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Shannon's Sampling Theorem

**Author:**

Buğra ÇINAROĞLU - 2577849

Ogan Altuğ OKUTAN - .....

Abdullah Naci BODUR - .....

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# 1 Sampling

We live in an analog world. Whether recording sounds, capturing images, or processing an electromagnetic wave, many sources of information are of analog or continuous-time (CT) nature. We usually want to store, exchange, or manipulate this information using digital computers, microprocessors, and so on. **Sampling:** The process of converting a CT signal to a discrete-time (DT) signal (i.e., a discrete sequence of numbers)

$$x(t) \longrightarrow \dots, x(-2T), x(-T), x(0), x(T), x(2T), \dots$$

**Sampling theorem** states that *under certain conditions* a CT signal can be completely represented by its values (‘samples’) at points equally spaced in time. That is, no information is lost in the sampling process.

**Why is sampling theorem very important/useful?**

- It provides a bridge between continuous-time signals and discrete-time signals.
- Sampling theorem allows us to represent continuous-time signals with discrete samples.
- With the development of digital technology, it became much easier to process discrete-time signals.

## 2 Shannon Sampling Theorem

### 2.1 Definitions

The **Shannon Sampling Theorem** states that a continuous-time signal  $x(t)$  that is *bandlimited* to a maximum angular frequency  $\omega_M$  can be *completely reconstructed* from its samples if the sampling frequency  $\omega_s$  satisfies

$$\omega_s > 2\omega_M \quad \text{or equivalently} \quad f_s > 2f_M$$

where  $f_s$  is the sampling frequency and  $f_M$  is the highest frequency component of  $x(t)$ .

This condition ensures that the replicas of the signal spectrum in the frequency domain do not overlap, thereby preventing *aliasing*. When this criterion is satisfied, the original signal  $x(t)$  can be exactly recovered from its discrete samples  $x(nT)$  using an ideal low-pass filter.

**Definition:** A signal  $x(t)$  is bandlimited if

$$X(j\omega) = 0 \quad \text{for } |\omega| > \omega_m \text{ for some } \omega_m$$

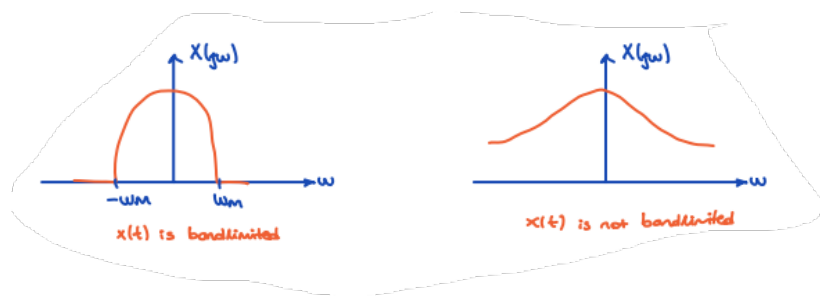


Figure 1: Example for Bandlimited Signal

**Ideal impulse-train sampling:**

Periodic impulse train

$$p(t) = \sum_{n=-\infty}^{\infty} \delta(t - nT)$$

Sampling of a CT signal  $x(t)$ :

$$x_p(t) = x(t)p(t)$$

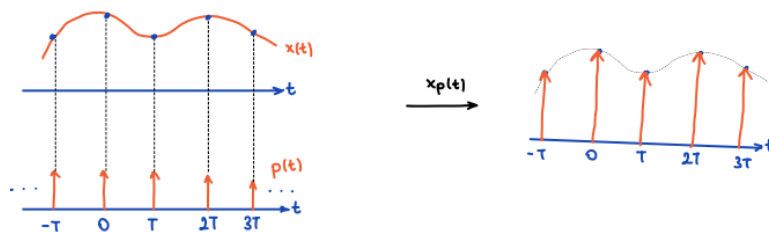


Figure 2: Example for an Impulse Train

$$x_p(t) = \sum_{n=-\infty}^{\infty} x(nT) \delta(t - nT)$$

**Find the spectrum of  $x_p(t)$ :**

$$p(t) = \sum_{n=-\infty}^{\infty} \delta(t - nT) \iff P(j\omega) = \frac{2\pi}{T} \sum_{k=-\infty}^{\infty} \delta(\omega - k\omega_s)$$

where

$$\omega_s = \frac{2\pi}{T} \quad (\text{sampling frequency}), \quad T : \text{sampling period.}$$

$T$ : sampling period

$$\omega_s = \frac{2\pi}{T} \quad : \text{sampling frequency}$$

$$X_p(j\omega) = \frac{1}{2\pi} X(j\omega) * P(j\omega) = \frac{1}{2\pi} X(j\omega) * \frac{2\pi}{T} \sum_{k=-\infty}^{\infty} \delta(\omega - k\omega_s) = \frac{1}{T} \sum_{k=-\infty}^{\infty} X(j(\omega - k\omega_s))$$

where  $X(j(\omega - k\omega_s))$  represents shifted copies of  $X(j\omega)$

When  $\omega_s > 2\omega_M$  (sampling frequency greater than twice the largest frequency in the signal)

$x(j\omega)$  can be recovered from  $x_p(j\omega)$  if  $\omega_s > 2\omega_M$

$x(t)$  can be recovered from  $x_p(t)$

Assume  $x(t)$  is bandlimited with maximal frequency  $\omega_M$ . When  $\omega_s > 2\omega_M$ ,

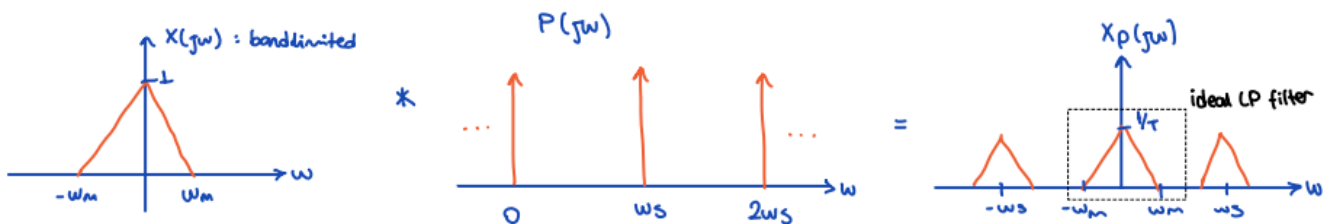


Figure 3: Example of the Method

By using a low-pass (LP) filter, we can get  $X(j\omega)$  from  $X_p(j\omega)$ .

