

Properly sampling a time-dependent signal

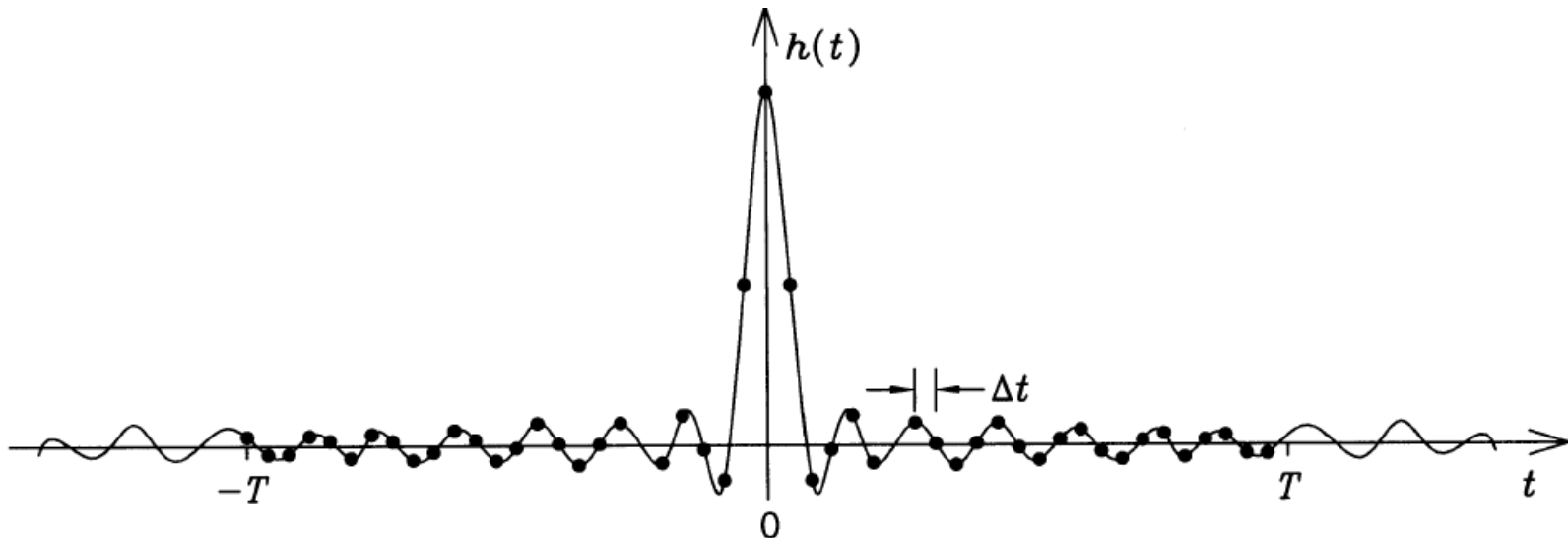
Brian Washburn

07/09/2018

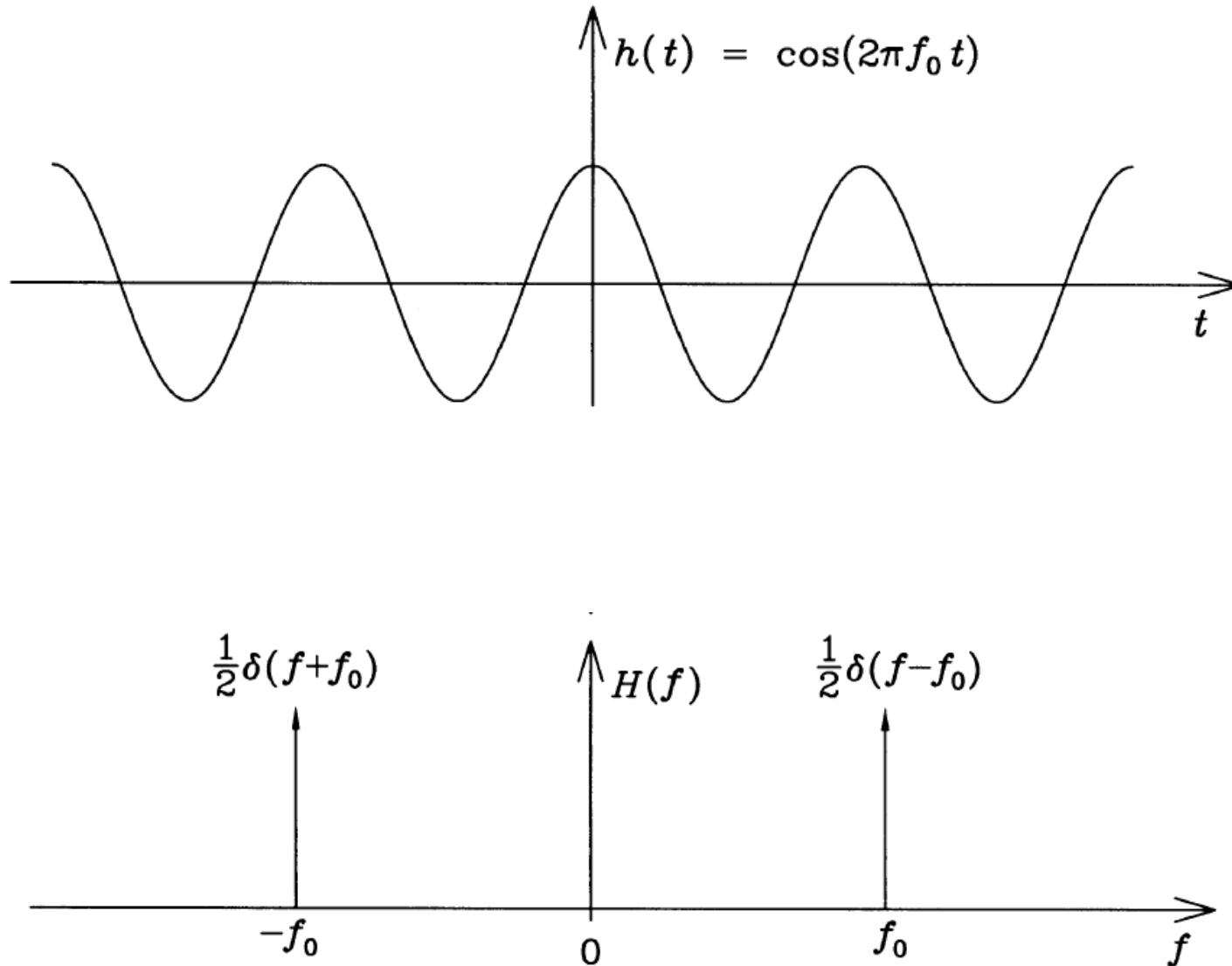
Consider measuring a signal as a function of time.

- **Sampling:** the measured signal will be sampled at the rate of $1/\Delta t$ hertz
- **Truncation:** the measured signal will be recorded over the time of $2T$ seconds

Improper sampling and truncation will cause the measured signal to have a different spectral content than the original signal.

