \right| < 1 \\ \gg 1 & \left| \frac{\Omega}{\Omega_p} \right| > 1 \end{cases}$$

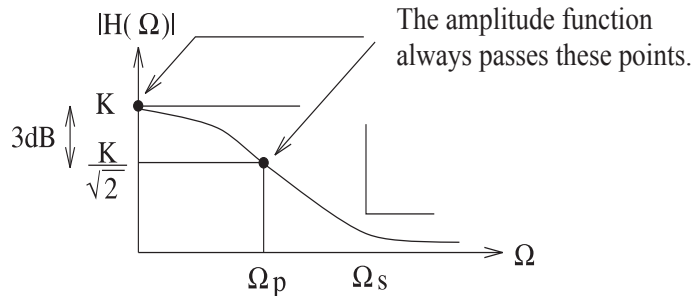
and  $\Omega_p$  is the cut-off frequency of the filter.

Sometimes it is suitable to normalize the angular frequency to  $\Omega_p$ .

In this section this corresponds to setting  $\Omega_p = 1$ .

##### 3.1.1 Butterworth Filters

$$|H(\Omega)| = \frac{K}{\sqrt{1 + \left( \frac{\Omega}{\Omega_p} \right)^{2N}}}$$



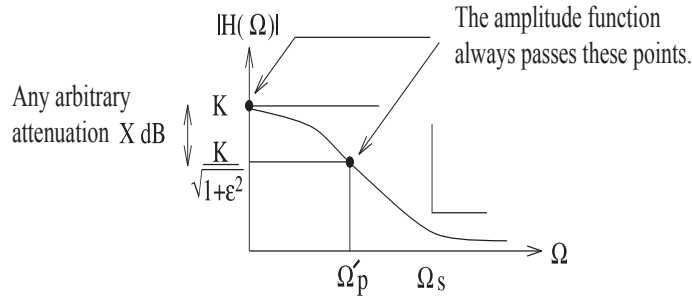
$K$  = The maximum value of the amplitude function.

$K$  = The value of the amplitude function for  $\Omega = 0$ .

Arbitrary attenuation in the pass band (X dB)

$$|H(\Omega)| = \frac{K}{\sqrt{1 + \varepsilon^2 \left( \frac{\Omega}{\Omega_p} \right)^{2N}}}$$

When  $\Omega'_p = \Omega_p$  (3dB) then  $\varepsilon^2 \approx 1$ .



Filter order:

$$N = \frac{\log_{10} \left( \left[ \sqrt{\frac{1}{\delta_2^2} - 1} \right] / \epsilon \right)}{\log_{10} \left( \frac{\Omega_s}{\Omega_p} \right)}$$

$\delta_2$  is the maximum allowed value for the amplitude function at the stop band edge:

$$\delta_2 = |H(\Omega_s)|_{\max}$$

The denominator of the system function is a Butterworth polynomial if  $\Omega_p = 1$ . These polynomials can be found in Table 2.1. For a general  $\Omega_p$

$$\mathcal{H}(s) = \frac{K}{\left(\frac{s}{\Omega_p}\right)^N + a_{N-1} \left(\frac{s}{\Omega_p}\right)^{N-1} + \cdots + a_1 \left(\frac{s}{\Omega_p}\right) + 1}$$

where  $a_1, \dots, a_{N-1}$  can be found in Table 4.1.



**Table 4.1**

Coefficients  $a_\nu$  in Butterworth polynomials  $s^N + a_{N-1}s^{N-1} + \dots + a_1s + 1$

$N$	$a_1$	$a_2$	$a_3$	$a_4$	$a_5$	$a_6$	$a_7$
1							
2	$\sqrt{2}$						
3	2	2					
4	2.613	3.414	2.613				
5	3.236	5.236	5.236	3.236			
6	3.864	7.464	9.141	7.464	3.864		
7	4.494	10.103	14.606	14.606	10.103	4.494	
8	5.126	13.138	21.848	25.691	21.848	13.138	5.126

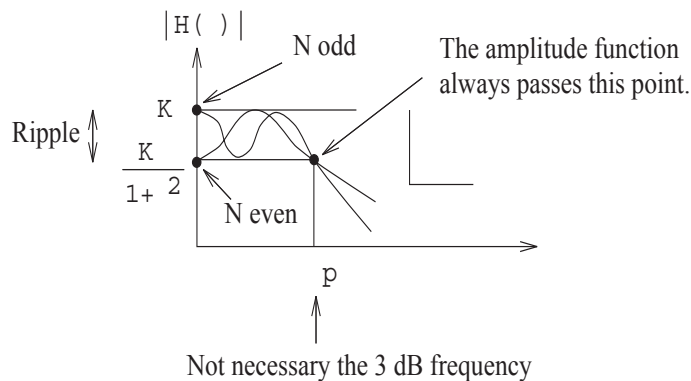
**Table 4.2**

Factorized Butterworth polynomials for  $\Omega_p = 1$ . For  $\Omega_p \neq 1$  let  $s \rightarrow s/\Omega_p$ .

$N$	
1	$(s + 1)$
2	$(s^2 + \sqrt{2}s + 1)$
3	$(s^2 + s + 1)(s + 1)$
4	$(s^2 + 0.76536s + 1)(s^2 + 1.84776s + 1)$
5	$(s + 1)(s^2 + 0.6180s + 1)(s^2 + 1.6180s + 1)$
6	$(s^2 + 0.5176s + 1)(s^2 + \sqrt{2}s + 1)(s^2 + 1.9318s + 1)$
7	$(s + 1)(s^2 + 0.4450s + 1)(s^2 + 1.2465s + 1)(s^2 + 1.8022s + 1)$
8	$(s^2 + 0.3896s + 1)(s^2 + 1.1110s + 1)(s^2 + 1.6630s + 1)(s^2 + 1.9622s + 1)$

### 3.1.2 Chebyshev Filters

$$|H(\Omega)| = \frac{K}{\sqrt{1 + \varepsilon^2 T_N^2\left(\frac{\Omega}{\Omega_p}\right)}}$$



Ripple =  $10 \cdot \log(1 + \varepsilon^2)$  dB.

$K$  = The maximum value of the amplitude function.

$K \neq$  the value of the amplitude function for  $\Omega = 0$  when  $N$  is even.  
 $T_N\left(\frac{\Omega}{\Omega_p}\right)$  is a Chebyshev polynomial. (Also denoted  $C_N\left(\frac{\Omega}{\Omega_p}\right)$ ). These can be found in Table 4.3 for  $\Omega_p = 1$ . For  $\Omega_p \neq 1$  let  $\Omega \rightarrow \frac{\Omega}{\Omega_p}$  in Table 4.3.

System function

$$\mathcal{H}(s) = \frac{K \cdot a_0 \cdot \begin{cases} 1 & N \text{ odd} \\ \frac{1}{\sqrt{1+\varepsilon^2}} & N \text{ even} \end{cases}}{\left(\frac{s}{\Omega_p}\right)^N + a_{N-1} \left(\frac{s}{\Omega_p}\right)^{N-1} + \dots + a_0}$$

where  $\varepsilon, a_0, \dots, a_{N-1}$  can be found in Table 4.4.

The locations of the poles for  $\mathcal{H}(s)$  can be found in Table 4.5 for  $\Omega_p = 1$ . For  $\Omega_p \neq 1$  the pole locations are multiplied with  $\Omega_p$ .

### Table 4.3

Chebyshev polynomials.

$$T_N(\Omega) = \begin{cases} \cos(N \arccos \Omega) & |\Omega| \leq 1 \\ \cosh(N \operatorname{arccosh} \Omega) & |\Omega| \geq 1 \end{cases} \quad \Omega = 2\pi F$$

or

$$T_N(\Omega) = \frac{(\Omega + \sqrt{\Omega^2 - 1})^N + (\Omega + \sqrt{\Omega^2 - 1})^{-N}}{2} \quad |\Omega| \geq 1$$

Recursive calculation

$$T_{N+1}(\Omega) = 2\Omega T_N(\Omega) - T_{N-1}(\Omega)$$

Filter order:

$$N = \frac{\operatorname{arccosh} \left( \left[ \sqrt{\frac{1}{\delta_2^2} - 1} \right] / \varepsilon \right)}{\operatorname{arccosh} \left( \frac{\Omega_s}{\Omega_p} \right)}$$

$\delta_2$  is the maximum allowed stop band ripple.

Alternative for  $\operatorname{arccosh}$ :

$$\operatorname{arccosh}(x) = \ln \left( x + \sqrt{x^2 - 1} \right)$$

$N$	$T_N(\Omega)$
0	1
1	$\Omega$
2	$2\Omega^2 - 1$
3	$4\Omega^3 - 3\Omega$
4	$8\Omega^4 - 8\Omega^2 + 1$
5	$16\Omega^5 - 20\Omega^3 + 5\Omega$
6	$32\Omega^6 - 48\Omega^4 + 18\Omega^2 - 1$
7	$64\Omega^7 - 112\Omega^5 + 56\Omega^3 - 7\Omega$
8	$128\Omega^8 - 256\Omega^6 + 160\Omega^4 - 32\Omega^2 + 1$
9	$256\Omega^9 - 576\Omega^7 + 432\Omega^5 - 120\Omega^3 + 9\Omega$
10	$512\Omega^{10} - 1280\Omega^8 + 1120\Omega^6 - 400\Omega^4 + 50\Omega^2 - 1$

**Table 4.4. Chebyshev filter coefficients  $a_\nu$ .**

0.5dB ripple ( $\varepsilon = 0.349$ ,  $\varepsilon^2 = 0.122$ ).

