

**Lecture 4**

**DT Processing of CT Signals  
& CT Processing of DT Signals: Fractional Delay**

**Reading:** Sections 4.1 - 4.5 in Oppenheim, Schaffer & Buck (OSB).

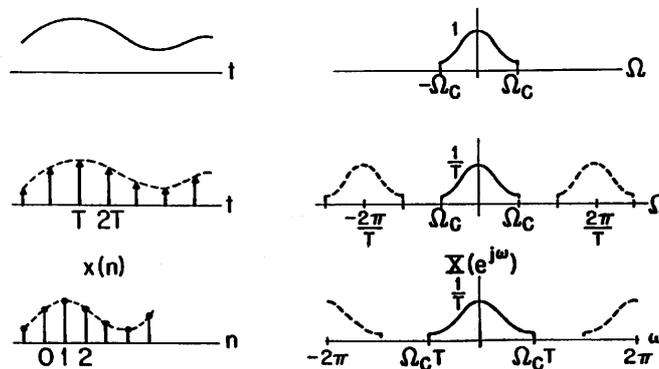
A typical discrete-time system used to process a continuous-time signal is shown in OSB Figure 4.15.  $T$  is the sampling/reconstruction period for the C/D and the D/C converters. When the input signal is bandlimited, the effective system in Figure 4.15 (b) is equivalent to the continuous-time system shown in Figure 4.15 (a).

### Ideal C/D Converter

The relationship between the input and output signals in an ideal C/D converter as depicted in OSB Figure 4.1 is:

$$\begin{aligned} \text{Time Domain:} \quad & x[n] = x_c(nT) \\ \text{Frequency Domain:} \quad & X(e^{j\omega}) = \frac{1}{T} \sum_{k=-\infty}^{\infty} X_c(j(\frac{\omega}{T} - \frac{2\pi k}{T})), \end{aligned}$$

where  $X(e^{j\omega})$  and  $X_c(j\Omega)$  are the DTFT and CTFT of  $x[n]$ ,  $x_c(t)$ . These relationships are illustrated in the figures below. See Section 4.2 of OSB for their derivations and more detailed descriptions of C/D converters.



Time/Frequency domain representation of C/D conversion

In the frequency domain, a C/D converter can be thought of in three stages: periodic repetition of the spectrum at  $2\pi$ , normalization of frequency by  $\frac{1}{T}$ , and scaling of the amplitude by  $\frac{1}{T}$ . The CT frequency  $\Omega$  is related to the DT frequency  $\omega$  by  $\Omega = \frac{\omega}{T}$ . If  $x_c(t)$  is not bandlimited to  $-\frac{\Omega_s}{2} < \Omega_{max} < \frac{\Omega_s}{2}$ , repetition at  $2\pi$  will cause aliasing. Note that the C/D converter here is an idealized sampler, representing the mathematical operations of sampling with a periodic impulse train followed by conversion from impulse train to a discrete-time sequence. It does *not* represent physical sampling circuits. In practice, the ideal C/D converter is approximated by analog-to-digital (A/D) converters, which will also quantize the input signal to a finite number of amplitude levels. OSB figure 4.2 illustrates C/D conversion of the same signal at two different sampling rates.

## Ideal D/C Converter

An ideal D/C converter is depicted in OSB Figure 4.10. As discussed in Section 4.3 of OSB, the input/output relationship of this system is

$$\begin{aligned} \text{Time Domain:} \quad x_r(t) &= \sum_n x[n] \text{sinc}\left(\frac{t-nT}{T}\right) \\ \text{Frequency Domain:} \quad X_r(j\Omega) &= \begin{cases} TX(e^{j\omega}) & \Omega \leq \frac{\pi}{T} \\ 0 & \text{otherwise} \end{cases} \end{aligned}$$

A D/C reconstructs the continuous signal by bandlimited interpolation. In the frequency domain, this is equivalent to ideal low-pass filtering of the scaled spectrum. That is

$$X_r(j\Omega) = H_r(j\Omega)X(e^{j\Omega T}),$$

where

$$H_r(j\Omega) = \begin{cases} T & |\Omega| \leq \frac{\pi}{T} \\ 0 & \text{otherwise} \end{cases}.$$

See OSB Figure 4.8 for the block diagram of such an ideal bandlimited filtering system, and OSB Figure 4.9 for the time-domain representation of bandlimited interpolation. Note that  $x[n] = x_r(nT)$  is *not a definition* of a D/C converter, because it does not specify the values of  $x_r(t)$  for time instances  $t \neq nT$ .