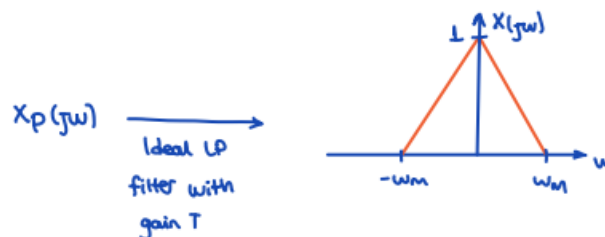


Figure 4: Recovery

So  $X(j\omega)$  can be recovered from its samples if  $\omega_s > 2\omega_M$ .

## 2.2 Effect of Undersampling: Aliasing

The problem of resulting overlap in the frequency domain is referred to as **aliasing**. If  $\omega_s < 2\omega_M$  (undersampling): We lose information  $\Rightarrow$  **Aliasing** occurs if  $\omega_s < 2\omega_M$ .  $X(j\omega)$  cannot be recovered from its samples when  $\omega_s < 2\omega_M$ .

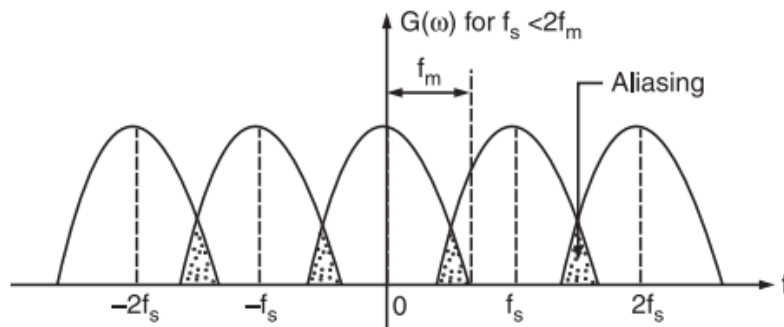


Figure 5: Aliasing

As the sampling frequency decreased the below results are obtained.

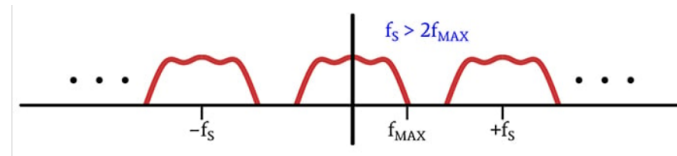


Figure 6: Aliasing Example

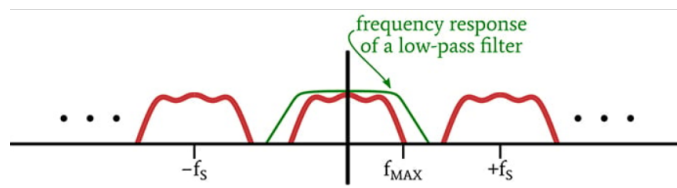


Figure 7: Aliasing Example

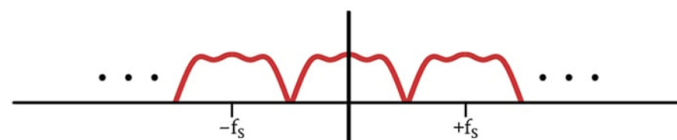


Figure 8: Aliasing Example

### 3 Examples

For the sampling frequency of  $\omega_s = 7$  Hz

$$y(t) = \cos(2\pi \times 15 \text{ Hz} \times t) \text{ where } \omega = 15 \text{ Hz}$$

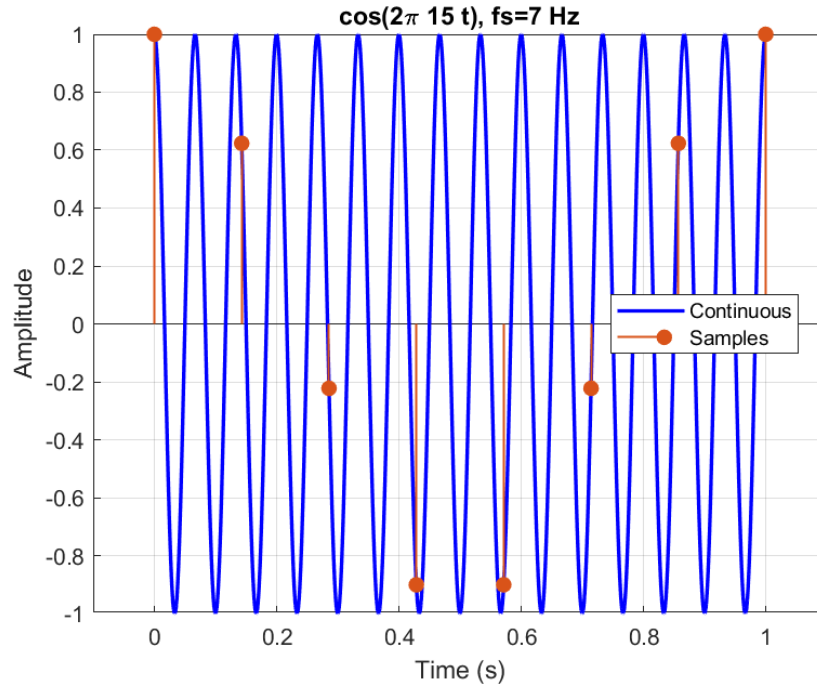


Figure 9: 15 Hz sampled with 7 Hz

If we fit reconstruct the signal from the sampled points.