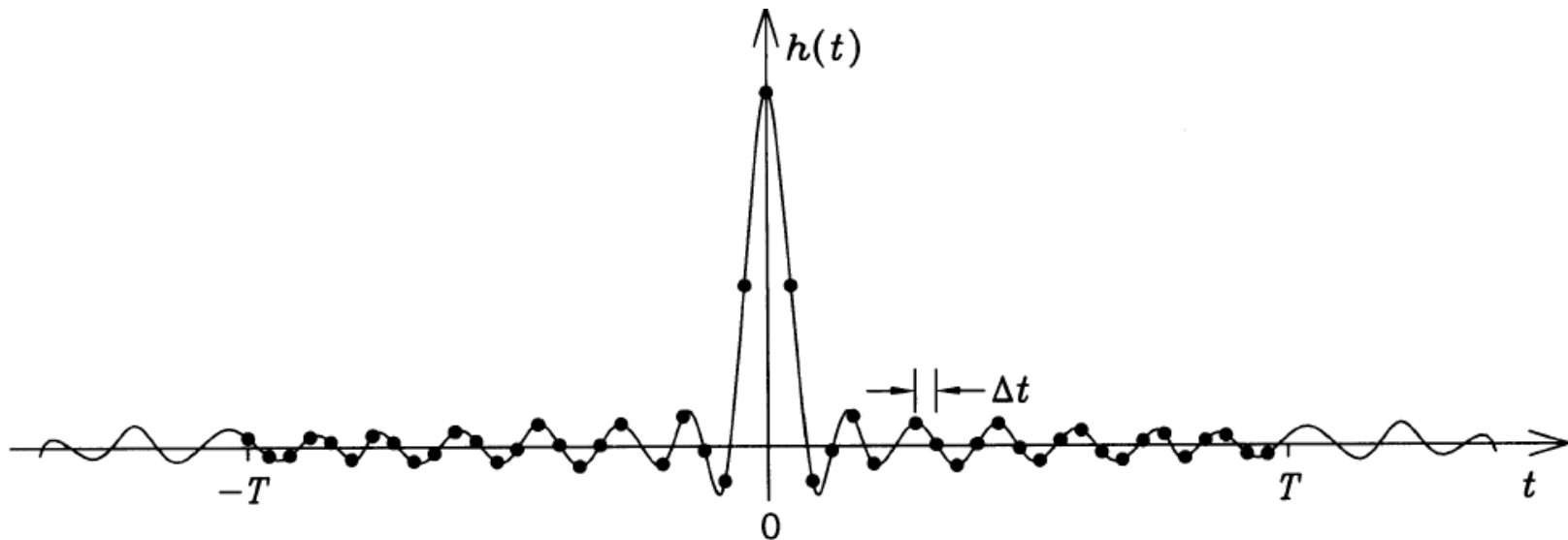


What is meant by the spectral content?
Consider the spectral content of a cosine

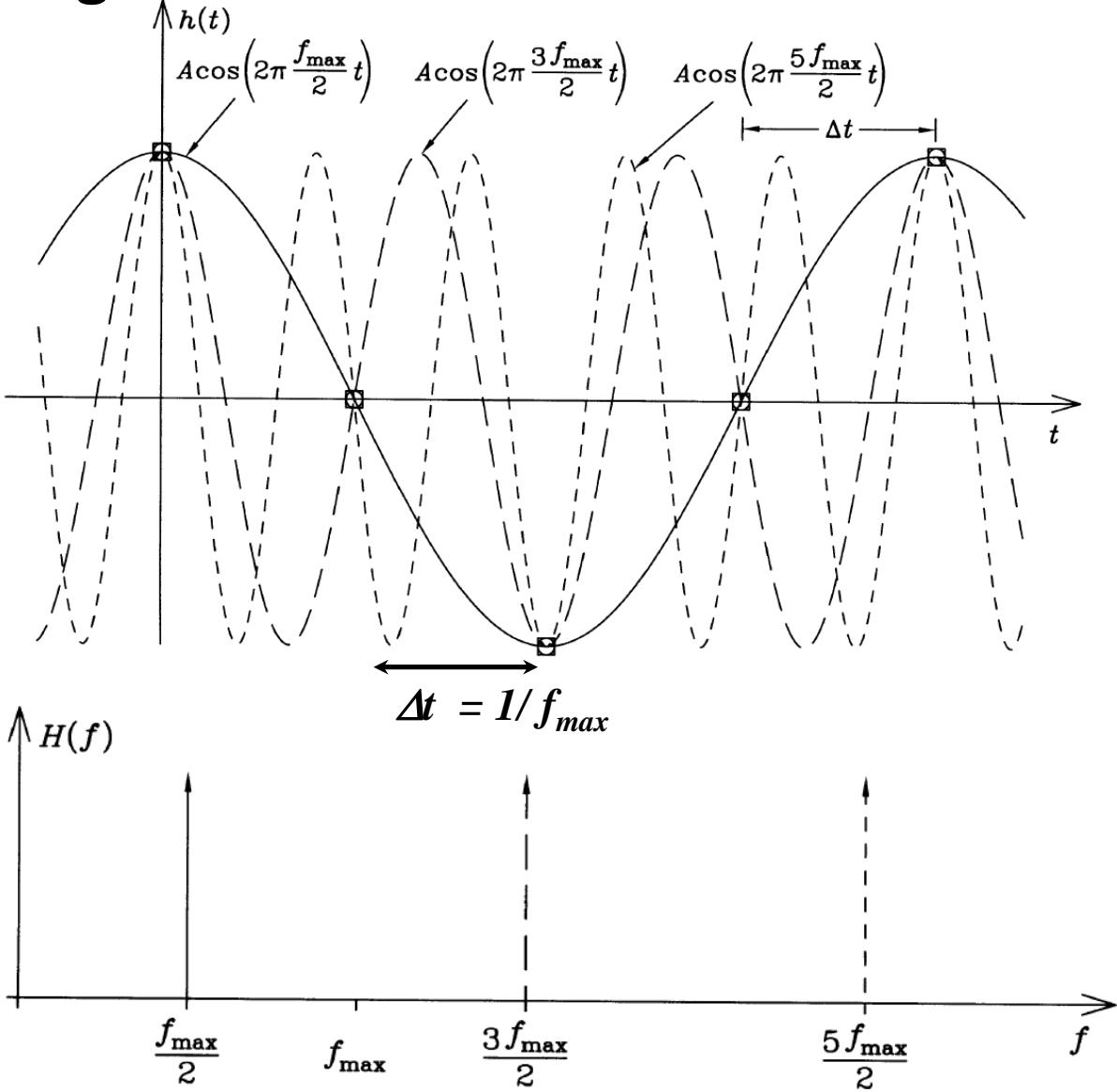


General Rules