N	$a_7$	$a_6$	$a_5$	$a_4$	$a_3$	$a_2$	$a_1$	$a_0$
1								2.863
2							1.426	1.516
3						1.253	1.535	0.716
4					1.197	1.717	1.025	0.379
5				1.172	1.937	1.309	0.752	0.179
6			1.159	2.172	1.589	1.172	0.432	0.095
7		1.151	2.413	1.869	1.648	0.756	0.282	0.045
8	1.146	2.657	2.149	2.184	1.148	0.573	0.152	0.024

1-dB ripple ( $\varepsilon = 0.509$ ,  $\varepsilon^2 = 0.259$ ).

N	$a_7$	$a_6$	$a_5$	$a_4$	$a_3$	$a_2$	$a_1$	$a_0$
1								1.965
2							1.098	1.102
3						0.989	1.238	0.491
4					0.953	1.454	0.743	0.276
5				0.937	1.689	0.974	0.580	0.123
6			0.928	1.931	1.202	0.939	0.307	0.069
7		0.923	2.176	1.429	1.357	0.549	0.214	0.031
8	0.920	2.423	1.655	1.837	0.447	0.448	0.107	0.017

2-dB ripple ( $\varepsilon = 0.765$ ,  $\varepsilon^2 = 0.585$ ).

N	$a_7$	$a_6$	$a_5$	$a_4$	$a_3$	$a_2$	$a_1$	$a_0$
1								1.307
2							0.804	0.823
3						0.738	1.022	0.327
4					0.716	1.256	0.517	0.206
5				0.705	1.499	0.693	0.459	0.082
6			0.701	1.745	0.867	0.771	0.210	0.051
7		0.698	1.994	1.039	1.144	0.383	0.166	0.020
8	0.696	2.242	1.212	1.579	0.598	0.359	0.073	0.013

3-dB\*) ripple ( $\varepsilon = 0.998$ ,  $\varepsilon^2 = 0.995$ ).

N	$a_7$	$a_6$	$a_5$	$a_4$	$a_3$	$a_2$	$a_1$	$a_0$
1								1.002
2							0.645	0.708
3						0.597	0.928	0.251
4					0.581	1.169	0.405	0.177
5				0.575	1.415	0.549	0.408	0.063
6			0.571	1.663	0.691	0.699	0.163	0.044
7		0.568	1.911	0.831	1.052	0.300	0.146	0.016
8	0.567	2.161	0.972	1.467	0.472	0.321	0.056	0.011

\*) The table was calculated with “exactly” 3dB, not with  $20 \cdot \log \sqrt{2} \approx 3.01$ dB. Hence  $\varepsilon \neq 1$  and  $a_0 \neq 1$  for  $N = 1$ .

**Table 4.5. Pole locations for Chebyshev filters.**

0.5dB ripple ( $\varepsilon = 0.349$ ,  $\varepsilon^2 = 0.122$ ).

$N = 1$	2	3	4	5	6	7	8
-2.863	-0.713	-0.626	-0.175	-0.362	-0.078	-0.256	-0.044
	$\pm j1.004$		$\pm j1.016$		$\pm j1.008$		$\pm j1.005$
		-0.313	-0.423	-0.112	-0.212	-0.057	-0.124
		$\pm j1.022$	$\pm j0.421$	$\pm j1.011$	$\pm j0.738$	$\pm j1.006$	$\pm j0.852$
				-0.293	-0.290	$\pm 0.160$	-0.186
				$\pm j0.625$	$\pm j0.270$	$\pm j0.807$	$\pm j0.570$
						-0.231	-0.220
						$\pm j0.448$	$\pm j0.200$

1-dB ripple ( $\varepsilon = 0.509$ ,  $\varepsilon^2 = 0.259$ ).

$N = 1$	2	3	4	5	6	7	8
-1.965	-0.549	-0.494	-0.139	-0.289	-0.062	-0.205	-0.035
	$\pm j0.895$		$\pm j0.983$		$\pm j0.993$		$\pm j0.996$
		-0.247	-0.337	-0.089	-0.170	-0.046	-0.100
		$\pm j0.966$	$\pm j0.407$	$\pm j0.990$	$\pm j0.727$	$\pm j0.995$	$\pm j0.845$
				-0.234	-0.232	-0.128	-0.149
				$\pm j0.612$	$\pm j0.266$	$\pm j0.798$	$\pm j0.564$
						-0.185	-0.176
						$\pm j0.443$	$\pm j0.198$

2-dB ripple ( $\varepsilon = 0.765$ ,  $\varepsilon^2 = 0.585$ ).

$N = 1$	2	3	4	5	6	7	8
-1.307	-0.402	-0.369	-0.105	-0.218	-0.047	-0.155	-0.026
	$\pm j0.813$		$\pm j0.958$		$\pm j0.982$		$\pm j0.990$
		-0.184	-0.253	-0.067	-0.128	-0.034	-0.075
		$\pm j0.923$	$\pm 0.397$	$\pm j0.973$	$\pm 0.719$	$\pm j0.987$	$\pm j0.839$
				-0.177	-0.175	-0.097	-0.113
				$\pm j0.602$	$\pm j0.263$	$\pm j0.791$	$\pm j0.561$
						-0.140	-0.133
						$\pm j0.439$	$\pm j0.197$

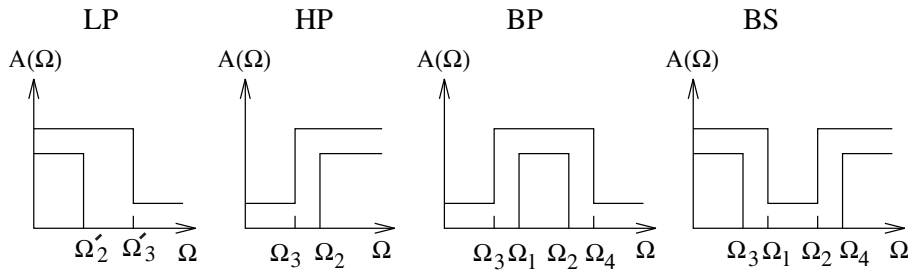
3-dB\*) ripple ( $\varepsilon = 0.998$ ,  $\varepsilon^2 = 0.995$ ).

$N = 1$	2	3	4	5	6	7	8
-1.002	-0.322	-0.299	-0.085	-0.177	-0.038	-0.126	-0.021
	$\pm j0.777$		$\pm j0.946$		$\pm j0.976$		$\pm 0.987$
		-0.1493	-0.206	-0.055	-0.104	-0.028	-0.061
		$\pm j0.904$	$\pm j0.392$	$\pm j0.966$	$\pm 0.715$	$\pm j0.983$	$\pm j0.836$
				-0.144	-0.143	-0.079	-0.092
				$\pm j0.597$	$\pm j0.262$	$\pm j0.789$	$\pm j0.559$
						-0.114	-0.108
						$\pm j0.437$	$\pm j0.196$

\*) See note in Table 4.4.

### 3.2 Frequency Transformations of Analogue Filter

1. Start out from the frequencies in the filter specification in the analogue high pass-, band pass- or band stop filter.
2. Transform to the frequencies of the LP filter  $\Omega'_2$  and  $\Omega'_3$ .
3. Find the coefficients of the LP filter coefficients.
4. Transform back to the original filter (HP, BP, BS) by replacing  $s$  i the low pass filter  $H(s)$ ; see below.



$$\text{HP} \Rightarrow \text{LP} \quad \Omega'_2 = \frac{1}{\Omega_2} \quad \Omega'_3 = \frac{1}{\Omega_3}$$

$$\text{BP} \Rightarrow \text{LP} \quad \Omega'_2 = \Omega_2 - \Omega_1 \quad \Omega'_3 = \Omega_4 - \Omega_3 \quad \Omega_1\Omega_2 = \Omega_3\Omega_4 = \Omega_I^2$$

$$\text{BS} \Rightarrow \text{LP} \quad \Omega'_2 = \frac{\Omega_I^2}{\Omega_4 - \Omega_3} \quad \Omega'_3 = \frac{\Omega_I^2}{\Omega_2 - \Omega_1} \quad \Omega_1\Omega_2 = \Omega_3\Omega_4 = \Omega_I^2$$

LP	HP	BP	BS	
$s$	$\rightarrow$	$\frac{1}{s}$	$s + \frac{\Omega_I^2}{s}$	$\frac{\Omega_I^2}{s + \frac{\Omega_I^2}{s}}$

## 4 Discrete Time Filters

### 4.1 FIR Filters and IIR Filters

$$\begin{aligned}\mathcal{H}(z) &= b_0 + b_1 z^{-1} + \dots + b_{M-1} z^{-M+1} \\ h(n) &= \begin{cases} b_n & 0 \leq n \leq M-1 \\ 0 & \text{Else} \end{cases}\end{aligned}$$

Linear phase FIR Filter :

Symmetrical impulse response  $h(n) = h(M-1-n)$

Anti-symmetrical impulse response  $h(n) = -h(M-1-n)$

IIR Filter

$$\begin{aligned}\mathcal{H}(z) &= \frac{b_0 + b_1 z^{-1} + \dots + b_M z^{-M}}{1 + a_1 z^{-1} + \dots + a_N z^{-N}} \\ h(n) &= Z^{-1}\{\mathcal{H}(z)\}\end{aligned}$$

### 4.2 Construction of FIR Filter

#### 4.2.1 FIR Filter using the window method

Impulse response

$$h(n) = h_d(n) \cdot w(n)$$

with desired impulse response  $h_d(n)$  and time window  $w(n)$ .