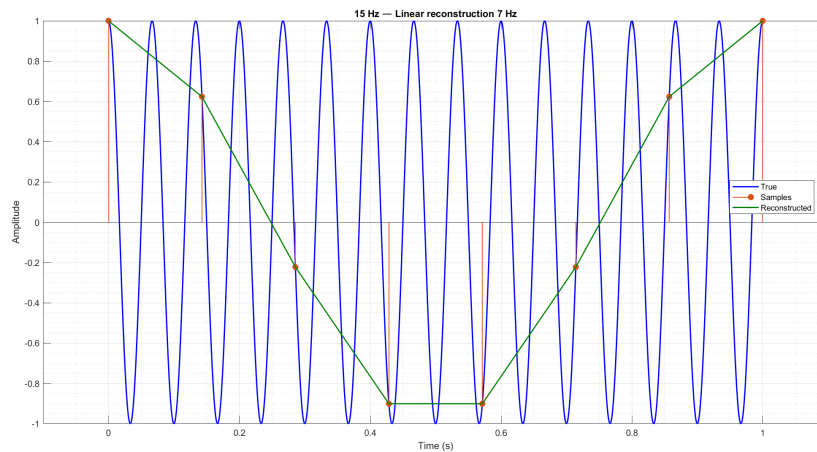


Figure 10: 15 Hz sampled with 7 Hz Reconstruction

$$y(t) = \cos(2\pi \times 13 \text{ Hz} \times t) \text{ where } \omega = 13 \text{ Hz}$$

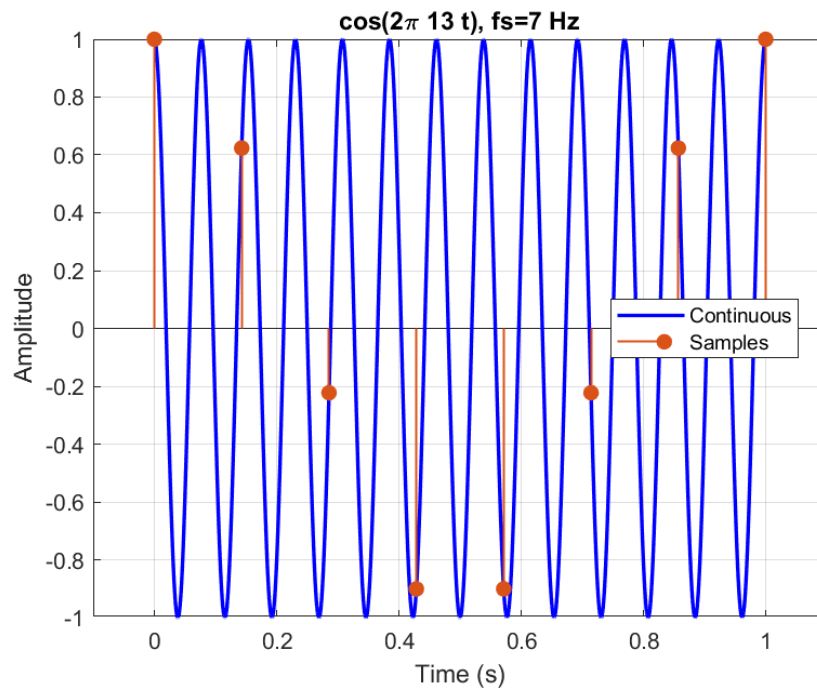


Figure 11: 13 Hz sampled with 7 Hz

If we fit reconstruct the signal from the sampled points.

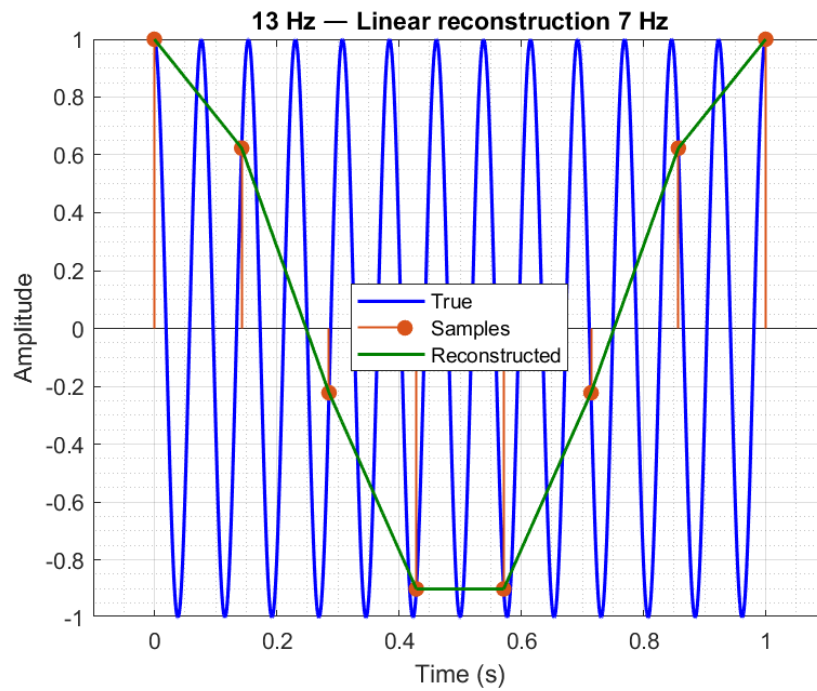


Figure 12: 13 Hz sampled with 7 Hz Reconstruction



$$y(t) = \cos(2\pi \times 8 \text{ Hz} \times t) \text{ where } \omega = 8 \text{ Hz}$$

