

Multirate Digital Signal Processing: Part I

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Discrete-Time Signals and Systems

Reference:

Sections 11.1-11.3 of

John G. Proakis and Dimitris G. Manolakis, *Digital Signal Processing: Principles, Algorithms, and Applications*, 4th edition, 2007.

Multirate DSP

- ▶ **sampling rate conversion**: process of converting a given discrete-time signal at a given rate to a different rate
- ▶ **multirate digital signal processing systems**: systems that employ multiple sampling rates

Sampling vs. Sampling Rate Conversion

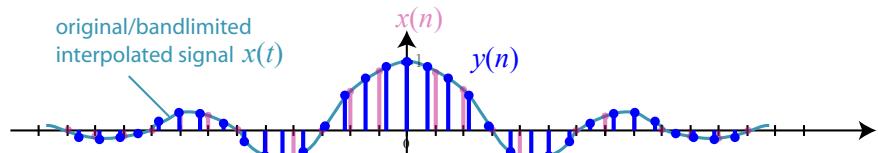
Sampling:

- ▶ conversion from cts-time to dst-time by taking “samples” at discrete time instants
- ▶ E.g., uniform sampling: $x(n) = x_a(nT)$ where T is the sampling period

Sampling rate conversion approaches:

- ▶ convert original samples to **analog domain** and then resample to generate new samples
- ▶ filter original samples with a discrete-time **linear time-varying system** to generate new samples

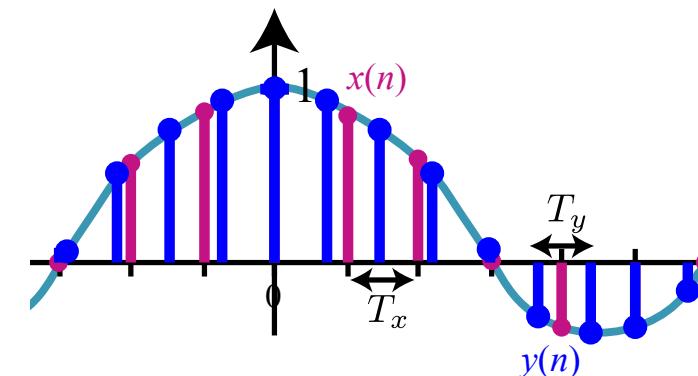
Sampling Rate Conversion



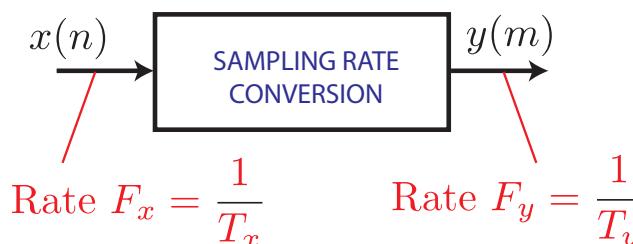
- ▶ $x(n)$: original samples at sampling rate F_x
- ▶ $y(n)$: new samples at sampling rate F_y

Ideal Sampling Rate Conversion

- ▶ $x(n)$: original samples at sampling rate $F_x = \frac{1}{T_x}$
- ▶ $y(n)$: new samples at sampling rate $F_y = \frac{1}{T_y}$



Parameter Relationships



Parameter/Variable	$x(n) \equiv x(nT_x)$	$y(m) \equiv y(mT_y)$
Rate	F_x	F_y
Period	T_x	T_y
Dst-time Frequency	ω_x	ω_y
Cts-time Frequency	F	F

Bridging the Parameter Relationships

- ▶ related to the ratio: $\boxed{\frac{T_y}{T_x}}$

$$F_y = \frac{T_x}{T_y} \cdot F_x$$

$$\omega_x = 2\pi f_x = \frac{2\pi F}{F_y}$$

$$\omega_y = 2\pi f_y = \frac{2\pi F}{F_y}$$

$$\omega_x = \frac{F_y}{F_x} \cdot \omega_y = \frac{T_x}{T_y} \cdot \omega_y$$

Implementation of Sampling Rate Conversion

We relate the original samples $x(nT_x)$ to the new samples $y(mT_y)$ by assuming we convert the signal to analog and resample. Using the interpolation formula

$$y(t) = \sum_{n=-\infty}^{\infty} x(nT_x)g(t - nT_x)$$

where

$$g(t) = \frac{\sin(\pi t/T_x)}{\pi t/T_x} \leftrightarrow G(F) = \begin{cases} T_x & |F| \leq F_x < 2 \\ 0 & \text{otherwise} \end{cases}$$

Note: $y(t) = x(t)$ if $x(t)$ is sampled above Nyquist.