Filters designed using the window method have linear phase.

Desired impulse response  $h_d(n)$  defined for  $0 \leq n \leq M-1$

Odd filter length

Low pass:

$$\begin{aligned}h_d(n) &= \frac{\omega_c}{\pi} \frac{\sin \omega_c \left(n - \frac{M-1}{2}\right)}{\omega_c \left(n - \frac{M-1}{2}\right)} \\ h_d\left(\frac{M-1}{2}\right) &= \frac{\omega_c}{\pi} \\ H_d(\omega) &= \begin{cases} e^{-j\omega (M-1)/2} & 0 \leq |\omega| < \omega_c \\ 0 & \text{Else} \end{cases}\end{aligned}$$

High pass:

$$\begin{aligned}h_d(n) &= \delta\left(n - \frac{M-1}{2}\right) - \frac{\omega_c}{\pi} \frac{\sin \omega_c \left(n - \frac{M-1}{2}\right)}{\omega_c \left(n - \frac{M-1}{2}\right)} \\ H_d(\omega) &= \begin{cases} e^{-j\omega (M-1)/2} & \omega_c < |\omega| < \pi \\ 0 & \text{Else} \end{cases}\end{aligned}$$

Band pass:

$$h_d(n) = 2 \cos \left( \omega_0 \left( n - \frac{M-1}{2} \right) \right) \cdot \frac{\omega_c}{\pi} \frac{\sin \omega_c \left( n - \frac{M-1}{2} \right)}{\omega_c \left( n - \frac{M-1}{2} \right)}$$

$$H_d(\omega) = \begin{cases} e^{-j\omega (M-1)/2} & \omega_0 - \omega_c < |\omega| < \omega_0 + \omega_c \\ 0 & \text{Else} \end{cases}$$

Band stop:

$$h_d(n) = \delta \left( n - \frac{M-1}{2} \right) - 2 \cos \left( \omega_0 \left( n - \frac{M-1}{2} \right) \right) \cdot \frac{\omega_c}{\pi} \frac{\sin \omega_c \left( n - \frac{M-1}{2} \right)}{\omega_c \left( n - \frac{M-1}{2} \right)}$$

$$H_d(\omega) = \begin{cases} e^{-j\omega (M-1)/2} & 0 < |\omega| < \omega_0 - \omega_c \text{ and } \omega_0 + \omega_c < |\omega| < \pi \\ 0 & \text{Else} \end{cases}$$

The spectrum of the filter  $H(\omega) = H_d(\omega) * W(\omega)$ .

At the cut-off frequency  $\omega_c$  the attenuation is 6dB.

When dimensioning filters, the tables below offers a rough estimation of the required filter length  $M$ .

**Table 5.1**

The main and sidelobe size for some common window functions.

Window	Approximate main lobe width (rad)	Largest sidelobe relative main lobe (dB)
Rektangular	$4\pi/M$	-13
Bartlett	$8\pi/M$	-27
Hanning	$8\pi/M$	-32
Hamming	$8\pi/M$	-43
Blackman	$12\pi/M$	-58

**Table 5.2**

Sidelobe size for some filters designed using the window method.

Window	Approximate largest sidelobe (dB)
Rektangulr	-20
Bartlett	-27
Hanning	-40
Hamming	-50
Blackman	-70
Kaiser ( $\beta = 4.54$ )	-50
Kaiser ( $\beta = 6.76$ )	-70
Kaiser ( $\beta = 8.96$ )	-90



Window functions defined for  $0 \leq n \leq M - 1$  (Odd filter length  $M$ )

Rectangular window

$$w(n) = 1$$

Bartlett (Triangular window)

$$w(n) = 1 - \frac{\left|n - \frac{M-1}{2}\right|}{\frac{M-1}{2}}$$

Hanning window

$$\begin{aligned} w(n) &= 0.5 \left( 1 + \cos \frac{2\pi \left(n - \frac{M-1}{2}\right)}{M-1} \right) = \\ &= 0.5 \left( 1 - \cos \frac{2\pi n}{M-1} \right) \end{aligned}$$

Hamming window

$$\begin{aligned} w(n) &= 0.54 + 0.46 \cos \frac{2\pi \left(n - \frac{M-1}{2}\right)}{M-1} = \\ &= 0.54 - 0.46 \cos \frac{2\pi n}{M-1} \end{aligned}$$

Blackman window

$$\begin{aligned} w(n) &= 0.42 + 0.5 \cos \frac{2\pi \left(n - \frac{M-1}{2}\right)}{M-1} + \\ &+ 0.08 \cos \frac{4\pi \left(n - \frac{M-1}{2}\right)}{M-1} = \\ &= 0.42 - 0.5 \cos \frac{2\pi n}{M-1} + 0.08 \cos \frac{4\pi n}{M-1} \end{aligned}$$

Kaiser window

$$w(n) = \frac{I_0 \left( \beta \sqrt{1 - [2(n - \frac{M-1}{2}) / (M-1)]^2} \right)}{I_0(\beta)}$$

$I_0(x)$  is a Bessel function, whose value can be calculated according to

$$I_0(x) = 1 + \sum_{k=1}^L \left[ \frac{(x/2)^k}{k!} \right]^2$$

Usually  $L < 25$ .

The value of  $\beta$  is decided from the stop band attenuation  $D$ .

Stop band attenuation $D$	$\beta$
$D \leq 21\text{dB}$	0
$21\text{dB} < D < 50\text{dB}$	$0,5842(D - 21)^{0,4} + 0,07886(D - 21)$
$D \geq 50\text{dB}$	$0,1102(D - 8,7)$

Filter length  $M$ :

$$M \geq \frac{D - 7,95}{14,36\Delta f}$$

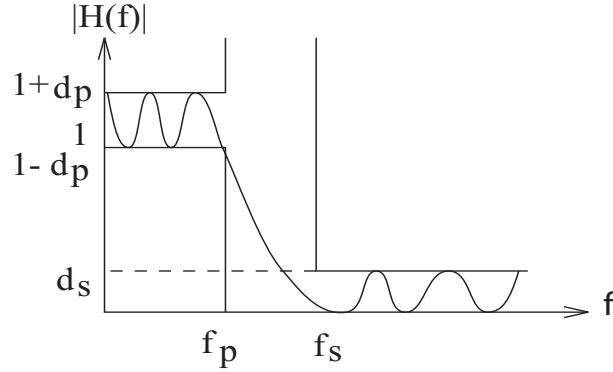
$$\Delta f = f_s - f_p$$

$f_p$  Pass band frequency (normalized frequency)

$f_s$  Stop band frequency (normalized frequency)

### 4.2.2 Equiripple FIR Filter

Dimensioning of equiripple filters when using the Remez algorithm. Approximately according to Kaiser.



$$\text{Filter length } M = \frac{-20 \log_{10} \left( \sqrt{\delta_p \delta_s} \right) - 13}{14.6 \Delta f} + 1$$

$$\Delta f = f_s - f_p$$

### 4.2.3 FIR Filters using Least-Squares

Minimizing

$$\mathcal{E} = \sum_n [x(n) * h(n) - d(n)]^2$$

yields

$$\sum_{n=0}^{M-1} h(n) r_{xx}(n - \ell) = r_{dx}(\ell) \quad \ell = 0, \dots, M - 1$$

and

$$\mathcal{E}_{\min} = r_{dd}(0) - \sum_{k=0}^{M-1} h(k) r_{dx}(k)$$

where  $r_{xx}(\ell)$  is the correlation function for  $x(n)$  and  $r_{dx}(\ell)$  is the cross correlation between  $d(n)$  and  $x(n)$ .

This can be written on matrix form as

$$\begin{aligned} \mathbf{R}_{xx} \cdot \mathbf{h} &= \mathbf{r}_{dx} \\ \mathbf{h} &= \mathbf{R}_{xx}^{-1} \cdot \mathbf{r}_{dx} \\ \mathcal{E}_{\min} &= r_{dd}(0) - \mathbf{h}^T \cdot \mathbf{r}_{dx} \end{aligned}$$

## 4.3 Constructing IIR Filters starting from Analogue Filters

### 4.3.1 The Impulse-Invariant Method

$$h(n) = h_a(t)|_{t=nT} = h_a(nT)$$

$$H(z) = \sum_{n=0}^{\infty} h_a(nT)z^{-n}$$

1.

$$h_a(t) = e^{-\sigma_0 t} \longleftrightarrow \mathcal{H}_a(s) = \frac{1}{s + \sigma_0}$$

$$\Rightarrow \mathcal{H}(z) = \frac{1}{1 - e^{-\sigma_0 T} z^{-1}}$$

2.

$$h_a(t) = e^{-\sigma_0 t} \cos \Omega_0 t \longleftrightarrow \mathcal{H}_a(s) = \frac{s + \sigma_0}{(s + \sigma_0)^2 + \Omega_0^2}$$

$$\Rightarrow \mathcal{H}(z) = \frac{1 - z^{-1} e^{-\sigma_0 T} \cos \Omega_0 T}{1 - 2z^{-1} e^{-\sigma_0 T} \cos \Omega_0 T + z^{-2} e^{-2\sigma_0 T}}$$

3.

$$h_a(t) = e^{-\sigma_0 t} \sin \Omega_0 t \longleftrightarrow \mathcal{H}_a(s) = \frac{\Omega_0}{(s + \sigma_0)^2 + \Omega_0^2}$$

$$\Rightarrow \mathcal{H}(z) = \frac{z^{-1} e^{-\sigma_0 T} \sin \Omega_0 T}{1 - 2z^{-1} e^{-\sigma_0 T} \cos \Omega_0 T + z^{-2} e^{-2\sigma_0 T}}$$

### 4.3.2 Bilinear Transformation

Frequency transformation (“prewarp”)

$$\Omega_{\text{prewarp}} = \frac{2}{T} \tan \frac{\omega}{2}$$

Analogue construction of filters in the variable  $\Omega_{\text{prewarp}} = 2\pi F_{\text{prewarp}}$  ( $\omega = 2\pi f$ )

$$\mathcal{H}(z) = \mathcal{H}_a(s) \text{ where } s = \frac{2}{T} \frac{1 - z^{-1}}{1 + z^{-1}}$$

$T$  is a normalization factor (can often be set to 1).