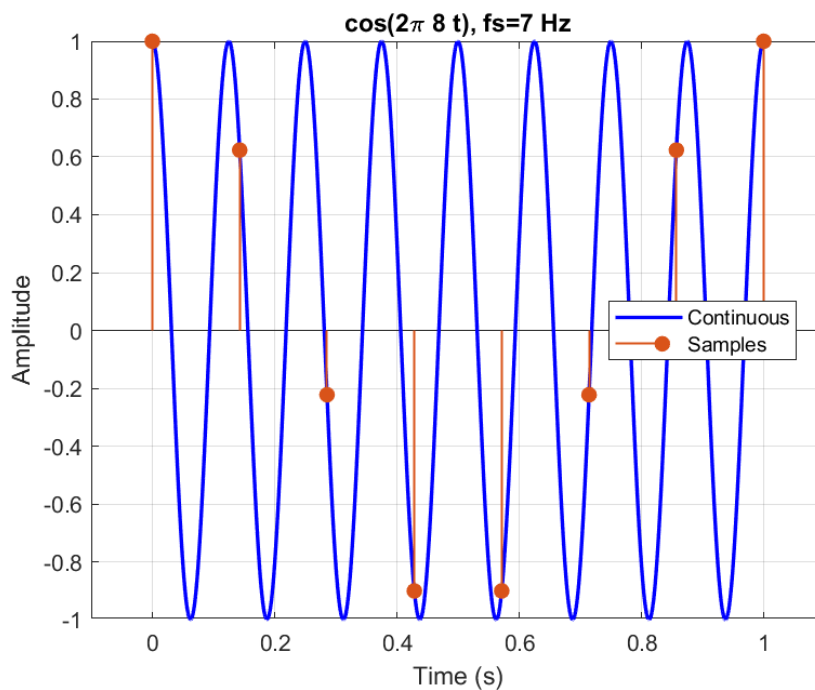


Figure 13: 8 Hz sampled with 7 Hz

If we fit reconstruct the signal from the sampled points.

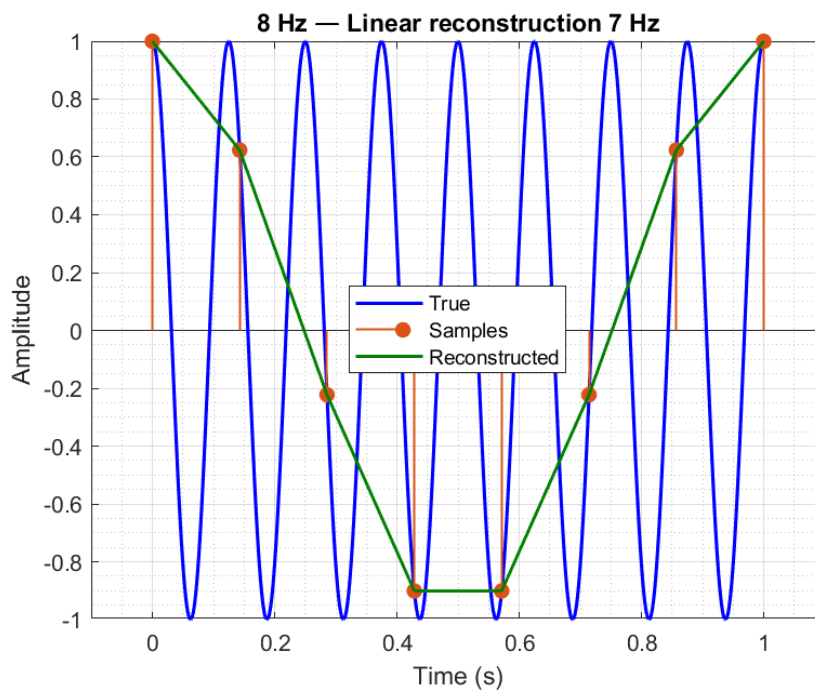


Figure 14: 8 Hz sampled with 7 Hz Reconstruction

$$y(t) = \cos(2\pi \times 6 \text{ Hz} \times t) \text{ where } \omega = 6 \text{ Hz}$$

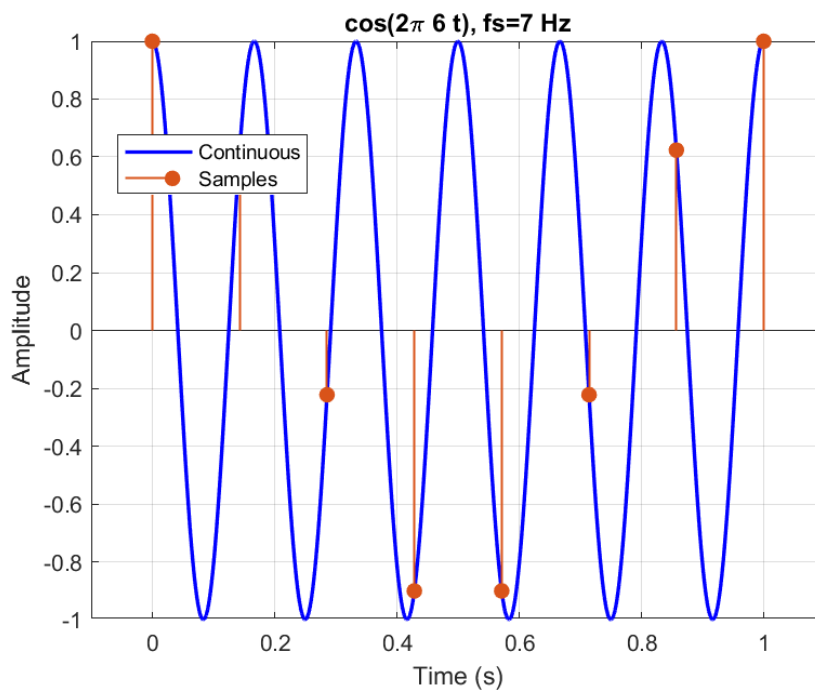


Figure 15: 6 Hz sampled with 7 Hz

If we fit reconstruct the signal from the sampled points.

