2.161 Signal Processing: Continuous and Discrete Fall 2008

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# MASSACHUSETTS INSTITUTE OF TECHNOLOGY DEPARTMENT OF MECHANICAL ENGINEERING

2.161 Signal Processing - Continuous and Discrete Fall Term 2008

# <u>Lecture $13^1$ </u>

### Reading:

- Proakis & Manolakis, Chapter 3 (The z-transform)
- Oppenheim, Schafer & Buck, Chapter 3 (The z-transform)

# **1** Introduction to Time-Domain Digital Signal Processing

Consider a continuous-time filter

$$f(t) \longrightarrow \begin{array}{c} \text{Continuous} \\ \text{system} \\ (h(t), H(s)) \end{array} \longrightarrow y(t)$$

such as simple first-order RC high-pass filter:



described by a transfer function

$$H(s) = \frac{RCs}{RCs+1}.$$

The ODE describing the system is

$$\tau \frac{\mathrm{d}y}{\mathrm{d}t} + y = \tau \frac{\mathrm{d}f}{\mathrm{d}t}$$

where  $\tau = RC$  is the time constant.

Our task is to derive a simple discrete-time equivalent of this prototype filter based on samples of the input f(t) taken at intervals  $\Delta T$ .



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If we use a *backwards-difference* numerical approximation to the derivatives, that is

$$\frac{\mathrm{d}x}{\mathrm{d}t} \approx \frac{(x(n\Delta T) - x((n-1)\Delta T))}{\Delta T}$$

and adopt the notation  $y_n = y(n\Delta T)$ , and let  $a = \tau/\Delta T$ ,

$$a(y_n - y_{n-1}) + y_n = a(f_n - f_{n-1})$$

and solving for  $y_n$ 

$$y_n = \frac{a}{1+a}y_{n-1} + \frac{a}{1+a}f_n - \frac{a}{1+a}f_{n-1}$$

which is a first-order *difference equation*, and is the computational formula for a sampleby-sample implementation of digital high-pass filter derived from the continuous prototype above. Note that

- The "fidelity" of the approximation depends on  $\Delta T$ , and becomes more accurate when  $\Delta T \ll \tau$ .
- At each step the output is a linear combination of the present and/or past samples of the output and input. This is a recursive system because the computation of the current output depends on prior values of the output.

In general, regardless of the design method used, a LTI digital filter implementation will be of a similar form, that is

$$y_n = \sum_{i=1}^N a_i y_{n-i} + \sum_{i=0}^M b_i f_{n-i}$$

where the  $a_i$  and  $b_i$  are constant coefficients. Then as in the simple example above, the current output is a weighted combination of past values of the output, and current and past values of the input.

• If  $a_i \equiv 0$  for  $i = 1 \dots N$ , so that

$$y_n = \sum_{i=0}^M b_i f_{n-i}$$

The output is simply a weighted sum of the current and prior inputs. Such a filter is a non-recursive filter with a finite-impulse-response (FIR), and is known as a *moving* average (MA) filter, or an *all-zero* filter.

• If  $b_i \equiv 0$  for  $i = 1 \dots M$ , so that

$$y_n = \sum_{i=0}^N a_i y_{n-i} + b_0 f_n$$

only the current input value is used. This filter is a recursive filter with an infiniteimpulse-response (IIR), and is known as an *auto-regressive* (AR) filter, or an *all-pole* filter. • With the full difference equation

$$y_n = \sum_{i=1}^{N} a_i y_{n-i} + \sum_{i=0}^{M} b_i f_{n-i}$$

the filter is a recursive filter with an infinite-impulse response (IIR), and is known as an *auto-regressive moving-average* (ARMA) filter.

# 2 The Discrete-time Convolution Sum

For a continuous system

$$f(t) \longrightarrow \begin{array}{c} \text{Continuous} \\ \text{system} \\ (h(t), H(s)) \end{array} \longrightarrow y(t)$$

the output y(t), in response to an input f(t), is given by the convolution integral:

$$y(t) = \int_0^\infty f(\tau)h(t-\tau)d\tau$$

where h(t) is the system impulse response.

For a LTI discrete-time system, such as defined by a difference equation, we define the pulse response sequence  $\{h(n)\}$  as the response to a unit-pulse input sequence  $\{\delta_n\}$ , where

$$\delta_n = \begin{cases} 1 & n = 0\\ 0 & \text{otherwise.} \end{cases}$$

$$f_n \rightarrow f_n \rightarrow f_n$$

If the input sequence  $\{f_n\}$  is written as a sum of weighted and shifted pulses, that is

$$f_n = \sum_{k=-\infty}^{\infty} f_k \delta_{n-k}$$

then by superposition the output will be a sequence of similarly weighted and shifted pulse responses

$$y_n = \sum_{k=-\infty}^{\infty} f_k h_{n-k}$$

which defines the *convolution sum*, which is analogous to the convolution integral of the continuous system.