### 4.3.3 Quantizing of Coefficients

Change in pole locations when the coefficients  $a_1, \dots, a_k$  is changed  $\Delta a_1, \dots, \Delta a_k$

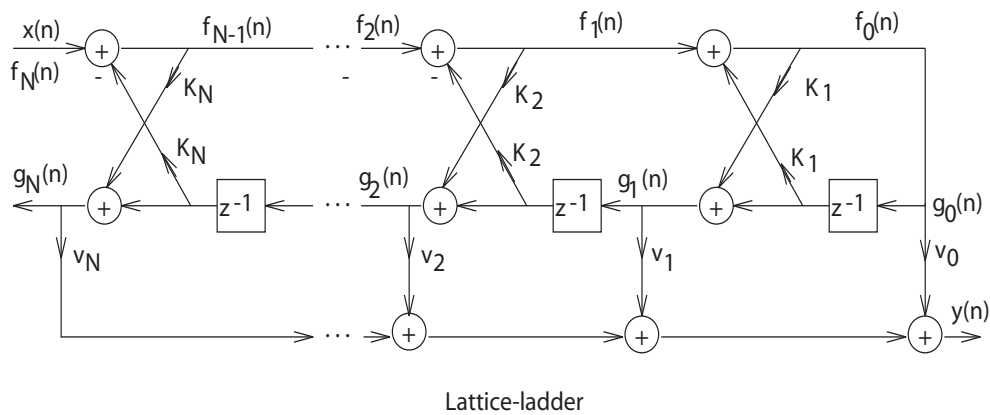
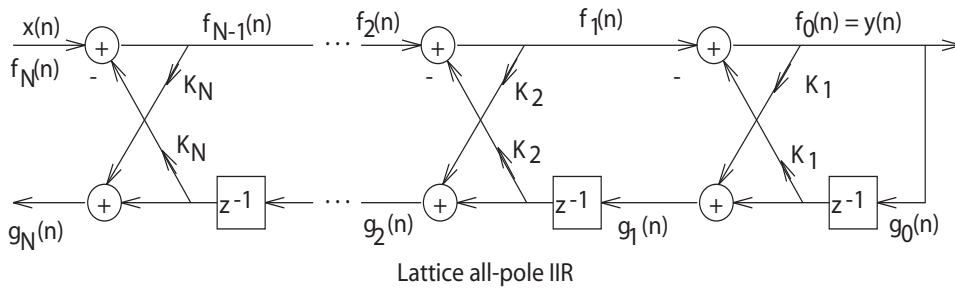
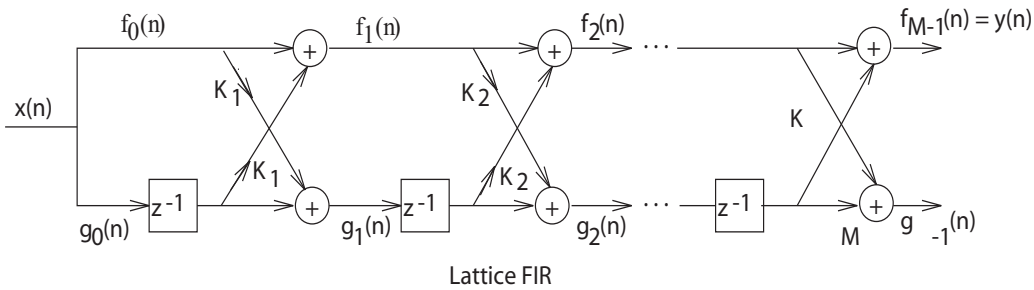
$$\Delta p_i \approx \frac{\partial p_i}{\partial a_1} \Delta a_1 + \dots + \frac{\partial p_i}{\partial a_k} \Delta a_k$$

If using normal form (Direct form II) then

$$\frac{\partial p_i}{\partial a_j} = \frac{-p_i^{k-j}}{\underbrace{(p_i - p_1)(p_i - p_2) \dots (p_i - p_k)}_{k-1 \text{ factors}}}$$

$(p_i - p_i)$  should not be included

## 4.4 Lattice Filters



Conversion from Lattice to Direct form:

$$A_0(z) = B_0(z) = 1$$

$$\begin{cases} A_m(z) = A_{m-1}(z) + K_m z^{-1} B_{m-1}(z) \\ B_m(z) = K_m A_{m-1}(z) + z^{-1} B_{m-1}(z) \end{cases}$$

$$m = 1, 2, \dots, M - 1$$

Relationship between  $A_m(z)$  and  $B_m(z)$

$$A_m(z) = \alpha_m(0) + \alpha_m(1)z^{-1} + \dots + \alpha_m(m-1)z^{-m+1} + \alpha_m(m)z^{-m}$$

$$B_m(z) = \beta_m(0) + \beta_m(1)z^{-1} + \dots + \beta_m(m-1)z^{-m+1} + \beta_m(m)z^{-m}$$

$$B_m(z) = z^{-m} A_m(z^{-1})$$

$$\beta_m(k) = \alpha_m(m-k)$$

$$k = 0, 1, \dots, m$$

Conversion from Direct form to Lattice:

$$A_{m-1}(z) = \frac{1}{1 - K_m^2} (A_m(z) - K_m B_m(z))$$

$$m = M - 1, M - 2, \dots, 1$$

Reflection coefficient:

$$K_m = \alpha_m(m)$$

**FIR Filter**

$$H(z) = A_N(z)$$

$$A_{M-1}(z) = A_N(z)$$

**IIR Filter (all-pole)**

$$H(z) = \frac{1}{A_N(z)}$$

$$A_{M-1}(z) = A_N(z)$$

**Lattice-Ladder**

Conversion from Direct form to Lattice-Ladder:

$$H(z) = \frac{C_M(z)}{A_N(z)} = \frac{c_0 + c_1 z^{-1} \dots c_M z^{-M}}{A_N(z)}$$

$$C_{m-1}(z) = C_m(z) - v_m B_m(z)$$

$$v_m = c_m(m) \quad m = 0, 1, \dots, M$$

## 5 Correlation

Correlation, Cross Correlation, spectrum, cross spectrum and coherence between input and output signal.

Continuous time process:

$$\begin{aligned}
 y(t) &= h(t) * x(t) \\
 Y(F) &= H(F) \cdot X(F) \\
 r_{yy}(\tau) &= r_{hh}(\tau) * r_{xx}(\tau) \\
 R_{yy}(F) &= |H(F)|^2 R_{xx}(F) \\
 r_{yx}(\tau) &= h(\tau) * r_{xx}(\tau) \\
 R_{yx}(F) &= H(F) \cdot R_{xx}(F) \\
 r_{xx}(\tau) &= \int_t x(t)x(t-\tau)dt \\
 r_{yx}(\tau) &= \int_t y(t)x(t-\tau)dt \\
 \gamma_{xx}(\tau) &= E\{x(t)x(t-\tau)\} \\
 \gamma_{yx}(\tau) &= E\{y(t)x(t-\tau)\}
 \end{aligned}$$

Discrete time process:

$$\begin{aligned}
 y(n) &= h(n) * x(n) \\
 Y(f) &= H(f) \cdot X(f) \\
 r_{yy}(n) &= r_{hh}(n) * r_{xx}(n) \\
 R_{yy}(f) &= |H(f)|^2 \cdot R_{xx}(f) \\
 r_{yx}(n) &= h(n) * r_{xx}(n) \\
 R_{yx}(f) &= H(f) \cdot R_{xx}(f) \\
 r_{xx}(n) &= \sum_{\ell} x(\ell)x(\ell-n) \\
 r_{yx}(n) &= \sum_{\ell} y(\ell)x(\ell-n) \\
 \gamma_{xx}(n) &= E\{x(\ell)x(\ell-n)\} \\
 \gamma_{yx}(n) &= E\{y(\ell)x(\ell-n)\}
 \end{aligned}$$

Normally distributed stochastic variables.  $X_i \in N(m_i, \sigma_i)$

$$\begin{aligned}
 E\{X_1 X_2 X_3 X_4\} &= E\{X_1 X_2\} E\{X_3 X_4\} + E\{X_1 X_3\} E\{X_2 X_4\} + \\
 &+ E\{X_1 X_4\} E\{X_2 X_3\} - 2m_1 m_2 m_3 m_4
 \end{aligned}$$

## 6 Spectrum Estimation

### Spectrum Estimation

$$\gamma_{xx}(m) = E\{x(n)x(n+m)\} \text{ autocorrelation}$$

$$\Gamma_{xx}(f) = \sum_{m=-\infty}^{\infty} \gamma_{xx} e^{-j2\pi f m} \text{ power spectrum}$$

$$r_{xx}(m) = \frac{1}{N} \sum_{n=0}^{N-m-1} x(n)x(n+m) \quad 0 \leq m \leq N-1 \quad \text{auto correlation (estimate)}$$

$$P_{xx}(f) = \sum_{m=-N+1}^{N-1} r_{xx}(m) e^{-j2\pi f m} = \frac{1}{N} \left| \sum_{m=0}^{N-1} x(m) e^{-j2\pi f m} \right|^2 \quad \text{power spectrum (estimate)}$$