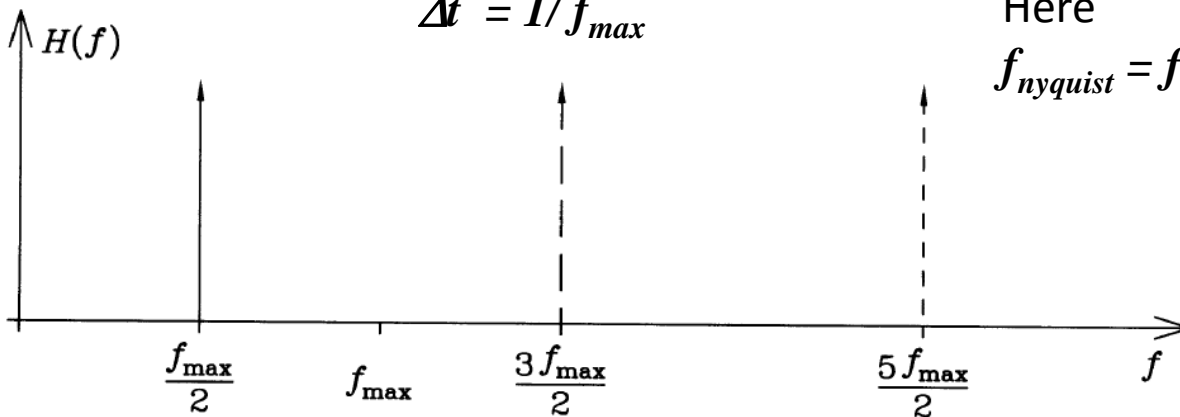
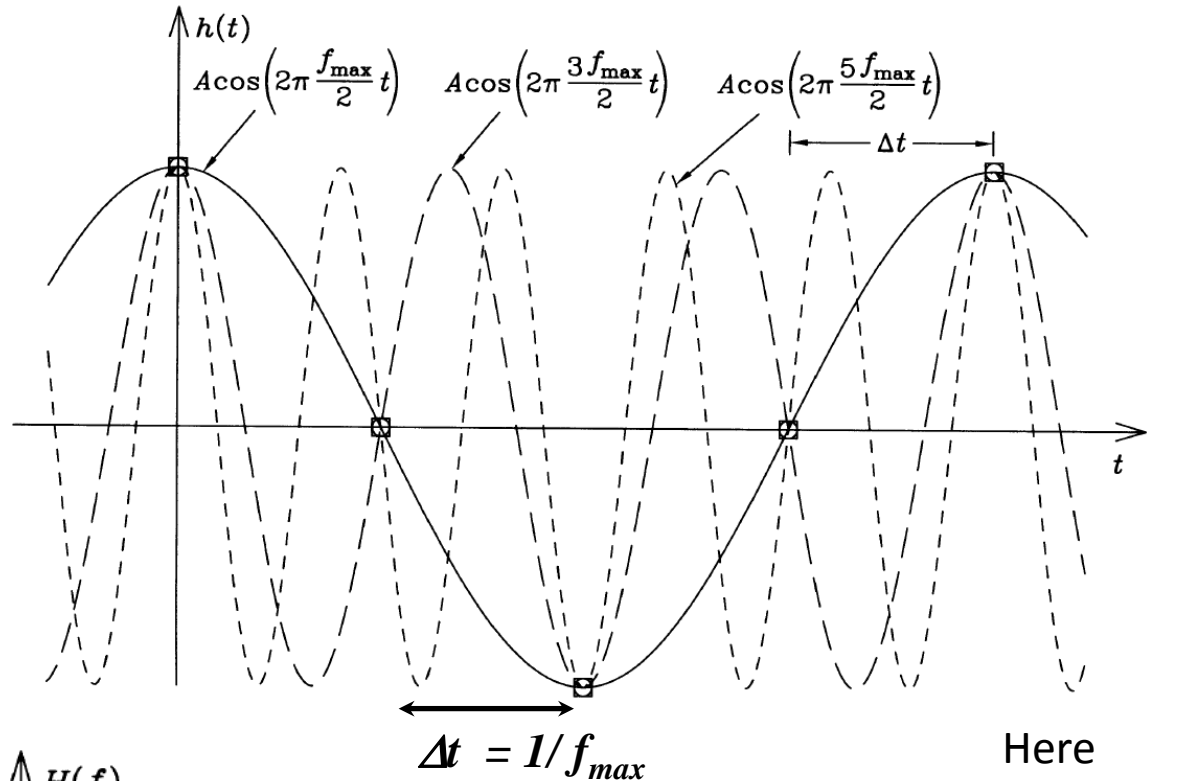
- For your signal determine highest frequency of interest f_{max}
- Nyquist Criteria: make sure that the sample rate is larger than twice the highest frequency of interest:
 $1/\Delta t > 2f_{max}$
- Make sure to take data for a time much longer than the time between points: $1/T \ll 1/(2\Delta t)$



