

1 Signals and Systems

1.1 Types of Signals

- $\ell_{\mathbb{R}}^1(\mathbb{Z})$ Set of all discrete-time real-valued absolutely summable signals
- $\ell_{\mathbb{C}}^2(\mathbb{Z})$ Set of all discrete-time complex-valued square summable signals
- $\ell_{\mathbb{R}}^{\infty}(\mathbb{Z})$ Set of all discrete-time real-valued bounded signals
- $L_{\mathbb{R}}^1(\mathbb{R})$ Set of all continuous-time real-valued absolutely summable signals
- $L_{\mathbb{C}}^1(\mathbb{R})$ Set of all continuous-time complex-valued square summable signals
- $L_{\mathbb{C}}^{\infty}(\mathbb{R})$ Set of all continuous-time complex-valued bounded signals

$$\ell_{\mathbb{R}}^1 \subset \ell_{\mathbb{R}}^2 \subset \ell_{\mathbb{R}}^{\infty} \quad \ell_{\mathbb{C}}^1 \subset \ell_{\mathbb{C}}^2 \subset \ell_{\mathbb{C}}^{\infty}$$

type of signal	continuous-time	discrete-time
absolutely summable	$\int_{-\infty}^{+\infty} f(x) dx < \infty$	$\sum_{k=-\infty}^{\infty} f[k] < \infty$
square summable	$\int_{-\infty}^{+\infty} f(x) ^2 dx < \infty$	$\sum_{k=-\infty}^{\infty} f[k] ^2 < \infty$
bounded:	$ f(x) < b, \forall x \in \mathbb{R}$	$ f[k] < b, \forall k \in \mathbb{Z}$

1.2 LTI System

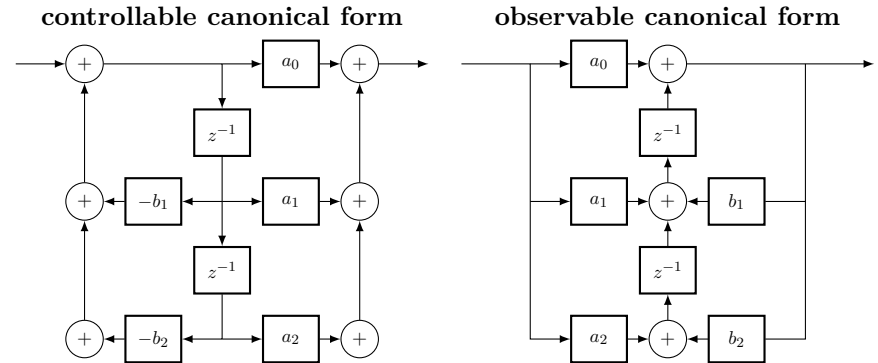
$$\begin{array}{c}
 u[\cdot] \rightarrow \boxed{h[\cdot]} \rightarrow y[\cdot] \\
 y[\cdot] = u[\cdot] * h[\cdot] = \sum_{k \in \mathbb{Z}} u[k] h[\cdot - k] \\
 y(t) = u(t) * h(t) = \int_{-\infty}^{+\infty} u(\tau) h(t - \tau) d\tau
 \end{array}$$

Theorem 1.1 An LTI system is **BIBO-stable**, if and only if its impulse response is a stable signal (i.e. absolutely summable or integrable)

f	g	$f * g$
right-sided	right-sided	right-sided
causal	causal	causal
anything	finite duration	well defined
bounded	absolutely summable	bounded
absolutely summable	absolutely summable	absolutely summable
square summable	absolutely summable	square summable
square summable	square summable	bounded

1.3 canonical forms

$$H(z) = \frac{a_0 + a_1 z^{-1} + a_2 z^{-2} + \dots + a_n z^{-n}}{1 + b_1 z^{-1} + b_2 z^{-2} + \dots + b_n z^{-n}} = \frac{a_0 z^n + a_1 z^{n-1} + \dots + a_n}{z^n + b_1 z^{n-1} + \dots + b_n}$$



$$x[k] \in \mathbb{R}^n, A \in \mathbb{R}^{n \times n}, B \in \mathbb{R}^{n \times 1}, C \in \mathbb{R}^{1 \times n}, D \in \mathbb{R}$$

$$A = \begin{bmatrix} -b_1 & -b_2 & -b_3 & \dots & -b_n \\ 1 & 0 & 0 & \dots & 0 \\ 0 & 1 & 0 & \dots & 0 \\ \vdots & \vdots & \vdots & \ddots & \vdots \\ 0 & 0 & \dots & 1 & 0 \end{bmatrix}, \quad B = \begin{bmatrix} 1 \\ 0 \\ \vdots \\ 0 \end{bmatrix}$$

$$C = [a_1 - a_0 b_1 \quad \dots \quad a_n - a_0 b_n], \quad D = a_0$$

1.4 Mason's Gain Formula

$$\frac{Y(z)}{U(z)} = \frac{\sum_{p \in P} T_p(z) \Delta_p(z)}{\Delta(z)}$$

P is the set of all forward paths from node u to node y . $T_p(z)$ is the transfer function of the forward path p . $\Delta_p(z)$ is the network determinant of that part of the block diagram that is not touched by the forward path p . $\Delta(z)$ is the network determinant.

$$\Delta(z) = 1 - \sum_{s \in S_1} T_s + \sum_{(s_1, s_2) \in S_2} T_{s_1} T_{s_2} - \sum_{(s_1, s_2, s_3) \in S_3} T_{s_1} T_{s_2} T_{s_3} \pm \dots$$

$T_s(z)$ denotes the transfer function of a simple loop s . S_1 is the set of all simple loops. S_2 is the set of all unordered pairs (s_1, s_2) of simple loops $s_1, s_2 \in S_1$ that do not touch. S_3 is defined as three not touching loops and so on...