Implementation of Sampling Rate Conversion

$$\begin{aligned} y(t) &= \sum_{n=-\infty}^{\infty} x(nT_x)g(t - nT_x) \\ y(mT_y) &= \sum_{n=-\infty}^{\infty} x(nT_x)g(mT_y - nT_x) \\ &= \sum_{n=-\infty}^{\infty} x(nT_x)g\left(T_x\left(\frac{mT_y}{T_x} - n\right)\right) \\ \frac{mT_y}{T_x} &= k_m + \Delta_m \end{aligned}$$

Implementation of Sampling Rate Conversion

$$\begin{aligned} y(t) &= \sum_{n=-\infty}^{\infty} x(nT_x)g(t - nT_x) \\ \underbrace{y(mT_y)}_{\text{desired samples}} &= \sum_{n=-\infty}^{\infty} \underbrace{x(nT_x)}_{\text{original samples}} \underbrace{g(mT_y - nT_x)}_{\text{samples of } g(t)} \end{aligned}$$

Implementation of Sampling Rate Conversion

$$\begin{aligned} \frac{mT_y}{T_x} &= \underbrace{k_m}_{\text{integer}} + \underbrace{\Delta_m}_{\text{remainder}} \\ k_m &= \left\lfloor \frac{mT_y}{T_x} \right\rfloor \\ \Delta_m &= \frac{mT_y}{T_x} - \left\lfloor \frac{mT_y}{T_x} \right\rfloor \in [0, 1) \end{aligned}$$

Implementation of Sampling Rate Conversion

$$\begin{aligned}
 y(mT_y) &= \sum_{n=-\infty}^{\infty} x(nT_x)g\left(T_x\left(\frac{mT_y}{T_x} - n\right)\right) \\
 &= \sum_{n=-\infty}^{\infty} x(nT_x)g(T_x(k_m + \Delta_m - n)) \\
 &\quad \text{let } k = k_m - n \\
 &= \sum_{k=-\infty}^{\infty} x((k_m - k)T_x)g(T_x(k + \Delta_m)) \\
 &= \underbrace{\sum_{k=-\infty}^{\infty} g((k + \Delta_m)T_x)x((k_m - k)T_x)}_{\text{weighted linear combination of orig samples}}
 \end{aligned}$$

Implementation of Sampling Rate Conversion

$$\begin{aligned}
 y(mT_y) &= \sum_{k=-\infty}^{\infty} g((k + \Delta_m)T_x)x((k_m - k)T_x) \\
 &= \sum_{k=-\infty}^{\infty} g((k + \Delta_m)T_x)x((k_m - k)T_x)
 \end{aligned}$$

- ▶ $g_m(nT_x)$ may have to be retrieved or computed
- ▶ in general, there are as many weights/coefficients required as input samples \times output values to compute
- ▶ in general, no simplification is possible making computation of $y(mT_y)$ from $x(nT_x)$ impractical

Implementation of Sampling Rate Conversion

$$y(mT_y) = \sum_{k=-\infty}^{\infty} g((k + \Delta_m)T_x)x((k_m - k)T_x)$$

- ▶ Δ_m : determines the set of weights
- ▶ k_m : specifies the set of input samples
- ▶ represents a discrete-time linear **time-varying** system
 - ▶ every output sample m requires use of a different impulse response/coefficient set:

$$\begin{aligned}
 g_m(nT_x) &= g((n + \Delta_m)T_x) \\
 \Delta_m &= \frac{mT_y}{T_x} - \left\lfloor \frac{mT_y}{T_x} \right\rfloor \in [0, 1)
 \end{aligned}$$

Linear Periodically Time-Varying Implementation

- ▶ significant simplification possible for $\frac{T_y}{T_x} = \frac{F_x}{F_y} = \frac{D}{I}$ where $D, I \in \mathbb{Z}^+$ and $\text{GCD}(D, I) = 1$

$$\begin{aligned}
 \Delta_m &= \frac{mT_y}{T_x} - \left\lfloor \frac{mT_y}{T_x} \right\rfloor = \frac{mD}{I} - \left\lfloor \frac{mD}{I} \right\rfloor \\
 &= \frac{1}{I} \left(mD - \left\lfloor \frac{mD}{I} \right\rfloor I \right) = \frac{1}{I}(mD) \bmod I \\
 &= \frac{1}{I}(mD)_I
 \end{aligned}$$