### Periodogram

$$P_{xx}(f) = \frac{1}{N} \left| \sum_{m=0}^{N-1} x(m) e^{-j2\pi f m} \right|^2 \quad \text{power spectrum (estimate)}$$

$$E\{r_{xx}(m)\} = \left(1 - \frac{|m|}{N}\right) \gamma_{xx}(m) \rightarrow \gamma_{xx}(m) \text{ when } N \rightarrow \infty$$

$$\text{var}(r_{xx}(m)) \approx \frac{1}{N} \sum_{n=-\infty}^{\infty} [\gamma_{xx}^2(n) + \gamma_{xx}(n-m)\gamma_{xx}(n+m)] \rightarrow 0 \text{ when } N \rightarrow \infty$$

$$E\{P_{xx}(f)\} = \int_{-1/2}^{1/2} \Gamma_{xx}(\alpha) W_B(f - \alpha) d\alpha$$

where  $W_B(f)$  is the Fourier transform of the Bartlett window  $\left(1 - \frac{|m|}{N}\right)$

$$\text{var}(P_{xx}(f)) = \Gamma_{xx}^2(f) \left[1 + \left(\frac{\sin 2\pi f N}{N \sin 2\pi f}\right)^2\right] \rightarrow \Gamma_{xx}^2(f) \text{ when } N \rightarrow \infty$$

if  $x(n)$  Gaussian.

### Periodogram using DFT:

$$P_{xx}\left(\frac{k}{N}\right) = \frac{1}{N} \left| \sum_{n=0}^{N-1} x(n) e^{-j2\pi \frac{nk}{N}} \right|^2 \quad k = 0, \dots, N-1$$



## Bartlett's method

Averaging periodograms.

Divide the  $N$  sample data sequence  $x(n)$  into  $K$  blocks with  $M$  samples in each block. The blocks are not overlapping ( $N = K \cdot M$ ).

$$x_i(n) = x(n + iM) \quad i = 0, 1, \dots, K - 1 \quad \text{and} \quad n = 0, 1, \dots, M - 1$$

$$P_{xx}^i(f) = \frac{1}{M} \left| \sum_{m=0}^{M-1} x_i(m) e^{-j2\pi f m} \right|^2 \quad i = 0, 1, \dots, K - 1$$

$$P_{xx}^B(f) = \frac{1}{K} \sum_{i=0}^{K-1} P_{xx}^i(f)$$

$$E\{P_{xx}^B(f)\} = \frac{1}{K} \sum_{i=0}^{K-1} E\{P_{xx}^i(f)\} = E\{P_{xx}^i(f)\}$$

$$E\{P_{xx}^i(f)\} = \frac{1}{M} \int_{-1/2}^{1/2} \Gamma_{xx}(\alpha) W_B(f - \alpha) d\alpha$$

where  $W_B(f)$  is the Fourier Transform of the Bartlett window  $(1 - \frac{|m|}{M})$

$$W_B(f) = \frac{1}{M} \left( \frac{\sin \pi f M}{\sin \pi f} \right)^2$$

$$\text{var}(P_{xx}^B(f)) = \frac{1}{K^2} \sum_{i=0}^{K-1} \text{var}(P_{xx}^i(f))$$

$$\text{var}(P_{xx}^B(f)) = \frac{1}{K} \Gamma_{xx}^2(f) \left[ 1 + \left( \frac{\sin 2\pi f N}{N \sin 2\pi f} \right)^2 \right]$$

## Welch's method

Averaging modified periodograms.

Divide the  $N$  sample data sequence  $x(n)$  into  $L$  block with  $M$  samples in each block. The blocks may be overlapping. The starting point for block  $i$  is given by  $iD$ .  $D = M$  means no block overlap and  $D = M/2$  yields 50% overlap.

$$x_i(n) = x(n + iD) \quad i = 0, 1, \dots, L - 1 \quad \text{and} \quad n = 0, 1, \dots, M - 1$$

Modified periodogram.

$$\tilde{P}_{xx}^i(f) = \frac{1}{MU} \left| \sum_{m=0}^{N-1} x_i(m)w(n)e^{-j2\pi fm} \right|^2 \quad i = 0, 1, \dots, L-1$$

$w(n)$  window function

$$U = \frac{1}{M} \sum_{n=0}^{N-1} w^2(n)$$

$$P_{xx}^W(f) = \frac{1}{L} \sum_{i=0}^{L-1} \tilde{P}_{xx}^i(f)$$

$$E\{P_{xx}^W(f)\} = \frac{1}{L} \sum_{i=0}^{L-1} E\{\tilde{P}_{xx}^i(f)\} = E\{\tilde{P}_{xx}^i(f)\}$$

$$E\{\tilde{P}_{xx}^i(f)\} = \int_{-1/2}^{1/2} \Gamma_{xx}(\alpha)W(f-\alpha)d\alpha$$

$$W(f) = \frac{1}{MU} \left| \sum_{n=0}^{M-1} w(n)e^{-j2\pi fn} \right|^2$$

No block overlap ( $L = K$ )

$$\begin{aligned} \text{var}(P_{xx}^W(f)) &= \frac{1}{L} \text{var}(\tilde{P}_{xx}^i(f)) \\ &\approx \frac{1}{L} \Gamma_{xx}^2(f) \end{aligned}$$

50% block overlap ( $L = 2K$ ) and Bartlett window (triangular window).

$$\text{var}(P_{xx}^W(f)) \approx \frac{9}{8L} \Gamma_{xx}^2(f)$$

## Averaging Periodograms

Quality factor

$$Q = \frac{[E\{P_{xx}(f)\}]^2}{\text{var}(P_{xx}(f))}$$

Relative variance  $\frac{1}{Q}$

$Q \approx$  time-bandwidth product.

Periodogram	$\Delta f = \frac{0.9}{M}$	$Q = 1$	
Bartlett ( $N = K \cdot M$ )	$\Delta f = \frac{0.9}{M}$	$Q = \frac{N}{M}$	Rectangular window No overlap
Welch ( $N = L \cdot M$ )	$\Delta f = \frac{1.28}{M}$	$Q = \frac{16}{9} \cdot \frac{N}{M}$	Triangular window 50% overlap
Welch ( $N = L \cdot M$ )	$\Delta f = \frac{1.50}{M}$	$Q = \frac{3}{1.08 \cdot 2} \cdot \frac{N}{M}$	Hanning window 50% overlap
Welch	$\Delta f = \frac{1.50}{M}$	$Q = \frac{3}{2} \cdot \frac{N}{M}$	Hanning window 62,5% overlap
Blackman/Tukey	$\Delta f = \frac{0.6}{M}$	$Q = \frac{1}{2} \cdot \frac{N}{M}$	Rectangular window
	$\Delta f = \frac{0.9}{M}$	$Q = \frac{3}{2} \cdot \frac{N}{M}$	Triangular window

The resolution  $\Delta f$  is calculated at the -3dB points from the main lobe of the window.

# A Basic Relationships

## A.1 Trigonometrical formulas

$$\begin{aligned}\sin \alpha &= \cos(\alpha - \pi/2) \\ \cos \alpha &= \sin(\alpha + \pi/2) \\ \cos^2 \alpha + \sin^2 \alpha &= 1 \\ \cos^2 \alpha - \sin^2 \alpha &= \cos 2\alpha \\ 2 \sin \alpha \cos \alpha &= \sin 2\alpha \\ \sin(-\alpha) &= -\sin \alpha \\ \cos(-\alpha) &= \cos \alpha \\ \cos^2 \alpha &= \frac{1}{2}(1 + \cos 2\alpha)\end{aligned}$$

$$\begin{aligned}\sin(\alpha + \beta) &= \sin \alpha \cos \beta + \cos \alpha \sin \beta \\ \cos(\alpha + \beta) &= \cos \alpha \cos \beta - \sin \alpha \sin \beta \\ 2 \sin \alpha \sin \beta &= \cos(\alpha - \beta) - \cos(\alpha + \beta) \\ 2 \sin \alpha \cos \beta &= \sin(\alpha + \beta) + \sin(\alpha - \beta) \\ 2 \cos \alpha \cos \beta &= \cos(\alpha + \beta) + \cos(\alpha - \beta) \\ \sin \alpha + \sin \beta &= 2 \sin \frac{\alpha + \beta}{2} \cos \frac{\alpha - \beta}{2} \\ \cos \alpha + \cos \beta &= 2 \cos \frac{\alpha + \beta}{2} \cos \frac{\alpha - \beta}{2}\end{aligned}$$

$$\cos \alpha = \frac{1}{2} (e^{j\alpha} + e^{-j\alpha}), \quad \sin \alpha = \frac{1}{2j} (e^{j\alpha} - e^{-j\alpha}), \quad e^{j\alpha} = \cos \alpha + j \sin \alpha$$

$$A \cos \alpha + B \sin \alpha = \sqrt{A^2 + B^2} \cos(\alpha - \beta)$$

$$\text{where } \cos \beta = \frac{A}{\sqrt{A^2 + B^2}}, \quad \sin \beta = \frac{B}{\sqrt{A^2 + B^2}}$$

$$\text{and } \beta = \begin{cases} \arctan \frac{B}{A} & \text{if } A \geq 0 \\ \arctan \frac{B}{A} + \pi & \text{if } A < 0 \end{cases}$$

$$A \cos \alpha + B \sin \alpha = \sqrt{A^2 + B^2} \sin(\alpha + \beta)$$

$$\text{where } \cos \beta = \frac{B}{\sqrt{A^2 + B^2}}, \quad \sin \beta = \frac{A}{\sqrt{A^2 + B^2}}$$

$$\text{and } \beta = \begin{cases} \arctan \frac{A}{B} & \text{if } B \geq 0 \\ \arctan \frac{A}{B} + \pi & \text{if } B < 0 \end{cases}$$