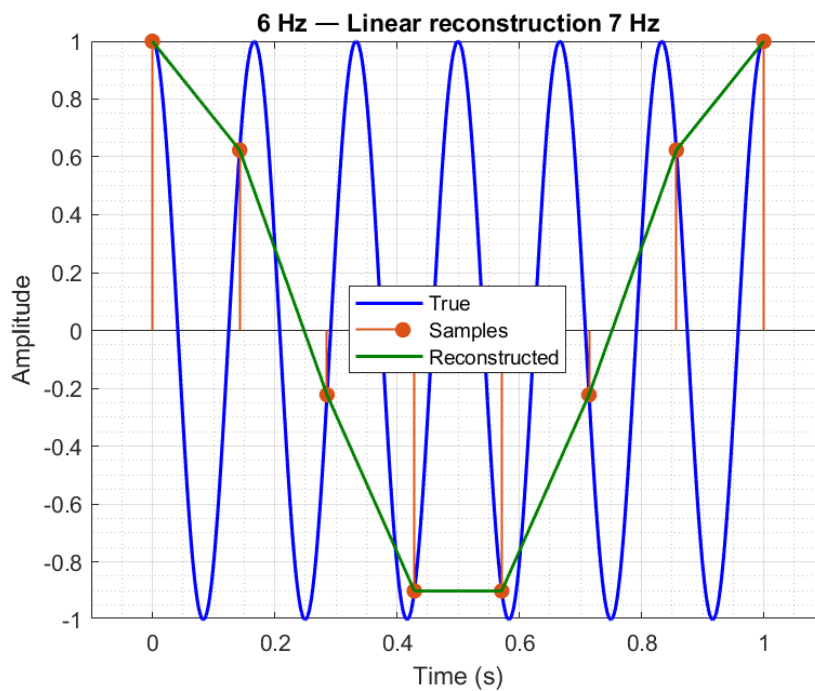


Figure 16: 6 Hz sampled with 7 Hz Reconstruction

$$y(t) = \cos(2\pi \times 3 \text{ Hz} \times t) \text{ where } \omega = 3 \text{ Hz}$$

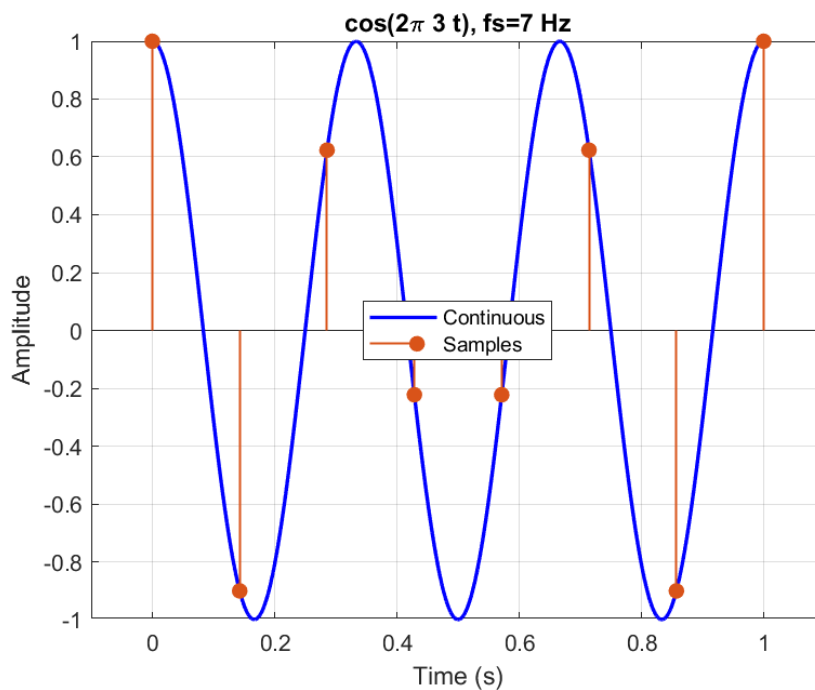


Figure 17: 3 Hz sampled with 7 Hz

If we fit reconstruct the signal from the sampled points.