Linear Periodically Time-Varying Implementation

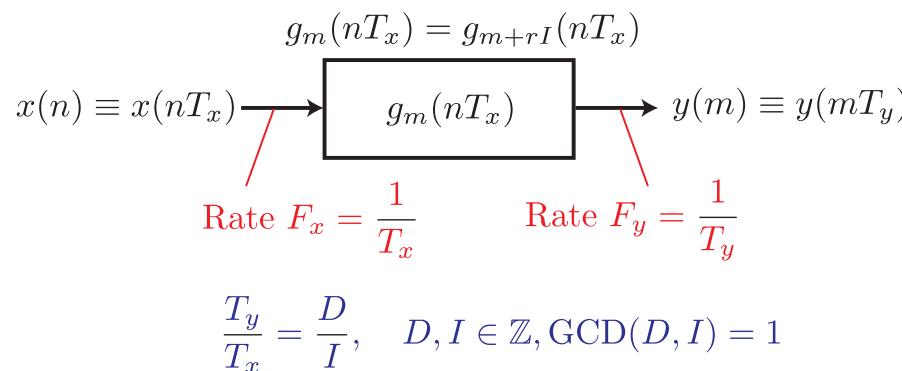
Note:

$$(mD)_I \in \{0, 1, 2, \dots, I-1\}$$

$$\Delta_m = \frac{1}{I}(mD)_I \in \{0, 1/I, 2/I, \dots, (I-1)/I\}$$

- $g_m(nT_x) = g((n + \Delta_m)T_x)$ consists **only of I distinct sets of coefficients!**

Linear Periodically Time-Varying Implementation



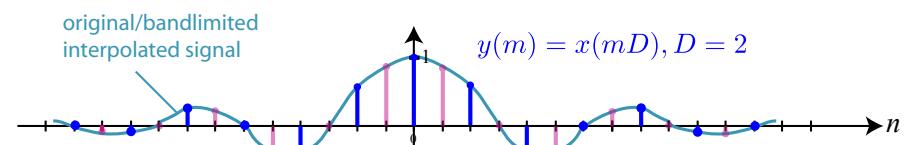
Linear Periodically Time-Varying Implementation

Furthermore, for $r \in \mathbb{Z}$:

$$\begin{aligned} \Delta_{m+rl} &= \frac{1}{I}((m+rl)D)_I = \frac{1}{I}(mD + rID)_I \\ &= \frac{1}{I}(mD)_I = \Delta_m \\ \therefore g_{m+rl}(nT_x) &= g_m(nT_x), \quad r \in \mathbb{Z} \end{aligned}$$

- $g_m(nT_x) = g_{m+rl}(nT_x)$ represents a discrete-time **periodically time-varying system!**

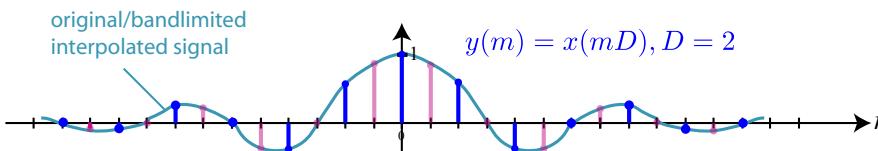
Decimation/Downsampling



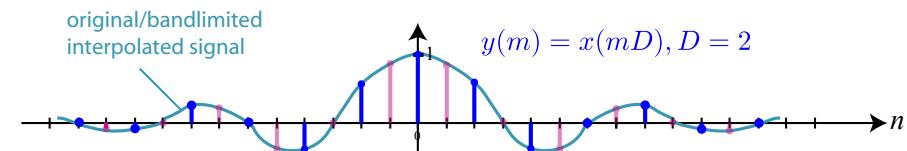
$$T_y = DT_x \implies \frac{T_y}{T_x} = D, \quad D \in \mathbb{Z}^+$$

$$k_m = \left\lfloor \frac{mT_y}{T_x} \right\rfloor = \lfloor mD \rfloor = mD \quad \because mD \in \mathbb{Z}$$

$$\Delta_m = \frac{mT_y}{T_x} - \left\lfloor \frac{mT_y}{T_x} \right\rfloor = mD - \lfloor mD \rfloor = mD - mD = 0$$



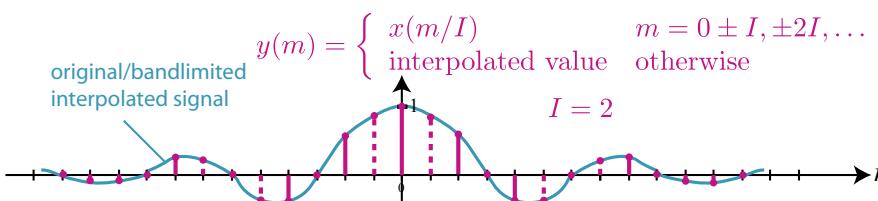
$$\begin{aligned} y(mT_y) &= \sum_{n=-\infty}^{\infty} g((n + \Delta_m)T_x)x((k_m - n)T_x) \\ &= \sum_{n=-\infty}^{\infty} g((n + 0)T_x)x((mD - n)T_x) \\ &= \underbrace{\sum_{n=-\infty}^{\infty} g(nT_x)x((mD - n)T_x)}_{\text{dst-time convolution}} \end{aligned}$$



$$\begin{aligned} y(mT_y) &= \sum_{n=-\infty}^{\infty} g(nT_x)x((mD - n)T_x) \\ &= \sum_{n=-\infty}^{\infty} \frac{\sin(\pi n)}{\pi n} x((mD - n)T_x) \\ &= \sum_{n=-\infty}^{\infty} \delta(n)x((mD - n)T_x) = x(mDT_x) \end{aligned}$$

See [Figure 11.1.3 of text](#).