1.5 z -Transform

Definition 1.1 The z -transform of a real or complex signal $f[\cdot]$ is the function

$$\text{ROC}(f) \rightarrow \mathbb{C} : z \mapsto F(z) := \sum_{k=-\infty}^{\infty} f[k]z^{-k}$$

with $\text{ROC}(f) := \{z \in \mathbb{C} : r_1 < |z| < r_2\}$

$$r_2 := \left(\limsup_{\ell \rightarrow \infty} |f([- \ell])|^{1/\ell} \right)^{-1} \quad r_1 := \limsup_{k \rightarrow \infty} |f[k]|^{1/k}$$

Theorem 1.2 Let $f[\cdot]$ be a real or complex discrete-time signal with rational z -transform. Then $f[\cdot]$ is **stable** $\iff \text{ROC}(f)$ contains the unit circle. If $f[\cdot]$ is right-sided, then $f[\cdot]$ is **stable** \iff all poles of $F(z)$ are strictly inside the unit circle.

Addition $\text{ROC}(f+g) \subseteq \text{ROC}(f) \cap \text{ROC}(g)$

Multiplication $H(z) := F(z)G(z) \longrightarrow \text{ROC}(h) \subseteq \text{ROC}(f) \cap \text{ROC}(g)$

Scalar multipl. $\text{ROC}(af) = \text{ROC}(a)$

Time shift $H(z) := z^m F(z) \longrightarrow \text{ROC}(h) = \text{ROC}(f)$

Time reversal $h[k] := f[-k] \longrightarrow \text{ROC}(h) = \{z \in \mathbb{C} : 1/r_2 < |z| < 1/r_1\}$

$f[k]$	$F(z)$	
$f[k]$	$F(z) = \sum_{k=-\infty}^{\infty} f[k]e^{-k}$	definition
$\alpha f[k] + \beta g[k]$	$\alpha F(z) + \beta G(z)$	linearity
$f[k-m]$	$z^m \cdot F(z)$	time delay
$f[k] * g[k]$	$F(z) \cdot G(z)$	convolution - multiplicity

1.6 Inverse z -Transform

$F(z)$	$\text{ROC}(f)$	$f[\cdot]$
$\frac{z}{z-p}$	$ z > p $	$f[k] = \begin{cases} 0 & k < 0 \\ p^k & k \geq 0 \end{cases}$
	$ z < p $	$f[k] = \begin{cases} -p^k & k < 0 \\ 0 & k \geq 0 \end{cases}$
$\frac{Az}{z-p} + \frac{\bar{A}z}{z-\bar{p}}$	$ z > p $	$f[k] = \begin{cases} 0 & k < 0 \\ 2 A p ^k \cos \Omega k + \varphi & k \geq 0 \end{cases}$
	$ z < p $	$f[k] = \begin{cases} -2 A p ^k \cos \Omega k + \varphi & k < 0 \\ 0 & k \geq 0 \end{cases}$

where $A = |A| \cdot e^{i\varphi}$ and $p = |p| \cdot e^{i\Omega}$

1.7 Spectrum of Discrete-time Signals

Theorem 1.3 The spectrum of a discrete-time signal is the z -transform on the unit circle.

$$F(e^{i\Omega}) := F(z)|_{z=e^{i\Omega}}$$

Theorem 1.4 For a stable real-valued discrete-time signal $f[\cdot] \in \mathbb{R} \forall k \in \mathbb{Z}$:

$$F(e^{-i\omega}) = \overline{F(e^{i\Omega})}$$

$$F(z) = c \cdot \frac{\prod_{k=1}^m (z - z_k)}{\prod_{\ell=1}^n (z - p_\ell)} \quad |F(e^{i\Omega})| = |c| \cdot \frac{\prod_{k=1}^m |e^{i\Omega} - z_k|}{\prod_{\ell=1}^n |e^{i\Omega} - p_\ell|}$$

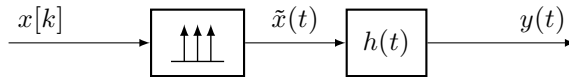
2 Discrete Time \leftrightarrow Continuous Time

