

Module 1

Sampling and Aliasing, System Functions and z-transforms

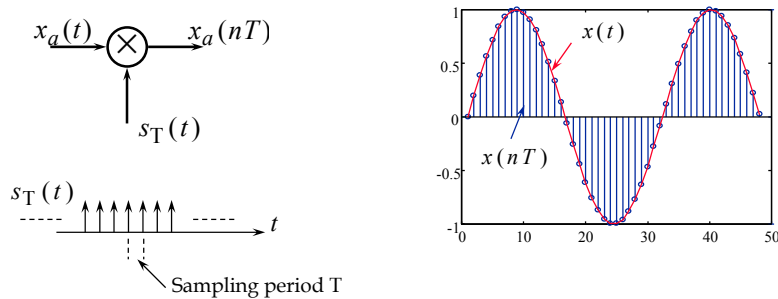
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Contents

- ◆ Sampling and Aliasing
 - Concept of sampling a continuous-time signal
 - Periodic nature of discrete-time signals in the frequency domain, the effect of aliasing and Nyquist sampling criterion
 - Reconstruction of continuous-time signals from their samples
- ◆ System Functions
 - Transfer function
 - Frequency response
 - Example
- ◆ z-transforms
 - Definition and properties
 - Inverse z-transform
 - Causal and anticausal systems
 - Stability

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Sampling



- ◆ Analog signal $x_a(t)$
- ◆ Sampling function $s_T(t)$ is a sequence of impulses

$$s_T(t) = \sum_{m=-\infty}^{\infty} \delta(t - mT)$$

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Symbols

- | | |
|-------------------------------------|------------------|
| ◆ continuous time signal | $x_a(t)$ |
| ◆ samples of continuous time signal | $x_a(nT)$ |
| ◆ discrete-time signal | $x(n)$ |
| ◆ frequency | Ω |
| ◆ digital frequency | ω |
| ◆ FT of $x_a(t)$ | $X_a(\Omega)$ |
| ◆ DFT of $x(n)$ | $X(e^{j\omega})$ |

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Analysis of Sampled Signal

- ◆ Spectrum of discrete-time signal = spectrum of continuous-time signal + images at multiples of 2π

- From *IFT*:

$$x(n) = x_a(nT) = \frac{1}{2\pi} \int_{-\infty}^{\infty} X_a(j\Omega) e^{j\Omega nT} d\Omega$$

- From *IDFT*:

$$x(n) = \frac{1}{2\pi} \int_{-\pi}^{\pi} X(e^{j\omega}) e^{j\omega n} d\omega$$

- Combining these:

$$X(e^{j\omega}) = \frac{1}{T} \sum_{r=-\infty}^{\infty} X_a\left(j\left(\frac{\omega}{T} + \frac{2\pi r}{T}\right)\right)$$

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◆ Proof

- write expression for $x_a(nT)$ as sum of integrals over intervals of length $2\pi/T$

$$x(n) = \frac{1}{2\pi} \sum_{r=-\infty}^{\infty} \int_{(2r-1)\pi/T}^{(2r+1)\pi/T} X_a(j\Omega) e^{j\Omega nT} d\Omega$$

- change of variable: replace Ω with $\Omega' + 2\pi r/T$

$$x(n) = \frac{1}{2\pi} \sum_{r=-\infty}^{\infty} \int_{-\pi/T}^{\pi/T} X_a\left(j\left(\Omega' + \frac{2\pi r}{T}\right)\right) e^{j\Omega' nT} e^{j2\pi r n} d\Omega'$$

- use $e^{j2\pi r n} = 1 \quad \forall (r, n)$ integer, reverse order of sum and integration and use $\Omega' = \omega/T$

$$x(n) = \frac{1}{2\pi} \int_{-\pi}^{\pi} \left[\frac{1}{T} \sum_{r=-\infty}^{\infty} X_a\left(j\left(\frac{\omega}{T} + \frac{2\pi r}{T}\right)\right) \right] e^{j\omega n} d\omega$$

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$$x(n) = \frac{1}{2\pi} \int_{-\pi}^{\pi} \left[\frac{1}{T} \sum_{r=-\infty}^{\infty} X_a\left(j\left(\frac{\omega}{T} + \frac{2\pi r}{T}\right)\right) \right] e^{j\omega n} d\omega$$