Interpolation/Upsampling



$$T_x = IT_y \implies \frac{T_y}{T_x} = \frac{1}{I}, \quad I \in \mathbb{Z}^+$$

$$k_m = \left\lfloor \frac{mT_y}{T_x} \right\rfloor = \left\lfloor \frac{m}{I} \right\rfloor$$

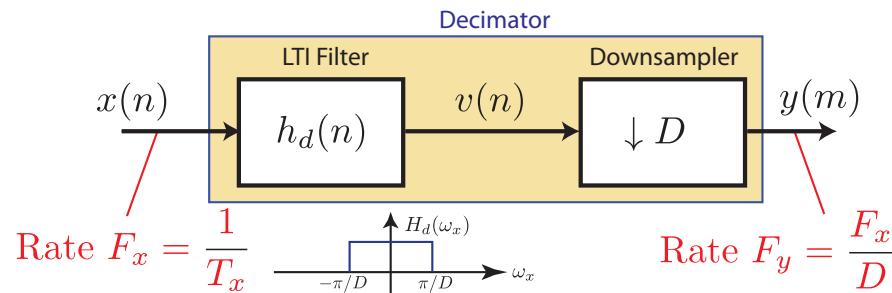
$$\begin{aligned} \Delta_m &= \frac{mT_y}{T_x} - \left\lfloor \frac{mT_y}{T_x} \right\rfloor = \frac{m}{I} - \left\lfloor \frac{m}{I} \right\rfloor \\ &\in \{0, 1/I, 2/I, \dots, (I-1)/I\} \end{aligned}$$

Interpolation/Upsampling

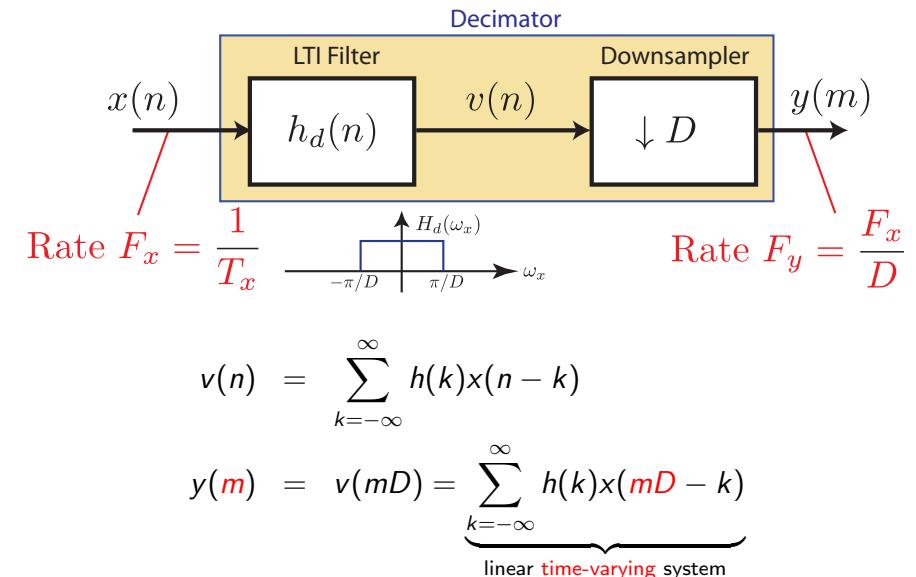
$$\begin{aligned} y(mT_y) &= \sum_{n=-\infty}^{\infty} x(nT_x)g(mT_y - nT_x) \\ &= \sum_{n=-\infty}^{\infty} x(nT_x)g\left(\frac{mT_x}{I} - nT_x\right) \\ &= \sum_{n=-\infty}^{\infty} x(nT_x) \frac{\sin(\pi \frac{m}{I} - \pi n)}{\pi \frac{m}{I} - \pi n} \\ &= \sum_{n=-\infty}^{\infty} x(nT_x) \underbrace{\frac{\sin(\frac{\pi}{I}(m - nl))}{\frac{\pi}{I}(m - nl)}}_{\text{sinc centered at } n = m/I} \end{aligned}$$

See [Figure 11.1.4 of text](#).