Aliasing: creating false spectral components by improper signal sampling

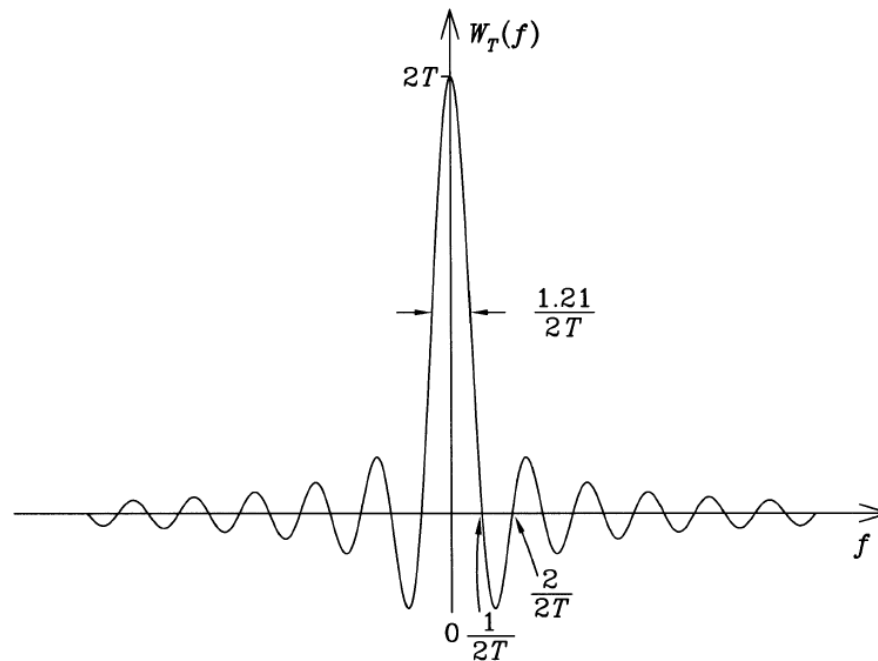
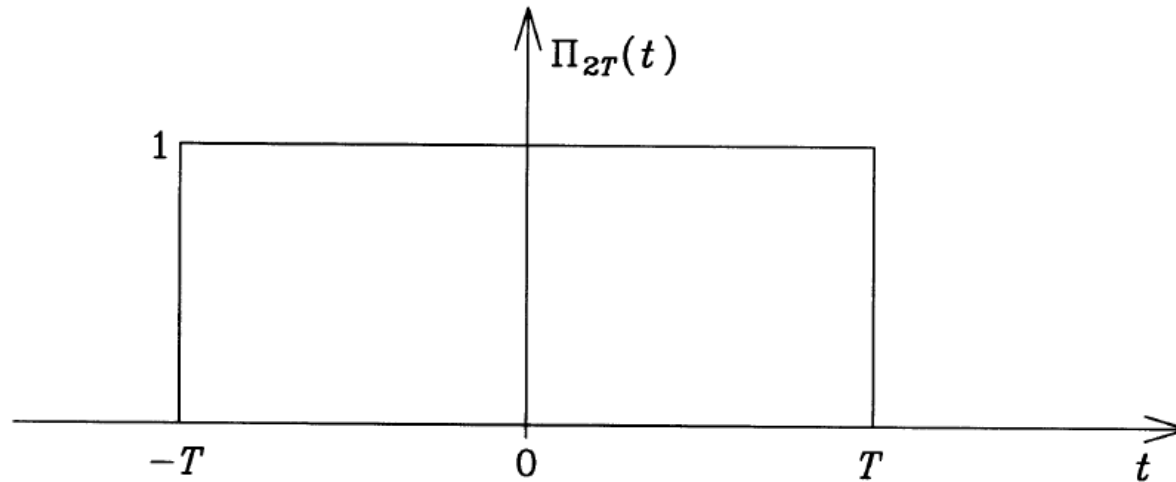