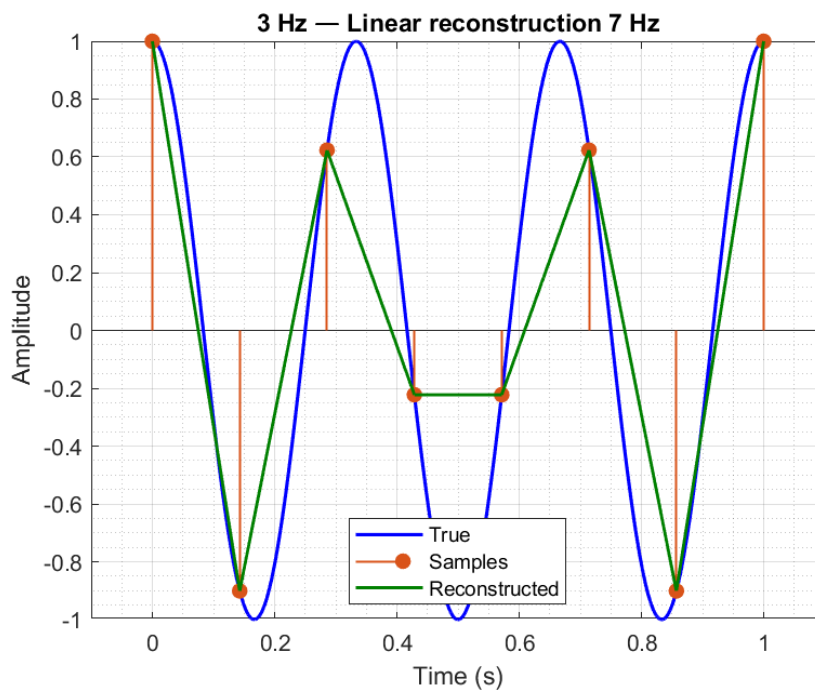


Figure 18: 3 Hz sampled with 7 Hz Reconstruction

$$y(t) = \cos(2\pi \times 1 \text{ Hz} \times t) \text{ where } \omega = 1 \text{ Hz}$$

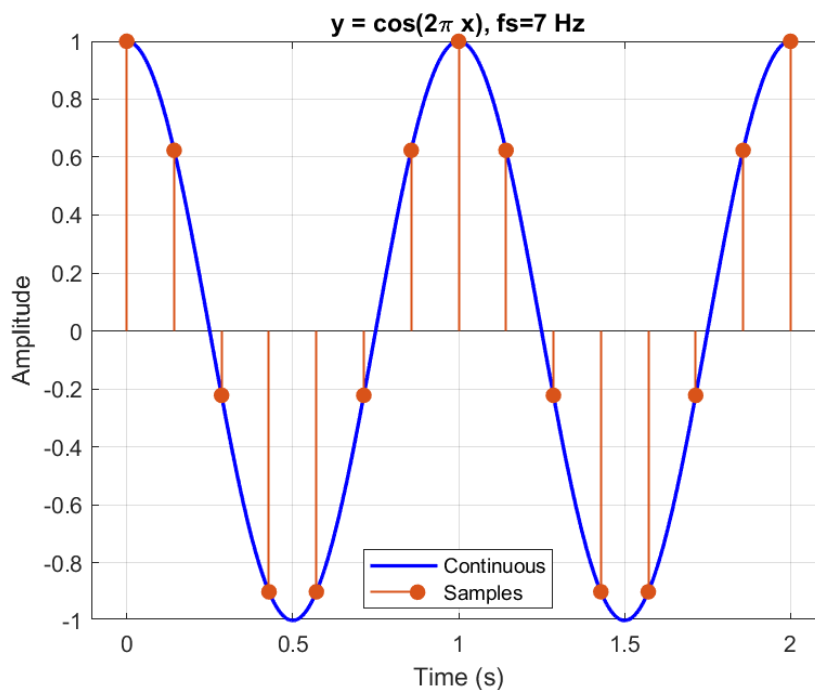


Figure 19: 1 Hz sampled with 7 Hz

If we fit reconstruct the signal from the sampled points.