Downsampling with Anti-Aliasing Filter



- Downsampling alone may cause aliasing, therefore, it is desirable to introduce an **anti-aliasing** filter $H_d(\omega_x)$



Downsampling: Frequency Domain Perspective

Goal: determine relationship between input-output spectra

Create an intermediate signal $\tilde{v}(n)$ at rate F_x but with the equivalent information as $y(m)$.

$$\begin{aligned}\tilde{v}(n) &= \begin{cases} v(n) & n = 0, \pm D, \pm 2D, \dots \\ 0 & \text{otherwise} \end{cases} \\ &= v(n) \cdot p(n) \quad \text{where } p(n) = \sum_{k=-\infty}^{\infty} \delta(n - kD)\end{aligned}$$

$$\begin{aligned}\text{Note: } y(m) &= v(mD) \cdot 1 = v(mD) \cdot p(mD) = \tilde{v}(mD) \\ y(m) &\longleftrightarrow \tilde{v}(n) \text{ (equiv info)}\end{aligned}$$

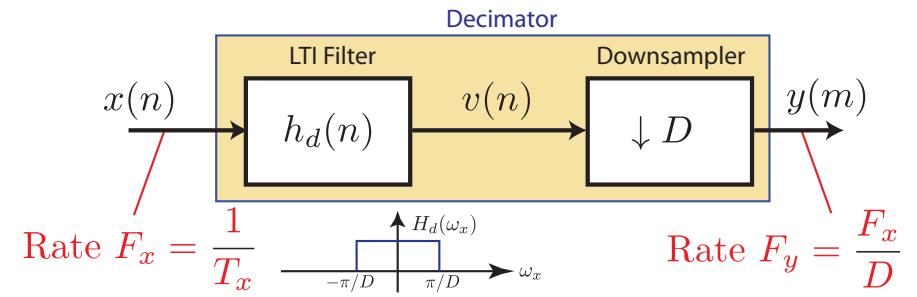
See [Figure 11.2.2 of text](#).

Aside: Impulse Train $p(n)$

$$\begin{aligned}p(n) &= \sum_{l=-\infty}^{\infty} \delta(n - lD) \quad (\text{periodic with period } D) \\ c_k &= \frac{1}{D} \sum_{n=0}^{D-1} p(n) e^{j2\pi kn/D} \\ &= \sum_{n=0}^{D-1} \underbrace{\sum_{l=-\infty}^{\infty} \delta(n - lD)}_{\text{zero for } l \neq 0} e^{j2\pi kn/D} \\ &= \frac{1}{D} \sum_{n=0}^{D-1} \delta(n) e^{j2\pi kn/D} = \frac{1}{D} \\ p(n) &= \sum_{k=0}^{D-1} c_k e^{j2\pi km/D} = \frac{1}{D} \sum_{k=0}^{D-1} e^{j2\pi km/D}\end{aligned}$$

Downsampling: Frequency Domain Perspective

$$\begin{aligned}
 Y(z) &= \sum_{m=-\infty}^{\infty} y(m)z^{-m} = \sum_{m=-\infty}^{\infty} \tilde{v}(mD)z^{-m} \\
 &= \cdots + \tilde{v}(-D)z^1 + \tilde{v}(0)z^0 + \tilde{v}(D)z^{-1} + \cdots \\
 &= \sum_{m'=-\infty}^{\infty} \tilde{v}(m')z^{-m'/D} \text{ since } \tilde{v}(m) = 0 \text{ for } m \notin \{0, \pm D, \pm 2D, \dots\} \\
 &= \sum_{m'=-\infty}^{\infty} v(m)p(m)z^{-m'/D} = \sum_{m=-\infty}^{\infty} v(m) \frac{1}{D} \sum_{k=0}^{D-1} e^{j2\pi km/D} z^{-m'/D} \\
 &= \frac{1}{D} \sum_{k=0}^{D-1} \underbrace{\sum_{m=-\infty}^{\infty} v(m)(e^{-j2\pi k/D} z^{1/D})^{-m}}_{\equiv V(e^{-j2\pi k/D} z^{1/D}) = H_d(e^{-j2\pi k/D} z^{1/D}) X(e^{-j2\pi k/D})} \\
 &= \frac{1}{D} \sum_{k=0}^{D-1} H_d(e^{-j2\pi k/D} z^{1/D}) X(e^{-j2\pi k/D} z^{1/D})
 \end{aligned}$$

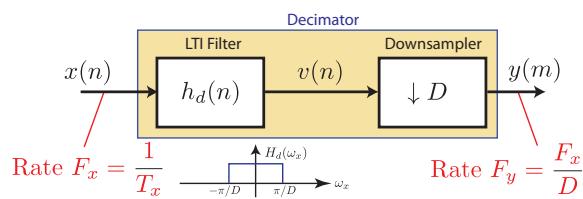


In the preceding analysis, we employed:

$$\begin{aligned}
 v(n) &\xleftrightarrow{z} V(z) \\
 h_d(n) &\xleftrightarrow{z} H_d(z) \\
 x(n) &\xleftrightarrow{z} X(z) \\
 V(z) &= \sum_{m=-\infty}^{\infty} v(m)z^{-m} \\
 V(z) &= H_d(z) \cdot X(z)
 \end{aligned}$$