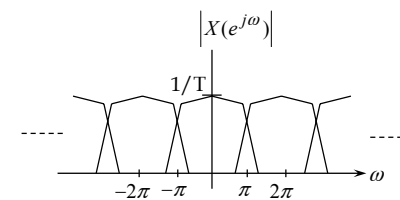
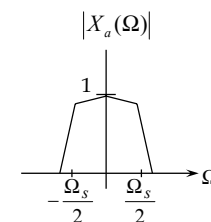
- Note that this is in the form of an IDFT since
- Hence

$$X(e^{j\omega}) = \frac{1}{T} \sum_{r=-\infty}^{\infty} X_a\left(j\left(\frac{\omega}{T} + \frac{2\pi r}{T}\right)\right) \quad [1]$$

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◆ Points to note:

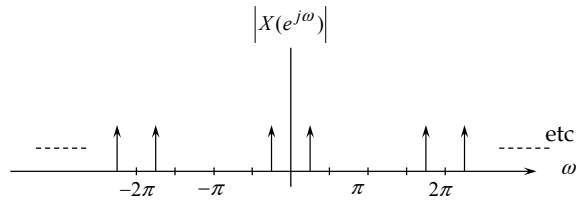
- spectrum of $x(n)$ is periodic in ω with period 2π
- if $X_a(\Omega)$ is not bandlimited to $\Omega_s/2$ then information in the signal is lost when sampled due to overlapping spectral images - this effect is called aliasing
- if $X_a(j\Omega)$ is bandlimited to $\Omega_s/2$ then the original continuous-time signal can be perfectly reconstructed from its discrete-time samples
 - this is known as the Nyquist Sampling Criterion
- Ω is the analog frequency, $\Omega = 2\pi f \quad 0 < \Omega < \infty$
- ω is the digital frequency $\omega = \Omega T = 2\pi f T = \frac{2\pi f}{f_s}$



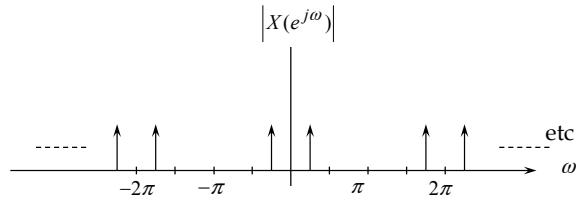
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◆ Examples of signal spectra after sampling

- 1) sinusoidal signal at 1 kHz, sampling frequency = 8 kHz

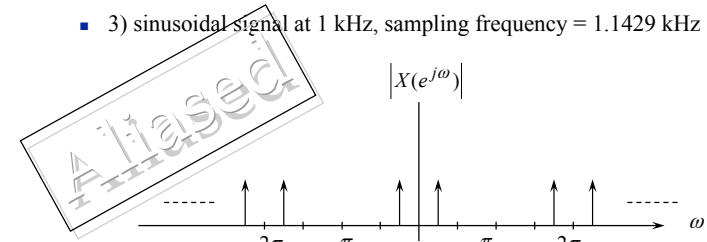


- 2) sinusoidal signal at 5.5125 kHz, sampling frequency = 44.1 kHz



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- 3) sinusoidal signal at 1 kHz, sampling frequency = 1.1429 kHz



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- ◆ Sampling in time domain => periodicity in frequency

- ◆ Sampling in frequency domain => periodicity in time

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Signal Reconstruction

- ◆ A continuous-time signal $x_a(t)$ can be reconstructed from its samples $\{x_a(nT)\}$ as

$$x_a(t) = \sum_{n=-\infty}^{\infty} x_a(nT)g(t-nT)$$

- where $g(t) = \frac{\sin(\pi t/T)}{\pi t/T}$

and corresponds to a lowpass filter with cut-off at the Nyquist frequency.