Degrees	Rad	sin	cos	tan	cot
0	0	0	1	0	$\pm\infty$
30	$\frac{\pi}{6}$	$\frac{1}{2}$	$\frac{\sqrt{3}}{2}$	$\frac{1}{\sqrt{3}}$	$\sqrt{3}$
45	$\frac{\pi}{4}$	$\frac{1}{\sqrt{2}}$	$\frac{1}{\sqrt{2}}$	1	1
60	$\frac{\pi}{3}$	$\frac{\sqrt{3}}{2}$	$\frac{1}{2}$	$\sqrt{3}$	$\frac{1}{\sqrt{3}}$
90	$\frac{\pi}{2}$	1	0	$\pm\infty$	0

## A.2 Some Common Relationships

Sum of geometrical series

$$\sum_{n=0}^{N-1} a^n = \begin{cases} N & \text{if } a = 1 \\ \frac{1-a^N}{1-a} & \text{if } a \neq 1 \end{cases}$$

Summation of sinusoid over an even number of periods

$$\sum_{n=0}^{N-1} e^{j2\pi kn/N} = \begin{cases} N & \text{if } k = 0, \pm N, \dots \\ 0 & \text{Else} \end{cases}$$

## A.3 Matrix Theory

**Matrix notation  $\mathbf{A}$  and vector  $\mathbf{x}$**

A matrix  $\mathbf{A}$  with dimension  $m \times n$  and a vector  $\mathbf{x}$  with dimension  $n$  is defined by

$$\mathbf{A} = \begin{bmatrix} a_{11} & a_{12} & \dots & a_{1n} \\ a_{21} & a_{22} & \dots & a_{2n} \\ \vdots & \vdots & & \vdots \\ a_{m1} & a_{m2} & \dots & a_{mn} \end{bmatrix} \quad \mathbf{x} = \begin{bmatrix} x_1 \\ x_2 \\ \vdots \\ x_n \end{bmatrix}$$

The matrix  $\mathbf{A}$  is symmetrical if  $a_{ij} = a_{ji} \forall ij$ .

$\mathbf{I}$  denotes the unity matrix.

**Transposing matrix  $\mathbf{A}$**

$$\mathbf{B} = \mathbf{A}^T \text{ where } b_{ij} = a_{ji}$$

$$(\mathbf{AB})^T = \mathbf{B}^T \mathbf{A}^T$$

## Matrix $\mathbf{A}$ determinant

$$\det \mathbf{A} = |\mathbf{A}| = \begin{vmatrix} a_{11} & a_{12} & \cdots & a_{1n} \\ a_{21} & a_{22} & \cdots & a_{2n} \\ \vdots & \vdots & & \vdots \\ a_{m1} & a_{m2} & \cdots & a_{mn} \end{vmatrix} = \sum_{i=1}^n a_{ij} (-1)^{i+j} \det \mathbf{M}_{ij}$$

where  $\mathbf{M}_{ij}$  is the resulting matrix if row  $i$  and column  $j$  in matrix  $\mathbf{A}$  is deleted.

$$\det \mathbf{AB} = \det \mathbf{A} \cdot \det \mathbf{B}$$

Especially, for a  $2 \times 2$  matrix:

$$\det \mathbf{A} = a_{11}a_{22} - a_{12}a_{21}$$

## Inverse of matrix $\mathbf{A}$

$$\mathbf{A}^{-1}\mathbf{A} = \mathbf{AA}^{-1} = \mathbf{I} \quad (\text{if } \det \mathbf{A} \neq 0)$$

$$\mathbf{A}^{-1} = \frac{1}{\det \mathbf{A}} \cdot \mathbf{C}^T$$

where  $\mathbf{C}$  is defined by

$$c_{ij} = (-1)^{i+j} \cdot \det \mathbf{M}_{ij}$$

$$(\mathbf{AB})^{-1} = \mathbf{B}^{-1}\mathbf{A}^{-1}$$

Especially for a  $2 \times 2$  matrix:

$$\mathbf{A}^{-1} = \frac{1}{\det \mathbf{A}} \begin{pmatrix} a_{22} & -a_{12} \\ -a_{21} & a_{11} \end{pmatrix}$$

## Eigenvalues and Eigenvectors

The eigenvalues ( $\lambda_i$ ,  $i = 1, 2, \dots, n$ ) and the eigenvectors ( $\mathbf{q}_i$ ,  $i = 1, 2, \dots, n$ ) are solutions for the equation system

$$\mathbf{A}\mathbf{q} = \lambda\mathbf{q} \text{ or } (\mathbf{A} - \lambda\mathbf{I})\mathbf{q} = 0$$

The eigenvalues can be calculated as solutions to the characteristic equation (secular equation) for  $\mathbf{A}$

$$\det(\lambda\mathbf{I} - \mathbf{A}) = \lambda^n + \alpha_{n-1}\lambda^{n-1} + \cdots + \alpha_0 = 0$$

$\det(\lambda\mathbf{I} - \mathbf{A})$  is called the characteristic polynomial (the secular polynomial) for  $\mathbf{A}$ .

# B Transforms

## B.1 The Laplace Transform

### B.1.1 The Laplace transform of causal signals

In the table below  $f(t) = 0$  for  $t < 0$  (i.e.  $f(t) \cdot u(t) = f(t)$ ).

1.	$f(t) = \frac{1}{2\pi j} \int_{\sigma-j\infty}^{\sigma+j\infty} \mathcal{F}(s)e^{st} ds$	$\longleftrightarrow$	$\mathcal{F}(s) = \int_{0-}^{\infty} f(t)e^{-st} dt$
2.	$\sum_{\nu} a_{\nu} f_{\nu}(t)$	$\longleftrightarrow$	$\sum_{\nu} a_{\nu} \mathcal{F}_{\nu}(s)$ Linearity
3.	$f(at)$	$\longleftrightarrow$	$\frac{1}{a} \mathcal{F}\left(\frac{s}{a}\right)$ Scaling
4.	$\frac{1}{a} f\left(\frac{t}{a}\right)$	$\longleftrightarrow$	$\mathcal{F}(as)$ $a > 0$ Scaling
5.	$f(t - t_0); t \geq t_0$	$\longleftrightarrow$	$\mathcal{F}(s) e^{-st_0}$ Time shift
6.	$f(t) \cdot e^{-at}$	$\longleftrightarrow$	$\mathcal{F}(s + a)$ Frequency shift
7.	$\frac{d^n f}{dt^n}$	$\longleftrightarrow$	$s^n \mathcal{F}(s)$ Derivation
8.	$\int_{0-}^t f(\tau) d\tau$	$\longleftrightarrow$	$\frac{1}{s} \mathcal{F}(s)$ Integration
9.	$(-t)^n f(t)$	$\longleftrightarrow$	$\frac{d^n \mathcal{F}(s)}{ds^n}$ Derivation in frequency domain
10.	$\frac{f(t)}{t}$	$\longleftrightarrow$	$\int_s^{\infty} \mathcal{F}(z) dz$ Integration in frequency domain
11.	$\lim_{t \rightarrow 0} f(t) = \lim_{s \rightarrow \infty} s \cdot \mathcal{F}(s)$		Initial value theorem
12.	$\lim_{t \rightarrow \infty} f(t) = \lim_{s \rightarrow 0} s \cdot \mathcal{F}(s)$		End value theorem
13.	$f_1(t) * f_2(t) = \int_0^t f_1(\tau) f_2(t - \tau) d\tau = \int_0^t f_1(t - \tau) f_2(\tau) d\tau$	$\longleftrightarrow$	$\mathcal{F}_1(s) \cdot \mathcal{F}_2(s)$ Convolution in time domain
14.	$f_1(t) \cdot f_2(t)$	$\longleftrightarrow$	$\frac{1}{2\pi j} \mathcal{F}_1(s) * \mathcal{F}_2(s) = \frac{1}{2\pi j} \int_{\sigma-j\infty}^{\sigma+j\infty} \mathcal{F}_1(z) \cdot \mathcal{F}_2(s - z) \cdot dz$ Convolution in frequency domain
15.	$\int_{0-}^{\infty} f_1(t) \cdot f_2(t) dt = \frac{1}{2\pi j} \int_{\sigma-j\infty}^{\sigma+j\infty} \mathcal{F}_1(s) \cdot \mathcal{F}_2(-s) ds$		Parseval's relation