Let $z = e^{j\omega_y}$:

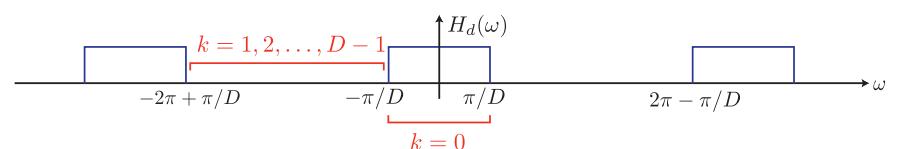
$$\begin{aligned}
 Y(z) &= \frac{1}{D} \sum_{k=0}^{D-1} H_d(e^{-j2\pi k/D} z^{1/D}) X(e^{-j2\pi k/D} z^{1/D}) \\
 Y(e^{j\omega_y}) &= \frac{1}{D} \sum_{k=0}^{D-1} H_d(e^{-j2\pi k/D} e^{j\omega_y 1/D}) X(e^{-j2\pi k/D} e^{j\omega_y 1/D}) \\
 &= \frac{1}{D} \sum_{k=0}^{D-1} H_d(e^{j(\omega_y - 2\pi k)/D}) X(e^{j(\omega_y - 2\pi k)/D}) \\
 Y(\omega_y) &= \frac{1}{D} \sum_{k=0}^{D-1} H_d\left(\frac{\omega_y - 2\pi k}{D}\right) X\left(\frac{\omega_y - 2\pi k}{D}\right)
 \end{aligned}$$



$$Y(\omega_y) = \frac{1}{D} \sum_{k=0}^{D-1} H_d\left(\frac{\omega_y - 2\pi k}{D}\right) X\left(\frac{\omega_y - 2\pi k}{D}\right)$$

For $-\pi \leq \omega_y \leq \pi$,

$$\begin{aligned}
 -\frac{\pi}{D} &\leq \frac{\omega_y - 2\pi k}{D} \leq \frac{\pi}{D} && \text{for } k = 0 \\
 -\frac{3\pi}{D} &\leq \frac{\omega_y - 2\pi k}{D} \leq -\frac{\pi}{D} && \text{for } k = 1 \\
 &\vdots && \vdots \\
 -2\pi + \frac{\pi}{D} &\leq \frac{\omega_y - 2\pi k}{D} \leq -2\pi + \frac{3\pi}{D} && \text{for } k = D-1
 \end{aligned}$$



Note: For $-\pi \leq \omega_y \leq \pi$,

$$H_d\left(\frac{\omega_y - 2\pi k}{D}\right) = \begin{cases} 1 & \text{for } k = 0 \\ 0 & \text{for } k = 1, 2, \dots, D-1 \end{cases}$$

Therefore, for $-\pi \leq \omega_y \leq \pi$,

$$\begin{aligned} Y(\omega_y) &= \frac{1}{D} \sum_{k=0}^{D-1} H_d \left(\frac{\omega_y - 2\pi k}{D} \right) X \left(\frac{\omega_y - 2\pi k}{D} \right) \\ &= \underbrace{\frac{1}{D} H_d \left(\frac{\omega_y}{D} \right) X \left(\frac{\omega_y}{D} \right)}_{=1} + \underbrace{\frac{1}{D} \sum_{k=1}^{D-1} H_d \left(\frac{\omega_y - 2\pi k}{D} \right) X \left(\frac{\omega_y - 2\pi k}{D} \right)}_{=0} \end{aligned}$$

$$Y(\omega_y) = \frac{1}{D} X \left(\frac{\omega_y}{D} \right)$$

Note: $\omega_y = \frac{T_y}{T_x} \omega_x = D \omega_x$.

$-\frac{\pi}{D} \leq \omega_x \leq \frac{\pi}{D}$ of $X(\omega_x)$ is stretched into $-\pi \leq \omega_y \leq \pi$ for $Y(\omega_y)$.

See [Figure 11.2.3 of text](#).

Goal: determine relationship between input-output spectra

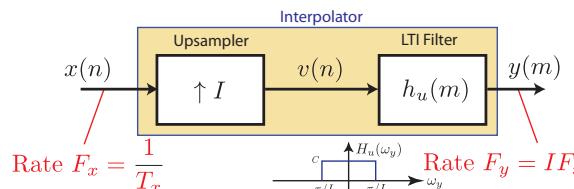
Consider an intermediate signal $v(n)$ at rate F_y but with the equivalent information as $x(m)$.

$$v(m) = \begin{cases} x(m/I) & m = 0, \pm 1, \pm 2I, \dots \\ 0 & \text{otherwise} \end{cases}$$