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◆ Proof

- write the *IFT* expression for $x_a(t)$ for the range $-\pi/T \leq \Omega \leq \pi/T$ or equivalently $-\pi \leq \omega \leq \pi$

$$x_a(t) = \frac{1}{2\pi} \int_{-\pi/T}^{\pi/T} X_a(\Omega) e^{j\Omega t} d\Omega$$

- from [1] we know that, in the range $-\pi \leq \omega \leq \pi$

$$X(e^{j\omega}) = X(e^{j\Omega T}) = \frac{1}{T} X_a(\Omega)$$

- giving

$$x_a(t) = \frac{1}{2\pi} \int_{-\pi/T}^{\pi/T} TX(e^{j\Omega T}) e^{j\Omega t} d\Omega$$

◆ Proof (continued)

- using the DTFT relation (described later)

$$X(e^{j\Omega T}) = \sum_{n=-\infty}^{\infty} x_a(nT) e^{-j\Omega nT}$$

- write

$$x_a(t) = \frac{T}{2\pi} \int_{-\pi/T}^{\pi/T} \left[\sum_{n=-\infty}^{\infty} x_a(nT) e^{-j\Omega nT} \right] e^{j\Omega t} d\Omega$$

- change order of summation and integration

$$\begin{aligned} x_a(t) &= \sum_{n=-\infty}^{\infty} x_a(nT) \left[\frac{T}{2\pi} \int_{-\pi/T}^{\pi/T} e^{j\Omega(t-nT)} d\Omega \right] \\ &= \sum_{n=-\infty}^{\infty} x_a(nT) \frac{\sin\left(\frac{\pi}{T}(t-nT)\right)}{\frac{\pi}{T}(t-nT)} \end{aligned}$$

◆ Proof (continued)

- This operation can be recognized as the convolution of $x_a(nT)$ with the sinc function

$$\frac{\sin(\pi/T)}{\pi/T}$$

- This convolution represents filtering with an “ideal” lowpass filter with a cut-off frequency of $\omega = \pi$
 - the Nyquist frequency

Reading: Proakis: Chapter 1, especially 1.4.1 to 1.4.7

System Functions

◆ Transfer function

- For a continuous-time system $H(s)$ with input $X(s)$ and output $Y(s)$,

its transfer function is defined as $H(s) = \frac{Y(s)}{X(s)}$

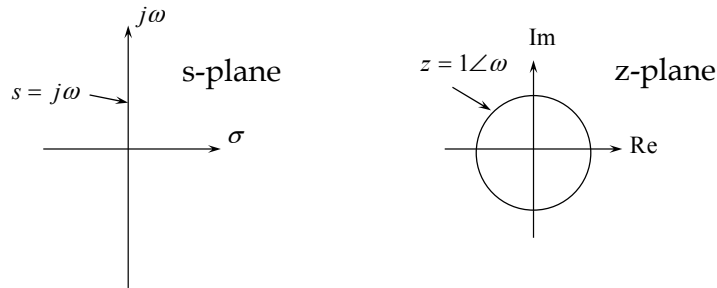
- For a discrete-time system $H(z)$ with input $X(z)$ and output $Y(z)$,

its transfer function is defined as $H(z) = \frac{Y(z)}{X(z)}$

- $H(\cdot)$, $Y(\cdot)$ and $X(\cdot)$ are polynomials in (\cdot)

◆ Frequency response

- For continuous-time systems, use $s = \sigma + j\omega$ and investigate the function $H(s)$ as a function of frequency ω only, i.e. write $s = j\omega$
- For discrete-time systems, use $z = e^{sT}$, $s = \alpha + j\omega$ and investigate the function $H(z)$ as a function of frequency ω only, i.e. write $z = e^{j\omega}$

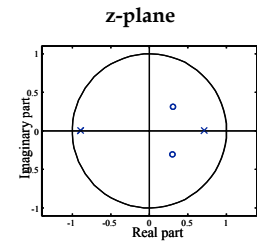


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◆ Example:

- Transfer Function

$$H(z) = \frac{z^2 - 0.6z + 0.18}{z^2 + 0.2z - 0.63}$$



- zeros at $z = 0.3 + j0.3$ and $z = 0.3 - j0.3$
- poles at $z = -0.9$ and $z = 0.7$

- Can be written in terms of z^{-1} as $H(z) = \frac{1 - 0.6z^{-1} + 0.18z^{-2}}{1 + 0.2z^{-1} - 0.63z^{-2}}$