In the frequency domain, a D/C can be interpreted as carrying out three operations in reverse of C/D conversion: bandlimiting to the base period  $-\frac{\Omega_r}{2} \leq \omega \leq \frac{\Omega_r}{2} = \frac{\pi}{T}$ , normalization of frequency by  $T$ , and scaling of the amplitude by  $T$ . Similar to C/D converters, a D/C converter is also ideal, representing the mathematical operations of conversion from sequence to an impulse train, followed by idealized reconstruction.

## DT Processing of CT Signals

OSB Figure 4.15 shows the cascade of a C/D converter, a discrete-time system, followed by a D/C converter. In this system, if  $x_c(t)$  is bandlimited, i.e.  $X_c(j\Omega) = 0$  for  $|\Omega| > \frac{\pi}{T}$ , the overall

system is linear and time invariant, and equivalent to a continuous time LTI system:

$$H_{\text{eff}}(j\Omega) = H(j\Omega) \quad \text{for } |\Omega| \leq \frac{\pi}{T} .$$

For example, Figure 4.13 shows the Fourier transform of signals through the overall system when the input is bandlimited. The discrete-time system in this case is an ideal lowpass filter with cutoff frequency  $\omega_c$ .

**Example:**

$$\begin{aligned} H_c(j\omega) &: \text{Lowpass filter, cutoff frequency} = \pi/4 \\ 1/T &: 20\text{kHz} \\ H_{\text{eff}}(j\Omega) &: \text{Lowpass filter, cutoff frequency} = ? \end{aligned}$$

## CT Processing of DT Signals: Fractional Delay

In addition to using discrete-time systems for processing continuous-time signals, the complementary situation is also possible, where a continuous time system is preceded by a D/C converter and followed by a C/D converter to process discrete-time signals, as depicted in OSB Figure 4.16. Section 4.5 of OSB analyzes such a system in detail. Given a CT system  $H_c(j\Omega)$ , the overall DT system behaves as:

$$H(e^{j\omega}) = H_c(j\frac{\omega}{T}), \quad |\omega| < \pi .$$

Equivalently, the overall system will be the same as a given  $H(e^{j\omega})$ , if the continuous-time system satisfies

$$H_c(j\Omega) = H(e^{j\Omega T}), \quad |\Omega| < \pi/T .$$

An important example of such a system is discussed in OSB Example 4.9: consider a discrete-time system with the following frequency response:

$$H(e^{j\omega}) = e^{-j\omega\Delta} .$$

When  $\Delta$  is an integer, this is a time delay by  $\Delta$  units:  $y[n] = x[n - \Delta]$ . When  $\Delta$  is not an integer, however, such an interpretation is incorrect, since a discrete sequence can only be shifted by integer amounts. Instead, consider choosing  $H_c(j\Omega)$  in OSB Figure 4.16 to be

$$H_c(j\Omega) = H(e^{j\Omega T}) = e^{-j\Omega\Delta T} .$$

It then follows that

$$y_c(t) = x_c(t - \Delta T) .$$

With some computations, we see that the output of this system can be interpreted as bandlimited interpolation of the input discrete-time signal, followed by non-integer time delay and re-sampling. Summarizing using a direct convolution representation with  $-\infty < n < \infty$ :

$$y[n] = \sum_{k=-\infty}^{\infty} x[k] \frac{\sin \pi(n - k - \Delta)}{\pi(n - k - \Delta)} = x[n] * \frac{\sin \pi(n - \Delta)}{\pi(n - \Delta)} = x[n] * h[n] .$$