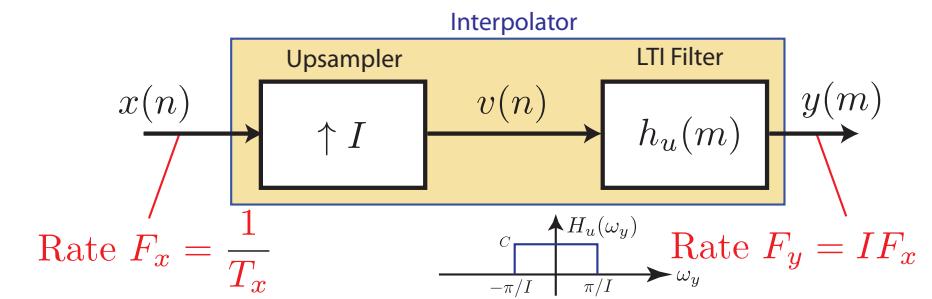
$$V(z) = \sum_{m=-\infty}^{\infty} v(m)z^{-m} = \dots + v(-I)z^I + v(0)z^0 + v(I)z^{-I} + \dots$$

$$= \sum_{m=-\infty}^{\infty} x(m)z^{-ml} = \sum_{m=-\infty}^{\infty} x(m)(z^I)^{-m} = X(z^I)$$

$$V(e^{j\omega_y}) = X(e^{j\omega_y I}) \implies V(\omega_y) = X(\omega_y I)$$



Interpolation by a Factor I



- ▶ Interpolation only increases the visible resolution of the signal.
- ▶ No information gain is achieved. At best $H_u(\omega_y)$ maintains the same information in $y(n)$ as exists in $x(n)$.

$$V(\omega_y) = X(\omega_y I)$$

See [Figure 11.3.1 of text](#).

$$H_u(\omega_y) = \begin{cases} I & 0 \leq |\omega_y| \leq \pi/I \\ 0 & \text{otherwise} \end{cases}$$

$$Y(\omega_y) = H_u(\omega_y) V(\omega_y) = \begin{cases} IX(\omega_y I) & 0 \leq |\omega_y| \leq \pi/I \\ 0 & \text{otherwise} \end{cases}$$

$$Y(\omega_y) = \begin{cases} IX(\omega_y I) & 0 \leq |\omega_y| \leq \pi/I \\ 0 & \text{otherwise} \end{cases}$$

Note: $\omega_y = \frac{T_y}{T_x} = \frac{\omega_x}{I}$.

$-\pi \leq \omega_x \leq \pi$ is compressed into $-\pi/I \leq \omega_y \leq \pi/I$

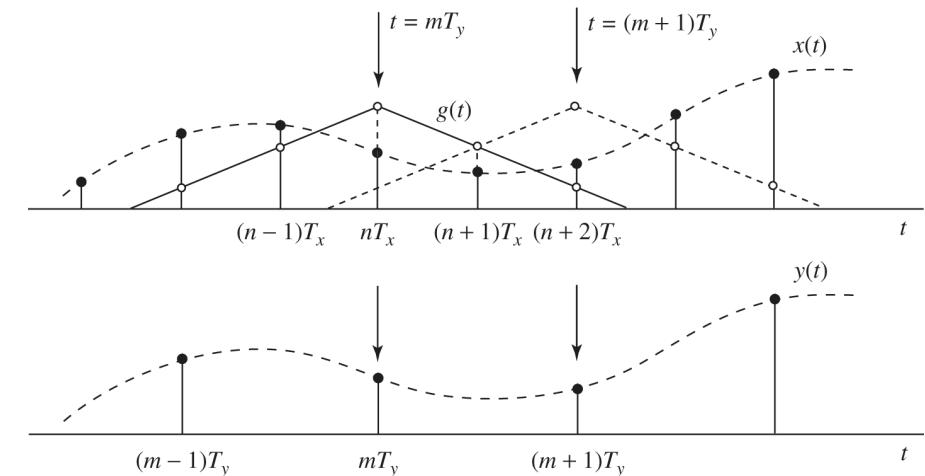
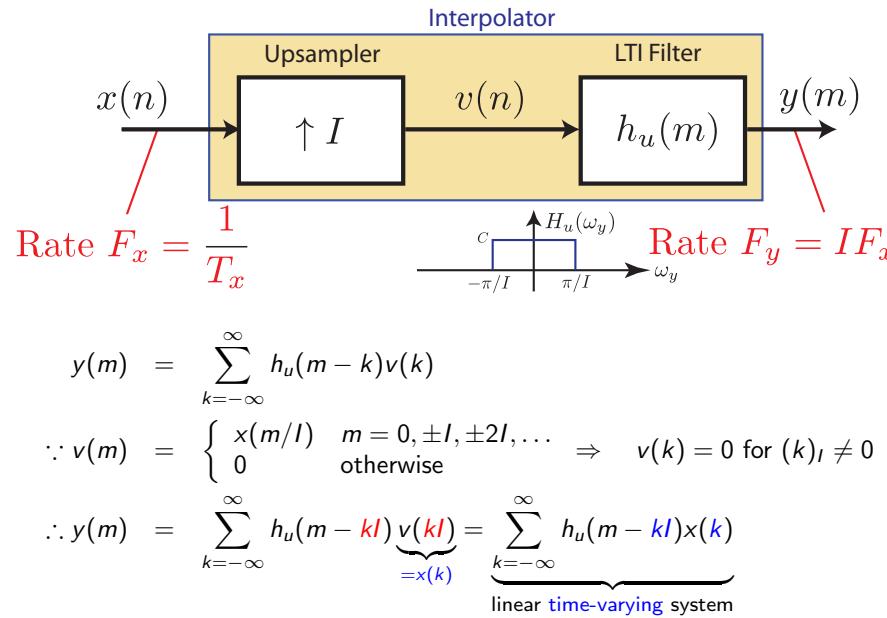