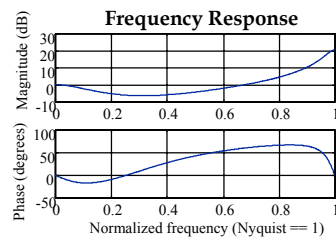
- Difference Equation:

$$y(n) = x(n) - 0.6x(n-1) + 0.18x(n-2) - 0.2y(n-1) + 0.63y(n-2)$$

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■ Frequency Response

- set $z = e^{j\omega}$
- plot magnitude and phase



- normally plot for $0 < \omega < \pi$ normalized such that $\pi = 1$

Reading: Proakis, Chapter 2 especially 2.4 and 2.5

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z-transform

◆ Definition

- The z-transform of the sequence $x(n)$ is given by

$$X(z) = \sum_{n=-\infty}^{\infty} x(n)z^{-n}$$

- The z domain for discrete-time signals is analogous to the s domain for continuous-time signals
- the z domain allows a signal (or system) to be compactly described

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Notation

- ◆ Z-transform denoted by

$$X(z) \equiv Z\{x(n)\}$$

- ◆ relationship indicated by

$$x(n) \xleftrightarrow{z} X(z)$$

Region of Convergence

$$X(z) = \sum_{n=-\infty}^{\infty} x(n)z^{-n}$$

- ◆ z-transform is an infinite power series
 - only exists for particular values of z for which the series converges
 - these are the values of z for which $X(z)$ has a finite value
- ◆ need to specify Region Of Convergence (ROC) when referring to z-transform

Examples

- ◆ Finite Duration Sequences

$$x(n) = \{1, 2, 3, 5, 8\} \quad X(z) = 1 + 2z^{-1} + 3z^{-2} + 5z^{-3} + 8z^{-4}$$

ROC: $z \neq 0$

$$x(n) = \{1, 2, 3, 5, 8\}$$

↑

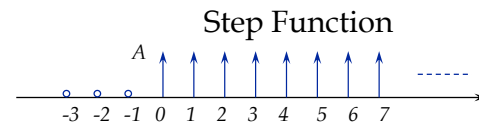
$$X(z) = z^2 + 2z^1 + 3 + 5z^{-1} + 8z^{-2}$$

ROC: $z \neq 0$ and $z \neq \infty$

$$x(n) = \delta(n) \quad X(z) = 1$$

ROC: $\forall z$

- ◆ Infinite Duration Sequences



$$x(n) = Au(n)$$

$$X(z) = \sum_{n=-\infty}^{\infty} Au(n)z^{-n}$$

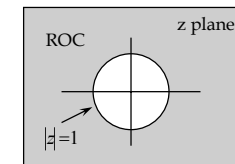
$$= A(1 + z^{-1} + z^{-2} + z^{-3} + \dots)$$

$$= \frac{A}{1 - z^{-1}} \quad \text{ROC: } |z^{-1}| < 1$$

Recall that an infinite geometric series

$$1 + A + A^2 + \dots = \frac{1}{1 - A}$$

if $|A| < 1$



- Sequences like $Au(n)$ are called “right-sided” sequences

◆ Another Example

$$x(n) = Au(n)a^n e^{jn\omega_0}$$

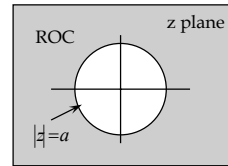
$$\begin{aligned} X(z) &= \sum_{n=-\infty}^{\infty} Au(n)a^n e^{jn\omega_0} z^{-n} \\ &= A \sum_{n=0}^{\infty} (ae^{j\omega_0} z^{-1})^n \\ &= \frac{A}{1 - ae^{j\omega_0} z^{-1}} \end{aligned}$$

- The region of convergence is

$$|ae^{j\omega_0} z^{-1}| < 1$$

or $|z| > |a|$

since $|e^{j\omega_0}| = 1$



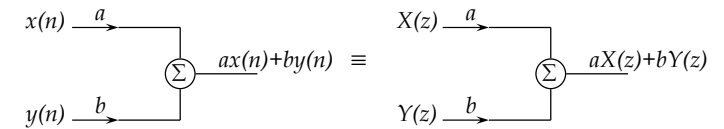
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◆ Properties of the z-transform