The Nyquist Frequency: to properly sample frequency f the sampling rate must be at least $2f$.

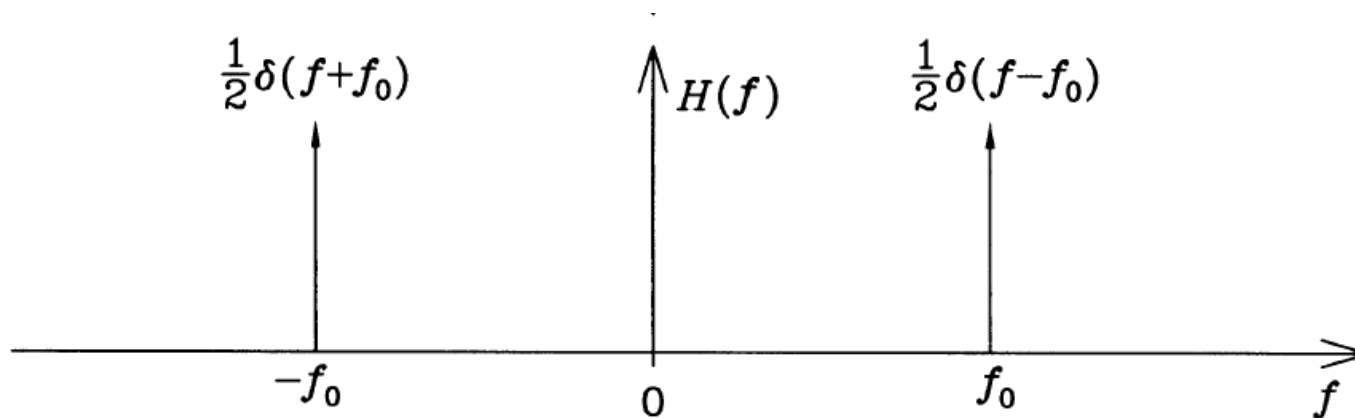
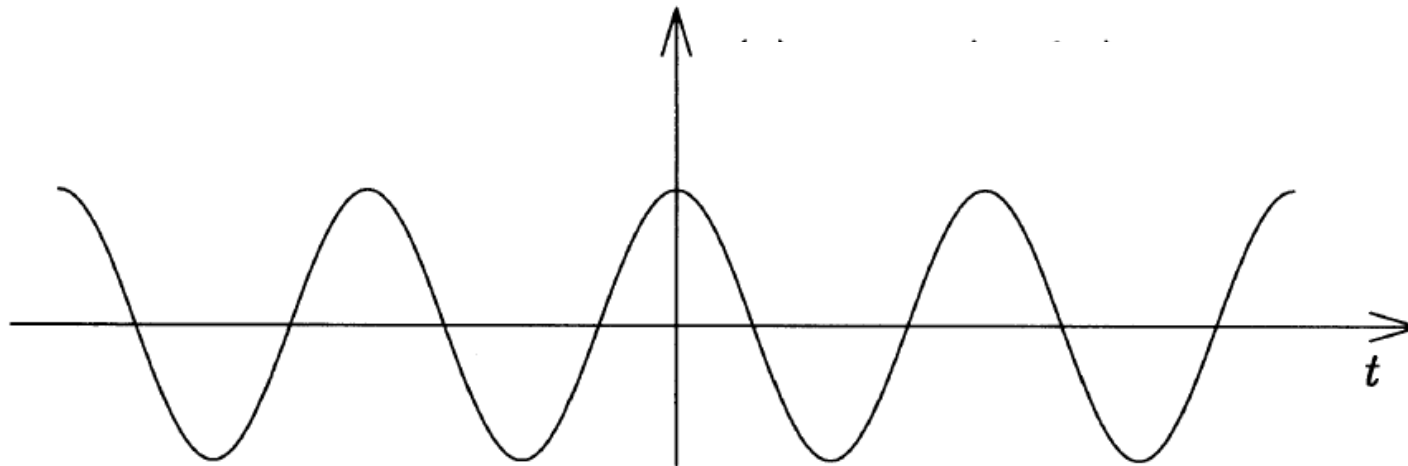