# **3** The *z*-Transform

The z-transform in discrete-time system analysis and design serves the same role as the Laplace transform in continuous systems. We begin here with a parallel development of both the z and Laplace transforms from the Fourier transforms.

#### The Laplace Transform

(1) We begin with causal f(t) and find its Fourier transform (Note that because f(t) is causal, the integral has limits of 0 and  $\infty$ ):

$$F(j\Omega) = \int_0^\infty f(t)e^{-j\Omega t}dt$$

(2) We note that for some functions f(t) (for example the unit step function), the Fourier integral does not converge.

(3) We introduce a weighted function

$$w(t) = f(t)e^{-\sigma t}$$

and note

$$\lim_{\sigma \to 0} w(t) = f(t)$$

The effect of the exponential weighting by  $e^{-\sigma t}$ is to allow convergence of the integral for a much broader range of functions f(t).

(4) We take the Fourier transform of w(t)

$$W(j\Omega) = \tilde{F}(j\Omega|\sigma) = \int_0^\infty \left(f(t)e^{-\sigma t}\right)e^{-j\Omega t}dt$$
$$= \int_0^\infty f(t)e^{-(\sigma+j\Omega)}dt$$

and define the complex variable  $s = \sigma + j\Omega$  so that we can write

$$F(s) = \tilde{F}(j\omega|\sigma) = \int_0^\infty f(t)e^{-st}dt$$

F(s) is the one-sided Laplace Transform. Note that the Laplace variable  $s = \sigma + j\Omega$  is expressed in Cartesian form.

#### The Z transform

(1) We sample f(t) at intervals  $\Delta T$  to produce  $f^*(t)$ . We take its Fourier transform (and use the sifting property of  $\delta(t)$ ) to produce

$$F^*(j\Omega) = \sum_{n=0}^{\infty} f_n e^{-jn\Omega\Delta T}$$

(2) We note that for some sequences  $f_n$  (for example the unit step sequence), the summation does not converge.

(3) We introduce a weighted sequence

$$\{w_n\} = \left\{f_n r^{-n}\right\}$$

and note

$$\lim_{r \to 1} \left\{ w_n \right\} = \left\{ f_n \right\}$$

The effect of the exponential weighting by  $r^{-n}$  is to allow convergence of the summation for a much broader range of sequences  $f_n$ .

(4) We take the Fourier transform of  $w_n$ 

$$W^*(j\Omega) = \tilde{F}^*(j\Omega|r) = \sum_{n=0}^{\infty} (f_n r^{-n}) e^{-jn\Omega\Delta T}$$
$$= \sum_{n=0}^{\infty} f_n (r e^{j\Omega\Delta T})^{-n}$$

and define the complex variable  $z = re^{j\Omega\Delta T}$  so that we can write

$$F(z) = \tilde{F}^*(j\Omega|r) = \sum_{n=0}^{\infty} f_n z^{-n}$$

F(z) is the one-sided Z-transform. Note that  $z = re^{j\Omega\Delta T}$  is expressed in polar form.

#### The Laplace Transform (contd.)

(5) For a causal function f(t), the region of convergence (ROC) includes the *s*-plane to the right of all poles of  $F(j\Omega)$ .



(6) If the ROC includes the imaginary axis, the FT of f(t) is  $F(j\Omega)$ :

$$F(j\Omega) = F(s)|_{s=j\Omega}$$

(7) The convolution theorem states

$$f(t) \otimes g(t) = \int_{-\infty}^{\infty} f(\tau) g(t - \tau) d\tau \stackrel{\mathcal{L}}{\longleftrightarrow} F(s) G(s)$$

(8) For an LTI system with transfer function H(s), the frequency response is

$$H(s)|_{s=j\Omega} = H(j\Omega)$$

if the ROC includes the imaginary axis.

From the above derivation, the Z-transform of a sequence  $\{f_n\}$  is

$$F(z) = \sum_{n = -\infty}^{\infty} f_n z^{-n}$$

where  $z = r e^{j\omega}$  is a complex variable. For a causal sequence  $f_n = 0$  for n < 0, the transform

#### The Z transform (contd.)

(5) For a right-sided (causal) sequence  $\{f_n\}$  the region of convergence (ROC) includes the *z*-plane at a radius greater than all of the poles of F(z).