- Linearity

- for signals $x(n)$ and $y(n)$ with z-transforms $X(z)$ and $Y(z)$

$$\sum_{k=1}^p C_k f_k(n) \leftrightarrow \sum_{k=1}^p C_k F_k(z)$$



- Shift

$$Z\{f(n-m)\} = z^{-m}F(z)$$

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- Multiplication by n

$$Z\{nf(n)\} = -z \frac{dF(z)}{dz}$$

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

- differentiating both sides gives

$$\frac{dF(z)}{dz} = \sum_{n=-\infty}^{\infty} -nf(n)z^{-(n+1)}$$

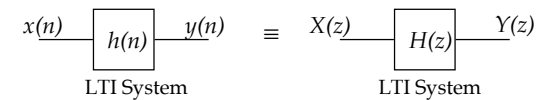
- multiplying both sides by $-z$ gives

$$-z \frac{dF(z)}{dz} = \sum_{n=-\infty}^{\infty} nf(n)z^{-n}$$

- where the RHS can be seen to be the z-transform of $nf(n)$

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- Convolution

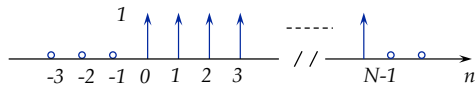


$$y(n) = \sum_{m=-\infty}^{\infty} h(m)x(n-m) \quad Y(z) = H(z)X(z)$$

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◆ Example

- Find the z-transform of the following function $p(n)$



- Write $p(n) = u(n) - u(n - N)$
- Using the shift and linearity properties we obtain

$$P(z) = \frac{1}{1-z^{-1}} - \frac{z^{-N}}{1-z^{-1}} = \frac{1-z^{-N}}{1-z^{-1}}$$

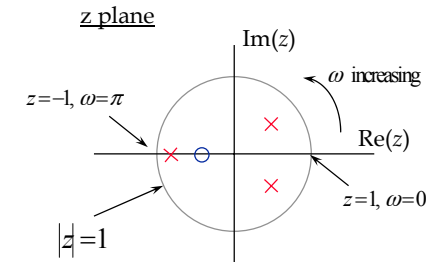
- The ROC is

$$|z^{-1}| < 1$$

◆ Plotting on the z-plane

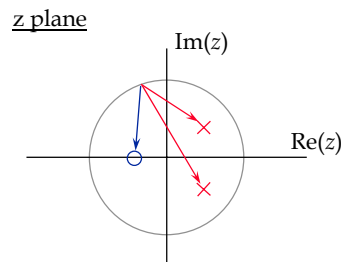
Given
$$H(z) = \frac{b_0 + b_1z^{-1} + b_2z^{-2} + \dots + b_Nz^{-N}}{1 + a_1z^{-1} + a_2z^{-2} + \dots + a_Mz^{-M}}$$

- Poles are roots of denominator \times
- Zeros are roots of numerator \circ



◆ Frequency Response from the z-plane plot

- The frequency response is given by $H(e^{j\omega}) = H(z)|_{z=e^{j\omega}}$
- Can be derived analytically from the transfer function in z
 - put $z = e^{j\omega}$
- Can be derived graphically
 - compute : $\frac{\text{Product of distances to all the zeros}}{\text{Product of distances to all the poles}}$ as z goes around the unit circle



Inverse z-transform

◆ Aim

- Given $X(z)$ find $x(n)$

◆ 4 methods

- Inspection (for power series)
- Long division
- Partial fractions and table look-up
- Inversion formula

Reading: Proakis Section 3.4

◆ Inverse z-transform by inspection

- Given a z-domain expression as a power series

$$X(z) = 1 + 2z^{-1} + 3z^{-2}$$

- use $Z\{A\delta(n-m)\} = Az^{-m}$

- to write $x(n) = \delta(n) + 2\delta(n-1) + 3\delta(n-2)$
 $= \{1, 2, 3\}$

◆ Inverse z-transform by long division

- Given a z-domain expression as a ratio of polynomials, the first few terms of the sequence can be found by long division.