16.	$\delta(t)$	$\longleftrightarrow$	1
17.	$\delta^n(t)$	$\longleftrightarrow$	$s^n$
18.	1	$\longleftrightarrow$	$\frac{1}{s}$
19.	$\frac{1}{n!} t^n$	$\longleftrightarrow$	$\frac{1}{s^{n+1}}$
20.	$e^{-\sigma_0 t}$	$\longleftrightarrow$	$\frac{1}{s + \sigma_0}$
21.	$\frac{1}{(n-1)!} t^{n-1} e^{-\sigma_0 t}$	$\longleftrightarrow$	$\frac{1}{(s + \sigma_0)^n}$
22.	$\sin \Omega_0 t$	$\longleftrightarrow$	$\frac{\Omega_0}{s^2 + \Omega_0^2}$
23.	$\cos \Omega_0 t$	$\longleftrightarrow$	$\frac{s}{s^2 + \Omega_0^2}$
24.	$t \cdot \sin \Omega_0 t$	$\longleftrightarrow$	$\frac{2\Omega_0 s}{(s^2 + \Omega_0^2)^2}$
25.	$t \cdot \cos \Omega_0 t$	$\longleftrightarrow$	$\frac{s^2 - \Omega_0^2}{(s^2 + \Omega_0^2)^2}$
26.	$e^{-\sigma_0 t} \sin \Omega_0 t$	$\longleftrightarrow$	$\frac{\Omega_0}{(s + \sigma_0)^2 + \Omega_0^2}$
27.	$e^{-\sigma_0 t} \cos \Omega_0 t$	$\longleftrightarrow$	$\frac{s + \sigma_0}{(s + \sigma_0)^2 + \Omega_0^2}$
28.	$e^{-\sigma_0 t} \sin(\Omega_0 t + \phi)$	$\longleftrightarrow$	$\frac{(s + \sigma_0) \sin \phi + \Omega_0 \cos \phi}{(s + \sigma_0)^2 + \Omega_0^2}$

### B.1.2 One-sided Laplace transform of non-causal signals

Notation

$\mathcal{F}^+(s) = \int_{0-}^{\infty} f(t) e^{-st} dt$	Single sided Laplace transform, $f(t)$ not necessarily causal.
$\mathcal{F}(s) = \mathcal{F}^+(s)$	For causal signals

Taking the derivative of  $f(t)$  yields

$\frac{d}{dt} f(t)$	$\longleftrightarrow$	$s \cdot \mathcal{F}^+(s) - f(0-)$	Single derivation
$\frac{d^n}{dt^n} f(t)$	$\longleftrightarrow$	$s^n \mathcal{F}^+(s) - s^{n-1} f(0-)$ $- s^{n-2} f^{(1)}(0-) - \dots - f^{(n-1)}(0-)$	$n$ derivations



## B.2 The Fourier Transform of a Continuous Time Signal

$$\Omega = 2\pi F$$

1.  $w(t) = \mathcal{F}^{-1}\{W(F)\} = \int_{-\infty}^{\infty} W(F)e^{j2\pi Ft}dF \longleftrightarrow W(F) = \mathcal{F}\{w(t)\} = \int_{-\infty}^{\infty} w(t)e^{-j2\pi Ft}dt$
2.  $\sum_{\nu} a_{\nu}w_{\nu}(t) \longleftrightarrow \sum_{\nu} a_{\nu}W_{\nu}(F)$
3.  $w^*(-t) \longleftrightarrow W^*(F)$
4.  $W(t) \longleftrightarrow w(-F)$
5.  $w(at) \longleftrightarrow \frac{1}{|a|} W\left(\frac{F}{a}\right)$
6.  $w(t - t_0) \longleftrightarrow W(F) \cdot e^{-j2\pi Ft_0}$
7.  $w(t) \cdot e^{j2\pi F_0 t} \longleftrightarrow W(F - F_0)$
8.  $w^*(t) \longleftrightarrow W^*(-F)$
9.  $\frac{d^n w(t)}{dt^n} \longleftrightarrow (j2\pi F)^n W(F)$
10.  $\int_{-\infty}^t w(\tau)d\tau \longleftrightarrow \frac{1}{j2\pi F} W(F)$  if  $W(F) = 0$  for  $F = 0$
11.  $-j2\pi t w(t) \longleftrightarrow \frac{dw}{dF}$
12.  $w_1(t) * w_2(t) \longleftrightarrow W_1(F) \cdot W_2(F)$
13.  $w_1(t) \cdot w_2(t) \longleftrightarrow W_1(F) * W_2(F)$
14.  $\int_{-\infty}^{\infty} |w(t)|^2 dt = \int_{-\infty}^{\infty} |W(F)|^2 dF$  Parseval's relation
15.  $\int_{-\infty}^{\infty} w_1(t) \cdot w_2(t) dt = \int_{-\infty}^{\infty} W_1(F) \cdot W_2^*(F) dF$   $w_1(t), w_2(t)$  real
16.  $\delta(t) \longleftrightarrow 1$
17.  $1 \longleftrightarrow \delta(F)$
18.  $u(t) \longleftrightarrow \frac{1}{j2\pi F} + \frac{1}{2} \delta(F)$
19.  $e^{-at}u(t) \longleftrightarrow \frac{1}{a+j\Omega}$

20.  $e^{-a|t|} \longleftrightarrow \frac{2a}{a^2 + \Omega^2}$
21.  $e^{j2\pi F_0 t} \longleftrightarrow \delta(F - F_0)$
22.  $\sin 2\pi F_0 t \longleftrightarrow j \frac{1}{2} \{\delta(F + F_0) - \delta(F - F_0)\}$
23.  $\sin 2\pi F_0 t \cdot u(t) \longleftrightarrow \frac{\Omega_0}{\Omega_0^2 - \Omega^2} + j \frac{1}{4} \{\delta(F + F_0) - \delta(F - F_0)\}$
24.  $\cos 2\pi F_0 t \longleftrightarrow \frac{1}{2} \{\delta(F + F_0) + \delta(F - F_0)\}$
25.  $\cos 2\pi F_0 t \cdot u(t) \longleftrightarrow \frac{j\Omega}{\Omega_0^2 - \Omega^2} + \frac{1}{4} \{\delta(F + F_0) + \delta(F - F_0)\}$
26.  $\frac{1}{\sqrt{2\pi\sigma^2}} e^{-t^2/2\sigma^2} \longleftrightarrow e^{-(\Omega\sigma)^2/2}$
27.  $e^{-at} \sin 2\pi F_0 t \cdot u(t) \longleftrightarrow \frac{\Omega_0}{(j\Omega + a)^2 + (\Omega_0)^2}$
28.  $e^{-a|t|} \sin 2\pi F_0 |t| \longleftrightarrow \frac{2\Omega_0(\Omega_0^2 + a^2 - \Omega^2)}{(\Omega^2 + a^2 - \Omega_0^2)^2 + 4a^2\Omega_0^2}$
29.  $e^{-at} \cos 2\pi F_0 t \cdot u(t) \longleftrightarrow \frac{j\Omega + a}{(j\Omega + a)^2 + (\Omega_0)^2}$
30.  $e^{-a|t|} \cos 2\pi F_0 |t| \longleftrightarrow \frac{2a(\Omega_0^2 + a^2 + \Omega^2)}{(\Omega^2 + a^2 - \Omega_0^2)^2 + 4a^2\Omega_0^2}$
31.  $\text{rect}(at) = \begin{cases} 1 & \text{for } |t| < \frac{1}{2a} \\ 0 & \text{Else} \end{cases} \longleftrightarrow \frac{1}{a} \text{sinc}\left(\frac{F}{a}\right) \quad a > 0$
32.  $\text{sinc}(at) = \frac{\sin(\pi at)}{\pi at} \longleftrightarrow \frac{1}{a} \text{rect}\left(\frac{F}{a}\right) \quad a > 0$
33.  $\text{rep}_T(w(t)) = \sum_{m=-\infty}^{\infty} w(t - mT) \longleftrightarrow \frac{1}{|T|} \text{comb}_{1/T}(W(F))$
34.  $|T| \text{comb}_T(w(t)) = \sum_{m=-\infty}^{\infty} w(mT) \delta(t - mT) \longleftrightarrow \text{rep}_{1/T}(W(F))$
35.  $\sum_{n=-\infty}^{\infty} c_n \delta(t - nT) \longleftrightarrow \sum_{n=-\infty}^{\infty} \frac{1}{T} c_n \delta\left(F - \frac{n}{T}\right) = \sum c_n e^{-j2\pi nTF}$

## B.3 The Z-transform

### B.3.1 The Z-transform of causal signals

- |     |   |   |
|-----|---|---|
| 1.  | $\mathcal{X}(z) = Z[x(n)] = \sum_{n=-\infty}^{\infty} x(n)z^{-n}$   | Transform                                   |
| 2.  | $x(n) = Z^{-1}[\mathcal{X}(z)] = \frac{1}{2\pi j} \int_{\Gamma} \mathcal{X}(z)z^{n-1}dz$  | Invers transform                            |
| 3.  | $\sum_{\nu} a_{\nu}x_{\nu}(n) \longleftrightarrow \sum_{\nu} a_{\nu}\mathcal{X}_{\nu}(z)$   | Linearity                                   |
| 4.  | $x(n - n_0) \longleftrightarrow z^{-n_0}\mathcal{X}(z)$   | Shift ( $n_0$ positive or negative integer) |
| 5.  | $nx(n) \longleftrightarrow -z \frac{d}{dz} \mathcal{X}(z)$  | Multiplication with $n$                     |
| 6.  | $a^n x(n) \longleftrightarrow \mathcal{X}\left(\frac{z}{a}\right)$  | Scaling                                     |
| 7.  | $x(-n) \longleftrightarrow \mathcal{X}\left(\frac{1}{z}\right)$   | Reflection of the time sequence             |
| 8.  | $\left[\sum_{\ell=-\infty}^n x(\ell)\right] \longleftrightarrow \frac{z}{z-1} \mathcal{X}(z)$   | Summation                                   |
| 9.  | $x * y \longleftrightarrow \mathcal{X}(z) \cdot \mathcal{Y}(z)$   | Convolution                                 |
| 10. | $x(n) \cdot y(n) \longleftrightarrow \frac{1}{2\pi j} \int_{\Gamma} \mathcal{Y}(\xi)\mathcal{X}\left(\frac{z}{\xi}\right) \xi^{-1}d\xi$ | Product                                     |
| 11. | $x(0) = \lim_{z \rightarrow \infty} \mathcal{X}(z)$ (if the limit value exist)  | Initial value theorem                       |
| 12. | $\lim_{n \rightarrow \infty} x(n) = \lim_{z \rightarrow 1} (z-1)\mathcal{X}(z)$<br>(if the unit circle is included in the ROC)          | End value theorem                           |
| 13. | $\sum_{\ell=-\infty}^{\infty} x(\ell)y(\ell) = \frac{1}{2\pi j} \int_{\Gamma} x(z)y\left(\frac{1}{z}\right) z^{-1}dz$                   | Parseval's teorem for real time sequences   |
| 14. | $\sum_{\ell=-\infty}^{\infty} x^2(\ell) = \frac{1}{2\pi j} \int_{\Gamma} \mathcal{X}(z)\mathcal{X}(z^{-1})z^{-1}dz$                     | - " -                                       |