Figure 11.1.3 Illustration of timing relations for sampling rate decrease by an integer factor $D = 2$. A single impulse response, sampled with period T_x , is shifted at steps equal to $T_y = DT_x$ to generate the output samples.

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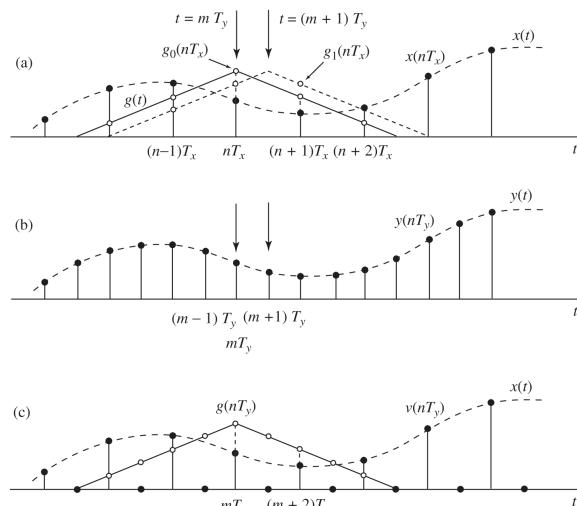


Figure 11.1.4 Illustration of timing relations for sampling rate increase by an integer factor $I = 2$. The approach in (a) requires one impulse response for the even-numbered and one for the odd-numbered output samples. The approach in (c) requires only one impulse response, obtained by interleaving the impulse responses in (a).

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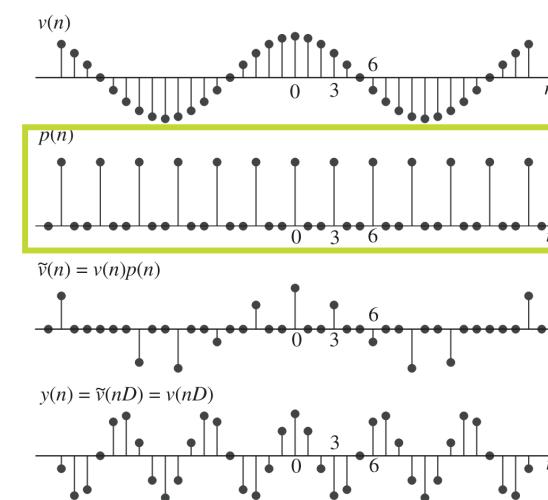
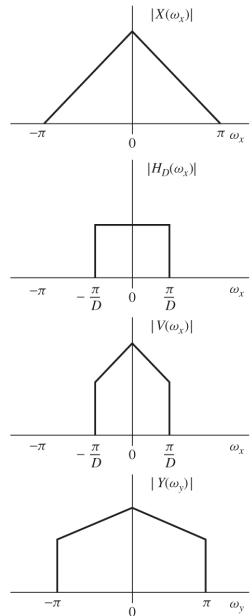


Figure 11.2.2 Steps required to facilitate the mathematical description of downsampling by a factor D , using a sinusoidal sequence for illustration.

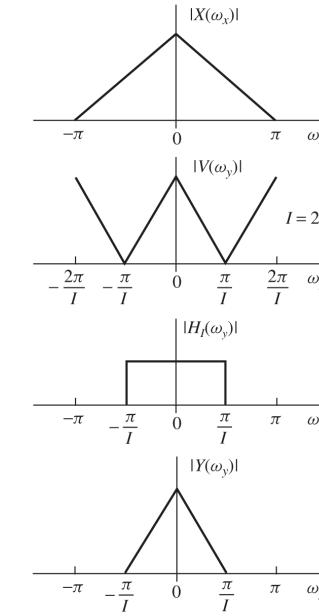
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Figure 11.2.3 Spectra of signals in the decimation of $x(n)$ by a factor D .

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Figure 11.3.1 Spectra of $x(n)$ and $v(n)$ where $V(\omega_y) = X(\omega_l)$.

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