- Start by converting ratio of polynomials to power series, then use inspection

- E.g.

$$X(z) = \frac{0.5z^2 + 0.5z}{z^2 - z + 0.5}$$

$$\begin{array}{r} 0.5 + 1.0z^{-1} + 0.75z^{-2} + \dots \\ z^2 - z + 0.5 \) \ 0.5z^2 + 0.5z \\ \hline 0.5z^2 - 0.5z + 0.25 \\ 0 \quad + 1.0z - 0.25 \\ \hline 1.0z - 1.00 + 0.50z^{-1} \\ 0 \quad + 0.75 - 0.50z^{-1} \\ \hline \text{etc} \end{array}$$

- and hence

$$x(n) = 0.5\delta(n) + 1\delta(n-1) + 0.75\delta(n-2) + \dots$$

$$= \{0.5, 1, 0.75, \dots\}$$

◆ Inverse z-transform by partial fractions and table look-up

- Use tables of standard transform pairs
- Use partial fraction expansion to re-write problem in terms of standard transform pairs
- E.g.

$$X(z) = \frac{4z^2}{z^2 - 0.25}$$

- Use PFE to write

$$X(z) = \frac{2z}{z-0.5} + \frac{2z}{z+0.5}$$

- Use standard transform pair

$$Z\{Aa^n u(n)\} = \frac{Az}{z-a}$$

- to give

$$x(n) = 2(0.5)^n u(n) + 2(-0.5)^n u(n)$$

◆ Inverse z-transform by the inversion formula

- The inverse z-transform is given by

$$x(n) = \frac{1}{2\pi j} \oint_C X(z)z^{n-1} dz$$

- This can be solved using the residue theorem

$$x(n) = \frac{1}{2\pi j} \oint_C X(z)z^{n-1} dz = \sum (\text{residues of } X(z)z^{n-1} \text{ at the poles inside contour } C)$$

- Express $X(z)z^{n-1}$ as $X(z)z^{n-1} = \frac{\varphi(z)}{(z-z_0)^s}$

which has s poles at $z = z_0$

- Then $\text{Res}[X(z)z^{n-1} \text{ at } z = z_0] = \frac{1}{(s-1)!} \left[\frac{d^{s-1} \varphi(z)}{dz^{s-1}} \right]_{z=z_0}$

◆ Example

Find the inverse z-transform of $X(z) = \frac{1}{1-az^{-1}}$ for $|z| > |a|$

Write $x(n) = \frac{1}{2\pi j} \oint_C \frac{z^{n-1}}{1-az^{-1}} dz = \frac{1}{2\pi j} \oint_C \frac{z^n}{z-a} dz$

C is a circular contour of radius greater than a .

Comparing with the form $X(z)z^{n-1} = \frac{\varphi(z)}{(z-z_0)^s}$

gives $s = 1$, $z_0 = a$ and $\varphi(z) = z^n$.

◆ For $n \geq 0$ the only pole of $X(z)z^{n-1}$ is at $z = a$ with a residue of a^n

◆ For $n < 0$ there is a multiple order pole at $z = 0$

For $n = -1$

- residue of pole at origin is $-a^{-1}$
- residue of pole at $z = a$ is a^{-1}



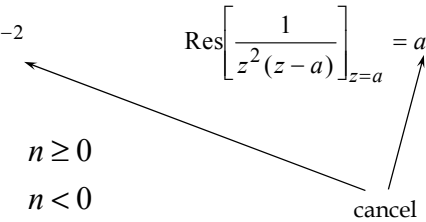
For $n = -2$

- residue of pole at origin is

residue of pole at $z = a$ is

$$\text{Res} \left[\frac{1}{z^2(z-a)} \right]_{z=0} = -a^{-2}$$

$$\text{Res} \left[\frac{1}{z^2(z-a)} \right]_{z=a} = a^{-2}$$



etc.