(6) If the ROC includes the unit circle, the DFT of  $\{f_n\}$ , n = 0, 1, ..., N - 1. is  $\{F_m\}$  where

$$F_m = F(z)|_{z=e^{j\omega_m}} = F(e^{j\omega_m}),$$

where  $\omega_m = 2\pi m/N$  for  $m = 0, 1, \dots, N-1$ . (7) The convolution theorem states

$$\{f_n\} \otimes \{g_n\} = \sum_{m=-\infty}^{\infty} f_m g_{n-m} \stackrel{\mathbb{Z}}{\longleftrightarrow} F(z)G(z)$$

(8) For a discrete LSI system with transfer function H(z), the frequency response is

$$H(z)|_{z=e^{j\omega}} = H(e^{j\omega}) \quad |\omega| \le \pi$$

if the ROC includes the unit circle.

can be written

$$F(z) = \sum_{n=0}^{\infty} f_n z^{-r}$$

**Example:** The finite sequence  $\{f_0, \ldots, f_3\} = \{5, 3, -1, 4\}$  has the z-transform

$$F(z) = 5z^0 + 3z^{-1} - z^{-2} + 4z^{-3}$$

**The Region of Convergence:** For a given sequence, the region of the *z*-plane in which the sum converges is defined as the *region of convergence* (ROC). In general, within the ROC

$$\sum_{n=-\infty}^{\infty} \left| f_n r^{-n} \right| < \infty$$

and the ROC is in general an annular region of the z-plane:



- (a) The ROC is a ring or disk in the z-plane.
- (b) The ROC cannot contain any poles of F(z).
- (c) For a finite sequence, the ROC is the entire z-plane (with the possible exception of z = 0 and  $z = \infty$ .
- (d) For a causal sequence, the ROC extends outward from the outermost pole.
- (e) for a left-sided sequence, the ROC is a disk, with radius defined by the innermost pole.
- (f) For a two sided sequence the ROC is a disk bounded by two poles, but not containing any poles.
- (g) The ROC is a connected region.

*z*-Transform Examples: In the following examples  $\{u_n\}$  is the unit step sequence,

$$u_n = \begin{cases} 0 & n < 0\\ 1 & n \ge 0 \end{cases}$$

and is used to force a causal sequence.

13 - 6

(1)  $\{f_n\} = \{\delta_n\}$  (the digital pulse sequence) From the definition of F(z):

$$F(z) = 1z^0 = 1 \qquad \text{for all } z.$$

(2)  $\{f_n\} = \{a^n u_n\}$ 

$$F(z) = \sum_{n=0}^{\infty} a^n z^{-n} = \sum_{n=0}^{\infty} \left(az^{-1}\right)^n$$
$$\boxed{\{a^n\} \stackrel{\mathcal{Z}}{\longleftrightarrow} F(z) = \frac{1}{1 - az^{-1}} = \frac{z}{z - a} \quad \text{for } |z| > a.}$$

since

$$\sum_{n=0}^{\infty} x^n = \frac{1}{1-x} \quad \text{for } x < 1.$$

(3)  $\{f_n\} = \{u_n\}$  (the unit step sequence).

$$F(z) = \sum_{n=0}^{\infty} z^{-n} = \frac{1}{1 - z^{-1}} = \frac{z}{z - 1} \quad \text{for } |z| < 1$$

from (2) with a = 1.

(4) 
$$\{f_n\} = \{e^{-bn}u_n\}.$$
  

$$F(z) = \sum_{n=0}^{\infty} e^{-bn}z^{-n} = \sum_{n=0}^{\infty} (e^{-b}z^{-1})^n$$

$$\left\{e^{-bn}\} \stackrel{\mathcal{Z}}{\longleftrightarrow} F(z) = \frac{1}{1 - e^{-b}z^{-1}} = \frac{z}{z - e^{-bn}} \quad \text{for } |z| > e^{-b}.$$

from (2) with  $a = e^{-b}$ .

(5)  $\{f_n\} = \{ e^{-b|n|} \}.$ 

$$F(z) = \sum_{n=-\infty}^{0} (e^{-b}z)^{-n} + \sum_{n=0}^{\infty} (e^{-b}z^{-1})^{n} - 1$$
$$= \frac{1}{1 - e^{-b}z} + \frac{1}{1 - e^{-b}z^{-1}} - 1$$

Note that the item  $f_0 = 1$  appears in each sum, therefore it is necessary to subtract 1.

$$\{ e^{-b|n|} \} \xrightarrow{\mathcal{Z}} F(z) = \frac{1 - e^{-2b}}{(1 - e^{-b}z)(1 - e^{-b}z^{-1})} \quad \text{for } e^{-b} < |z| < e^{b}.$$