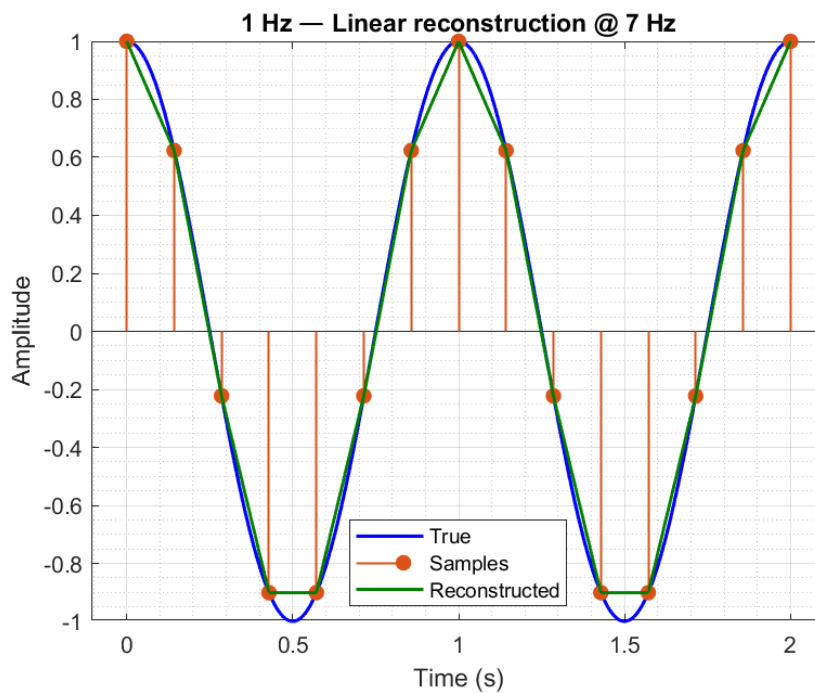


Figure 20: 1 Hz sampled with 7 Hz Reconstruction with Linear Fit

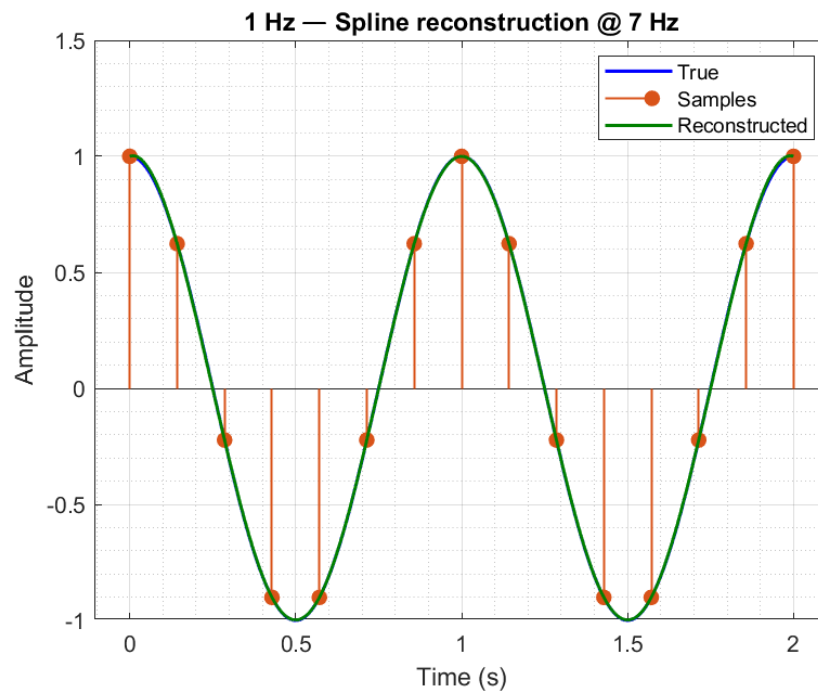


Figure 21: 1 Hz sampled with 7 Hz Reconstruction with Spline Fit

## 4 Appendix

### 4.1 Codes

```
function [hLine, hStem, t, ts, y, ys] = plot_true_with_samples(fun, fs, tspan, ...
    varargin)
    % ----- Parse inputs -----
    validateattributes(fs, {'numeric'}, {'scalar','real','positive'});
    validateattributes(tspan, {'numeric'}, {'numel',2,'real'});
    t0 = tspan(1); tf = tspan(2);
    if tf ≤ t0, error('tspan must satisfy [t0 tf] with tf > t0.');
```

end

```

    p = inputParser;
    addParameter(p, 'N', 4000, @(x) isnumeric(x) && isscalar(x) && x>0);
    addParameter(p, 'LineWidth', 1.6, @(x) isnumeric(x) && isscalar(x));
    addParameter(p, 'LineColor', [0 0 1], @(x) isnumeric(x) && numel(x)==3);
    addParameter(p, 'StemColor', [0.8500 0.3250 0.0980], @(x) isnumeric(x) && ...
        numel(x)==3);
    addParameter(p, 'Marker', 'filled', @(x) ischar(x) || isstring(x));
    addParameter(p, 'Title', '', @(x) ischar(x) || isstring(x));
    addParameter(p, 'XLabel', 'Time (s)', @(x) ischar(x) || isstring(x));
    addParameter(p, 'YLabel', 'Amplitude', @(x) ischar(x) || isstring(x));
    addParameter(p, 'Legend', true, @(x) islogical(x) && isscalar(x));
    parse(p, varargin{:});
    opt = p.Results;

    % ----- Build time vectors -----
    t = linspace(t0, tf, opt.N); % dense time for continuous curve
    Ts = 1/fs;
    % Include the endpoint tf if it aligns within numerical tolerance
    ts = t0:Ts:tf + 1e-12; % sample times

    % ----- Evaluate function -----
    y = fun(t);
    ys = fun(ts);

    % ----- Plot -----
    hold state = ishold;
    hLine = plot(t, y, 'LineWidth', opt.LineWidth, 'Color', opt.LineColor); ...
        hold on
    hStem = stem(ts, ys, 'filled', 'LineWidth', 1.0, 'Color', opt.StemColor, ...
        'BaseValue', 0); %ok<NASGU> % vertical stems
    grid on
    xlabel(opt.XLabel); ylabel(opt.YLabel);

    if isempty(opt.Title)
        ttl = sprintf('Continuous function with samples @ f_s = %.3g Hz', fs);
    else
        ttl = opt.Title;
    end
    title(ttl);

    if opt.Legend
        legend({'Continuous','Samples'}, 'Location', 'best');
    end
end
```

```

    if ~hold_state, hold off; end
end

fun = @(t) cos(2*pi*1*t);    % 1 Hz cosine
f_sampling = 7;
figure;
plot_true_with_samples(fun, f_sampling, [0 2], 'Title', 'y = cos(2\pi x), fs=7 ...
    Hz');

for f = [3 6 8 13 15]
    f_sampling = 7;
    figure;
    plot_true_with_samples(@(t) cos(2*pi*f*t), f_sampling, [0 1], 'Title', ...
        sprintf('cos(2\pi %d t), fs=%d Hz', f, f_sampling));
end

```

```

clc; clear; close all;
function [hTrue, hStem, hRec, t, ts, yTrue, yRec] = ...
    plot_reconstruction_with_samples(fun, fs, tspan, varargin)
    validateattributes(fs, {'numeric'}, {'scalar', 'real', 'positive'});
    validateattributes(tspan, {'numeric'}, {'numel', 2, 'real'});
    t0 = tspan(1); tf = tspan(2); if tf ≤ t0, error('tspan must satisfy tf > ...
        t0.');
```

```

    end

    p = inputParser;
    addParameter(p, 'N', 4000, @(x) isnumeric(x) && isscalar(x) && x>0);
    addParameter(p, 'Method', 'linear', @(s) ...
        any(strcmpi(s, {'linear', 'spline', 'sinc'})));
    addParameter(p, 'SincHalfLobes', 8, @(x) isnumeric(x) && isscalar(x) && x>0);
    addParameter(p, 'TrueColor', [0 0 1], @(x) isnumeric(x) && numel(x)==3);
    addParameter(p, 'RecColor', [0 0.5 0], @(x) isnumeric(x) && numel(x)==3);
    addParameter(p, 'StemColor', [0.8500 0.3250 0.0980], @(x) isnumeric(x) && ...
        numel(x)==3);
    addParameter(p, 'Title', '', @(x) ischar(x) || isstring(x));
    addParameter(p, 'Legend', true, @(x) islogical(x) && isscalar(x));
    parse(p, varargin{:});
    opt = p.Results;

    % Time and samples
    t = linspace(t0, tf, opt.N);
    Ts = 1/fs;
    ts = t0:Ts:tf+1e-12;

    % True & samples
    yTrue = fun(t);
    ys = fun(ts);

    % Reconstruction
    method = lower(opt.Method);
    switch method
        case 'linear'
            yRec = interp1(ts, ys, t, 'linear');
        case 'spline'