◆ Therefore $x(n) = \begin{cases} a^n & n \geq 0 \\ 0 & n < 0 \end{cases}$

Causal and Anticausal Systems

◆ Already seen that the ROC is the region of the z-plane for which the infinite sum of the z-transform converges

◆ Given a transfer function $H(z)$ the impulse response $h(n)$ depends on the ROC of $H(z)$

◆ Causal Example $h(n) = \begin{cases} a^n, & n \geq 0 \\ 0, & n < 0 \end{cases} = a^n u(n)$

◆ This has z-transform

$$H(z) = \sum_{n=-\infty}^{\infty} a^n u(n) z^{-n} = \sum_{n=0}^{\infty} (az^{-1})^n$$

$$= \frac{1}{1-az^{-1}} \text{ for } |z| > |a|$$

◆ Anticausal Example $h(n) = \begin{cases} 0, & n \geq 0 \\ -a^n, & n < 0 \end{cases} = -a^n u(-n-1)$

- This has z-transform

$$H(z) = \sum_{n=-\infty}^{\infty} -a^n u(-n-1) z^{-n} = \sum_{n=-\infty}^{-1} (-az^{-1})^n$$

$$= -\frac{1}{1-az^{-1}} \quad \text{for } |z| < |a|$$

◆ Causal and anticausal sequences have same form of z-transforms but different ROCs

◆ Generalisation

- A system with N poles with ROC $|z| > |p_i|$ where p_i is the pole farthest from $z = 0$ is causal.
- A system with N poles with ROC $|z| < |p_i|$ where p_i is the pole nearest to $z = 0$ is anticausal.

Stability

◆ Two equivalent definitions:

- A system $H(z)$ is stable if its inverse z-transform $h(n)$ satisfies

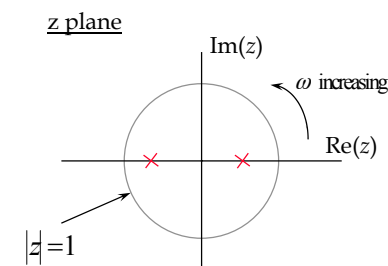
$$\sum_{n=-\infty}^{\infty} |h(n)| < \infty$$

- A system $H(z)$ is stable if its ROC includes the unit circle in the z-plane

- ◆ Causal systems are stable if all poles lie inside the unit circle
- ◆ Anticausal systems are stable if all poles lie outside the unit circle

◆ Example

$$H(z) = \frac{1}{(z-0.5)(z+0.75)} \quad \text{for } |z| > 0.75$$



- ROC lies outside both poles
 - therefore system is causal
- ROC includes unit circle (i.e. modulus of all poles < 1)
 - therefore system is stable

Schur-Cohn Stability Test

- ◆ Write the denominator of the system function as

$$A(z) = 1 + a_1 z^{-1} + a_2 z^{-2} + \dots + a_N z^{-N}$$

- ◆ Convert the polynomial coefficients a_k to reflection coefficients K_m
- ◆ $A(z)$ has roots within the unit circle iff $|K_m| < 1 \quad \forall m$
- ◆ Conversion to reflection coefficients can be done efficiently using a recursive algorithm
 - Levinson/Durbin
 - Uses N^2 multiplications
 - Better than direct factorisation of $A(z)$

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- ◆ Set $a_N(k) = a_k \quad k = 1, 2, \dots, N$ N is order of polynomial
 $K_N = a_N(N)$

- ◆ Then compute for $m = N, N-1, \dots, 1$

$$K_m = a_m(m) \quad a_{m-1}(0) = 1$$

$$b_m(k) = a_m(m-k) \quad k = 0, 1, \dots, m$$

$$a_{m-1}(k) = \frac{a_m(k) - K_m b_m(k)}{1 - K_m^2} \quad k = 1, 2, \dots, m-1$$

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Example

$$A(z) = 1 - 1.75z^{-1} - 0.5z^{-2}$$

$$N = 2$$

$$a_2(1) = -1.75, \quad a_2(2) = -0.5$$

$$K_2 = a_2(2) = -0.5$$

$$m = 2:$$

$$K_2 = a_2(2) = -0.5, \quad a_1(0) = 1$$

$$a_1(1) = \frac{a_2(1) - K_2 a_2(1)}{1 - K_2^2} = \frac{-1.75 - 0.5 * 1.75}{1 - 0.25} = -3.5$$

$$m = 1:$$

$$K_1 = a_1(1) = -3.5, \quad a_0(0) = 1$$

Reading: Proakis Chapter 3

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