Sequence	$\longleftrightarrow$	Transform
$x(n)$	$\longleftrightarrow$	$\mathcal{X}(z)$
15. $\delta(n)$	$\longleftrightarrow$	1
16. $u(n)$	$\longleftrightarrow$	$\frac{1}{1 - z^{-1}}$
17. $nu(n)$	$\longleftrightarrow$	$\frac{z^{-1}}{(1 - z^{-1})^2}$
18. $\alpha^n u(n)$	$\longleftrightarrow$	$\frac{1}{1 - \alpha z^{-1}}$
19. $(n + 1)\alpha^n u(n)$	$\longleftrightarrow$	$\frac{1}{(1 - \alpha z^{-1})^2}$
20. $\frac{(n + 1)(n + 2) \dots (n + r - 1)}{(r - 1)!} \alpha^n u(n)$	$\longleftrightarrow$	$\frac{1}{(1 - \alpha z^{-1})^r}$
21. $\alpha^n \cos \beta n u(n)$	$\longleftrightarrow$	$\frac{1 - z^{-1} \alpha \cos \beta}{1 - z^{-1} 2\alpha \cos \beta + \alpha^2 z^{-2}}$
22. $\alpha^n \sin \beta n u(n)$	$\longleftrightarrow$	$\frac{z^{-1} \alpha \sin \beta}{1 - z^{-1} 2\alpha \cos \beta + \alpha^2 z^{-2}}$
23. $\mathbf{F}^n u(n)$	$\longleftrightarrow$	$(\mathbf{I} - z^{-1} \mathbf{F})^{-1}$

### B.3.2 Single-Sided Z-transform of non-causal signals

Notation

$\mathcal{X}^+(z) = \sum_{n=0}^{\infty} x(n)z^{-n}$  Single-sided z-transform,  $x(n)$  not necessarily causal.

$\mathcal{X}(z) = \mathcal{X}^+(z)$  For causal signals

Shifting  $x(n)$  yields:

i) one step shift

$$x(n - 1) \longleftrightarrow z^{-1} \mathcal{X}^+(z) + x(-1)$$

$$x(n + 1) \longleftrightarrow z \mathcal{X}^+(z) - x(0) \cdot z$$

ii)  $n_0$  step shift ( $n_0 \geq 0$ )

$$x(n - n_0) \longleftrightarrow z^{-n_0} \mathcal{X}^+(z) + x(-1)z^{-n_0+1} + x(-2)z^{-n_0+2} + \dots + x(-n_0)$$

$$x(n + n_0) \longleftrightarrow z^{n_0} \mathcal{X}^+(z) - x(0)z^{n_0} - x(1)z^{n_0-1} - \dots - x(n_0 - 1)z$$

## B.4 Fourier Transform for Discrete Time Signal

1.  $X(f) = \mathcal{F}(x(n)) = \sum_{\ell=-\infty}^{\infty} x(\ell)e^{-j2\pi f\ell} \quad \omega = 2\pi f$  Transform
2.  $x(n) = \int_{-1/2}^{1/2} X(f)e^{j2\pi fn} df = \frac{1}{2\pi} \int_{-\pi}^{\pi} X(f)e^{j\omega n} d\omega$  Inverse transform
3.  $\sum a_{\nu}x_{\nu}(n) \longleftrightarrow \sum_{\nu} a_{\nu}X_{\nu}(f)$  Linearity
4.  $x(n - n_0) \longleftrightarrow X(f) \cdot e^{-j2\pi fn_0}$  Shift
5.  $x(n)e^{j2\pi f_0 n} \longleftrightarrow X(f - f_0)$  Frequency translation
6.  $x(n) \cdot \cos 2\pi f_0 n \longleftrightarrow \frac{1}{2} [X(f - f_0) + X(f + f_0)]$   
Modulation
7.  $x(n) \cdot \sin 2\pi f_0 n \longleftrightarrow \frac{1}{2j} [X(f - f_0) - X(f + f_0)]$   
Modulation
8.  $x * y \longleftrightarrow X(f) \cdot Y(f)$  Convolution
9.  $x \cdot y \longleftrightarrow \int_{-1/2}^{1/2} X(\lambda) \cdot Y(f - \lambda) d\lambda$  Product
10.  $\sum_{\ell=-\infty}^{\infty} x(\ell)y(\ell) = \int_{-1/2}^{1/2} X(f)Y^*(f)df$  Parseval's teorem  
for real time sequences
11.  $X(f) = \mathcal{X}(e^{j\omega})$  if  $x(n) = 0$  for  $n < n_0$  and  $\sum_{\ell=-\infty}^{\infty} |x(\ell)|^2 < \infty$   
(Valid for for example: 18,19,20,21 och 22 in the Z-transform table for  $|\alpha| < 1$ )
12.  $\delta(n) \longleftrightarrow 1$
13.  $\delta(n - n_0) \longleftrightarrow e^{-j\omega n_0}$
14.  $1 \forall n \longleftrightarrow \sum_{p=-\infty}^{\infty} \delta(f - p)$
15.  $u(n) \longleftrightarrow \frac{1}{2} \sum_{p=-\infty}^{\infty} \delta(f - p) + \frac{1}{2} + \frac{1}{j \cdot 2 \cdot \tan(\pi f)}$

16.  $2f_1 \cdot \text{sinc}(2f_1 \cdot n) = 2f_1 \frac{\sin(2\pi f_1 n)}{2\pi f_1 n}$
- $$\longleftrightarrow \text{rect}_p\left(\frac{f}{2f_1}\right) = \begin{cases} 1 & |f - n| < f_1 < 1/2, n \text{ integer} \\ 0 & \text{Else} \end{cases}$$
- Ideal LP filter
17.  $4f_1 \text{sinc}(2f_1 n) \cos(2\pi f_0 n)$
- $$\longleftrightarrow \text{rect}_p\left(\frac{f - f_0}{2f_1}\right) + \text{rect}_p\left(\frac{f + f_0}{2f_1}\right) \text{ Ideal BP Filter}$$
18.  $\frac{2\pi f_1 n \cos 2\pi f_1 n - \sin 2\pi f_1 n}{\pi n^2}$
- $$\longleftrightarrow (j2\pi f)_p = \begin{cases} j2\pi(f - n) & |f - n| < f_1 < 1/2, n \text{ integer} \\ 0 & \text{Else} \end{cases}$$
- “Derivating” system
19.  $\cos(2\pi f_0 n) \longleftrightarrow \frac{1}{2} \sum_{p=-\infty}^{\infty} [\delta(f - f_0 - p) + \delta(f + f_0 - p)]$
20.  $\alpha^{|n|} \longleftrightarrow \frac{1 - \alpha^2}{1 + \alpha^2 - 2\alpha \cos 2\pi f}$
21.  $\alpha^{|n|} \cos(2\pi f_0 n)$
- $$\longleftrightarrow \frac{1 - \alpha^2}{2} \left[ \frac{1}{1 + \alpha^2 - 2\alpha \cos 2\pi(f + f_0)} + \frac{1}{1 + \alpha^2 - 2\alpha \cos 2\pi(f - f_0)} \right]$$
22.  $p_r(n) = \begin{cases} 1 & |n| \leq \frac{M-1}{2} \\ 0 & \text{Else} \end{cases} \quad M \text{ odd}$
- $$\longleftrightarrow P_r(f) = \frac{\sin(\pi f M)}{\sin(\pi f)} \text{ Rectangular window}$$

## B.5 Some DFT Properties

Time	Frequency
$x(n), y(n)$	$X(k), Y(k)$
$x(n) = x(n + N)$	$X(k) = X(k + N)$
$x(N - 1)$	$X(N - k)$
$x((n - 1))_{<N>}$	$X(k)e^{-j2\pi k1/N}$
$x(n)e^{j2\pi 1n/N}$	$X((k - 1))_{<N>}$
$x^*(n)$	$X^*(N - k)$
$x_1(n) \otimes x_2(n)$	$X_1(k)X_2(k)$
$x(n) \otimes y^*(-n)$	$X(k)Y^*(k)$
$x_1(n)x_2(n)$	$\frac{1}{N} X_1(k) N X_2(k)$
$\sum_{n=0}^{N-1} x(n)y^*(n)$	$\frac{1}{N} \sum_{k=0}^{N-1} X(k)Y^*(k)$