```

```

        yRec = interp1(ts, ys, t, 'spline');
    case 'sinc'
        yRec = local_sinc_reconstruct(t, ts, ys, opt.SincHalfLobes);
    otherwise
        error('Unknown Method.');
```

end

```

% Plot
holdState = ishold;
hTrue = plot(t, yTrue, 'Color', opt.TrueColor, 'LineWidth', 1.5); hold on
hStem = stem(ts, ys, 'filled', 'Color', opt.StemColor, 'LineWidth', 1.0, ...
    'BaseValue', 0);
hRec = plot(t, yRec, 'Color', opt.RecColor, 'LineWidth', 1.4);
grid on; grid minor;
xlabel('Time (s)'); ylabel('Amplitude');
```

```

ttl = opt.Title;
if isempty(ttl), ttl = sprintf('Reconstruction: %s (f_s=%.3g Hz)', ...
    upper(method(1))+method(2:end), fs); end
title(ttl)
```

```

if opt.Legend
    lg = legend([hTrue hStem hRec], {'True', 'Samples', 'Reconstructed'}, ...
        'Location', 'best');
    set(lg, 'Interpreter', 'none');
end
if ~holdState, hold off; end
end
```

```

% --- Local sinc interpolation (windowed, truncated) ---
function y = local_sinc_reconstruct(t, ts, ys, halfLobes)
    T = ts(2) - ts(1);
    y = zeros(size(t));
    % For each sample, add shifted, windowed sinc
    for k = 1:numel(ts)
        tau = t - ts(k);
        u = tau / T;
        % ideal sinc
        s = sinc(u); % MATLAB sinc(x) = sin(pi x)/(pi x)
        % cosine window to truncate around halfLobes
        w = zeros(size(s));
        mask = abs(u) ≤ halfLobes;
        w(mask) = 0.5*(1 + cos(pi*u(mask)/halfLobes));
        y = y + ys(k) .* (s .* w);
    end
end
```

```

fun = @(t) cos(2*pi*1*t);
tspan = [0 2];
fs = 7;

figure;
plot_reconstruction_with_samples(fun, fs, tspan, 'Method', 'linear', ...
    'Title', '1 Hz Linear reconstruction @ 7 Hz');
```

```

figure;
```



```

plot_reconstruction_with_samples(fun, fs, tspan, 'Method','spline', ...
    'Title','1 Hz      Spline reconstruction @ 7 Hz');

figure;
plot_reconstruction_with_samples(fun, fs, tspan, 'Method','sinc', ...
    'SincHalfLobes', 8, ...
    'Title','1 Hz      Sinc reconstruction @ 7 Hz');

f_list = [3 6 8 13 15];
for fk = f_list
    funk = @(t) cos(2*pi*fk*t);
    fs = 7;
    figure;
    plot_reconstruction_with_samples(funk, fs, [0 1], 'Method','linear', ...
        'Title', sprintf('%d Hz      Linear reconstruction %d Hz', fk,fs));
end

```

```

clc; clear; close all;
% --- SETTINGS ---
fs      = 1.5;           % sampling rate (Hz)
tspan   = [0 2];        % [t0 tf] seconds
N       = 4000;         % points for smooth plot

% Define functions (edit as needed)
fun1 = @(t) cos(2*pi*1*t); % first function (e.g., 1 Hz)
fun2 = @(t) cos(2*pi*0.5*t); % second function (e.g., 13 Hz)

% --- TIME & SAMPLING (only for first function) ---
t0 = tspan(1); tf = tspan(2);
t  = linspace(t0, tf, N); % dense time for continuous curves
Ts = 1/fs;
ts = t0:Ts:tf + 1e-12;    % sampling instants (ONE set)

% --- EVALUATE ---
y1 = fun1(t);
y2 = fun2(t);
ys1 = fun1(ts);           % samples ONLY from first function

% --- FIGURE 1: first function+ its samples ---
figure(1); clf; hold on
plot(t, y1, 'b-', 'LineWidth', 1.6); % blue line
stem(ts, ys1, 'r', 'filled', 'LineWidth', 1.0); % red stems + filled red ...
    markers
grid on; xlabel('Time (s)'); ylabel('Amplitude');
title(sprintf('First function with samples, f_s = %g Hz', fs));
legend('f_1(t)', 'f_1 samples', 'Location', 'best');
hold off;

% FIGURE 2: first (blue), second (yellow), same red samples
figure(2); clf; hold on
plot(t, y1, 'b-', 'LineWidth', 1.6); % blue
plot(t, y2, 'g-', 'LineWidth', 1.6); % yellow
stem(ts, ys1, 'r', 'filled', 'LineWidth', 1.0); % red stems reused
grid on; xlabel('Time (s)'); ylabel('Amplitude');
title(sprintf('f_1 , f_2 , and reused f_1 samples , f_s = %g Hz', fs));

```

```
legend('f_1(t)', 'f_2(t)', 'f_1 samples', 'Location', 'best');  
hold off;
```