2.1 Laplace Transform

$$F(s) := \int_{-\infty}^{+\infty} f(t)e^{-st} \quad \text{ROC}(f) = \{s \in \mathbb{C} : r_1 < \text{Re}(s) < r_2\}$$

signal f	z -transform	Laplace transform
right-sided	$\text{ROC}(f) = \{z \in \mathbb{C} : z > \rho\}$	$\text{ROC}(f) = \{s \in \mathbb{C} : \text{Re}(s) > r\}$
left-sided	$\text{ROC}(f) = \{z \in \mathbb{C} : z < \rho\}$	$\text{ROC}(f) = \{s \in \mathbb{C} : \text{Re}(s) < r\}$
stable	$\text{ROC}(f)$ contains unit circle	$\text{ROC}(f)$ contains imaginary axis

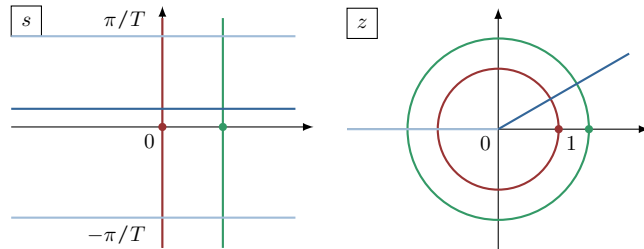
2.2 From Discrete-time to Continuous-time



$$\tilde{x}(t) := \sum_{k=-\infty}^{\infty} x[k]\delta(t - kT) \quad y(t) = (\tilde{x} * h)(t) = \sum_{k=-\infty}^{\infty} x[k]h(t - kT)$$

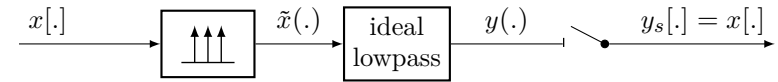
$$\tilde{X}(s) = X(z)|_{z=e^{sT}}, \quad \tilde{X}(i\omega) = F(z)|_{z=e^{i\omega T}}$$

The spectrum $\tilde{F}(i\omega)$ of a “continuous-time signal” $\tilde{f}(t) := \sum_{k \in \mathbb{Z}} f[k]\delta(t - kT)$ equals the **periodic spectrum** $F(e^{i\Omega})$ of the discrete-time signal $f[\cdot]$ with $\Omega = \omega T$



2.3 Ideal Lowpass and the Sampling Theorem

$$H(i\omega) = \begin{cases} 1 & |\omega| < \omega_c \\ 0 & |\omega| \geq \omega_c \end{cases} \quad h(t) = \frac{\sin(\omega_c t)}{\pi t} = \frac{\omega_c}{\pi} \text{sinc}\left(\frac{\omega_c t}{\pi}\right)$$



The system above is completely transparent and acts like a discrete-time filter with impulse response $\delta[\cdot]$

Theorem 2.1 (Nyquist-Shannon Sampling Theorem) Let $y(\cdot)$ be a signal with spectrum $Y(i\omega)$ satisfying

$$Y(i\omega) = 0 \quad \forall |\omega| \geq \pi/T \quad \text{or} \quad Y(i2\pi f) = 0 \quad \forall |f| \geq f_s/2,$$

where $\omega = 2\pi f$ and $f_s := 1/T$. Then, $y(\cdot)$ can be recovered from the discrete-time signal $y_s[k] = Ty(kT)$ by:

$$Y(i\omega) = Y_s(e^{i\omega T}) \quad \forall |\omega| < \pi/T, \quad y(t) = \sum_{k \in \mathbb{Z}} y(kT) \text{sinc}\left(\frac{t - kT}{T}\right).$$

The spectra of $y(\cdot)$ and $y_s[\cdot]$ are related by:

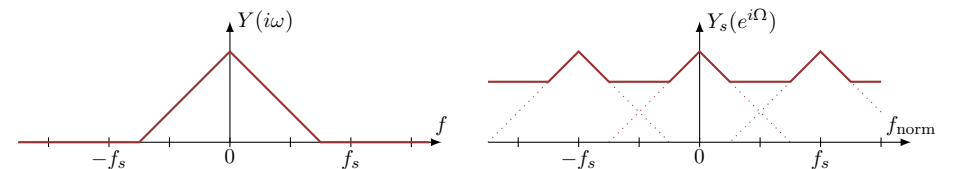
$$Y_s(e^{i\omega T}) = \sum_{n \in \mathbb{Z}} Y\left(i\left(\omega + \frac{2\pi n}{T}\right)\right) \quad Y_s(e^{i2\pi f/f_s}) = \sum_{n \in \mathbb{Z}} Y(i2\pi(f + nf_s))$$

Now, we define the digital frequency and the normalized frequency as following:

$$\Omega = \omega T = 2\pi \frac{f}{f_s}, \quad \Omega \in [-\pi, \pi] \text{ or } [0, 2\pi]$$

$$\frac{\omega}{2\pi} = f_{\text{norm}} = \frac{f}{f_s}, \quad f_{\text{norm}} \in \left[-\frac{1}{2}, \frac{1}{2}\right] \text{ or } [0, 1]$$

The spectrum $Y_s(e^{i\Omega})$ or $\tilde{Y}_s(i\omega)$ of a sampled signal is the sum of copies $Y(i\omega)$ that are shifted by $n \cdot 2\pi/T \quad \forall n \in \mathbb{Z}$.



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