(6) 
$$\{f_n\} = \{ e^{-j\omega_0 n} u_n \} = \{ \cos(\omega_0 n) u_n \} - j \{ \sin(\omega_0 n) u_n \} .$$
$$F(z) = \mathcal{Z} \{ \cos(\omega_0 n) u_n \} - j \mathcal{Z} \{ \sin(\omega_0 n) u_n \} \}$$

From (1)

$$F(z) = \frac{1}{1 - e^{-j\omega_0} z^{-1}} \quad \text{for } |z| > 1$$
  
=  $\frac{1 - \cos(\omega_0) z^{-1} - j\sin(\omega_0)}{1 - 2\cos(\omega_0) z_* - 1 + z^{-2}}$   
=  $\frac{z^2 - \cos(\omega_0) z - j\sin(\omega_0) z^2}{z^2 - 2\cos(\omega_0) z + 1}$ 

and therefore

$$\mathcal{Z}\left\{\cos(\omega_0 n)u_n\right\} = \frac{z^2 - \cos(\omega_0)z}{z^2 - 2\cos(\omega_0)z + 1} \quad \text{for } |z| > 1$$
$$\mathcal{Z}\left\{\sin(\omega_0 n)u_n\right\} = \frac{\sin(\omega_0)z^2}{z^2 - 2\cos(\omega_0)z + 1} \quad \text{for } |z| > 1$$

**Properties of the z-Transform:** Refer to the texts for a full description. We simply summarize some of the more important properties here.

### (a) Linearity:

$$a\{f_n\} + b\{g_n\} \xleftarrow{\mathcal{Z}} aF(z) + bG(z)$$
 ROC: Intersection of ROC<sub>f</sub> and ROC<sub>g</sub>.

(b) Time Shift:

$$\{f_{n-m}\} \stackrel{\mathbb{Z}}{\longleftrightarrow} z^{-m} F(z) \qquad \text{ROC: } \operatorname{ROC}_f \text{ except for } z = 0 \text{ if } k < 0, \text{ or } z = \infty \text{ if } k > 0.$$
  
If  $g_n = f_{n-m}$ ,

$$G(z) = \sum_{n=-\infty}^{\infty} f_{n-m} z^{-n} = \sum_{k=-\infty}^{\infty} f_k z^{-(k+m)} = z^{-m} F(z).$$

This is an important property in the analysis and design of discrete-time systems. We will often have recourse to a unit-delay block:

$$f_n \longrightarrow$$
 Unit Delay  $y_n = f_{n-1}$ 

### (c) Convolution:

$$\{f_n\} \otimes \{g_n\} \xleftarrow{\mathcal{Z}} F(z)G(z)$$
 ROC: Intersection of  $\operatorname{ROC}_f$  and  $\operatorname{ROC}_g$ .

where  $\{f_n\} \otimes \{g_n\} = \sum_{k=-\infty}^{\infty} f_k g_{n-k}$  is the convolution sum.

Let

$$Y(z) = \sum_{n=-\infty}^{\infty} y_n z^{-n} = \sum_{n=-\infty}^{\infty} \left( \sum_{k=-\infty}^{\infty} f_k g_{n-k} \right) z^{-n}$$
  
$$= \sum_{k=-\infty}^{\infty} f_k \left( \sum_{n=-\infty}^{\infty} g_{n-k} z^{-(n-k)} \right) z^{-k} = \sum_{k=-\infty}^{\infty} f_k z^{-k} \sum_{m=-\infty}^{\infty} g_m z^{-m}$$
  
$$= F(z)G(z)$$

(d) Conjugation of a complex sequence:

$$\left\{\overline{f}_n\right\} \xleftarrow{\mathcal{Z}} \overline{F}(z) \qquad \text{ROC: } \operatorname{ROC}_f$$

(e) Time reversal:

$$\{f_{-n}\} \xleftarrow{\mathcal{Z}} F(1/z) \qquad \text{ROC:} \quad \frac{1}{r_1} < |z| < \frac{1}{r_2}$$

where the ROC of F(z) lies between  $r_1$  and  $r_2$ .

(e) Scaling in the z-domain:

$$\left| \{a^n f_n\} \xleftarrow{\mathcal{Z}} F(a^{-1}z) \qquad \text{ROC:} \quad |a| \, r_1 < |z| < |a| \, r_2 \right|$$

where the ROC of F(z) lies between  $r_1$  and  $r_2$ .

### (e) Differentiation in the z-domain:

$$\{nf_n\} \xleftarrow{\mathcal{Z}} -z \frac{\mathrm{d}F(z)}{\mathrm{d}z} \qquad \mathrm{ROC:} \ r_2 < |z| < r_1$$

where the ROC of F(z) lies between  $r_1$  and  $r_2$ .