Here
 $f_{\text{nyquist}} = f_{\max} = 2 (f_{\max}/2)$

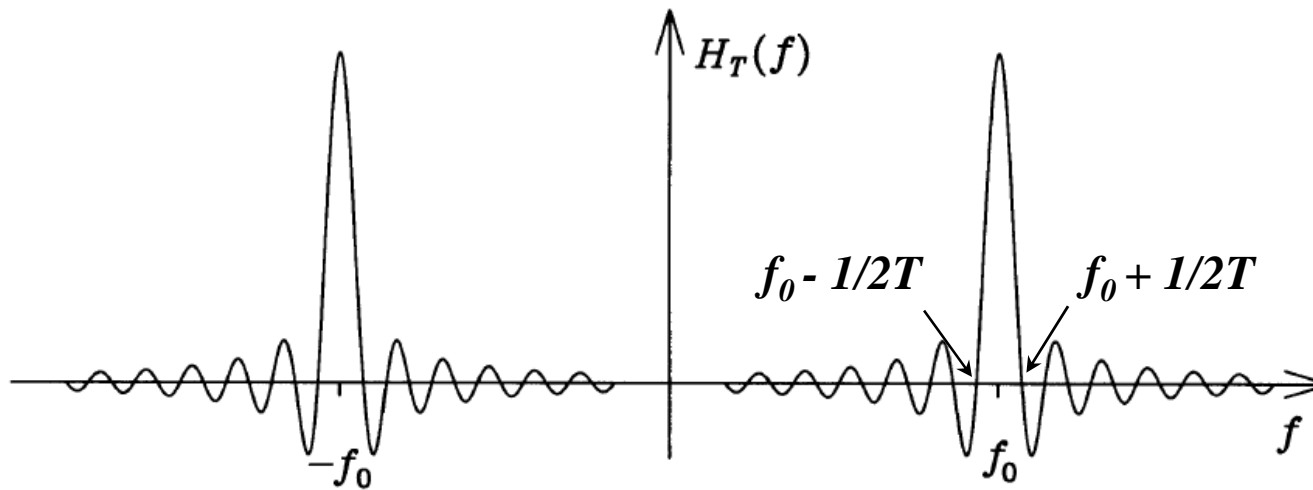
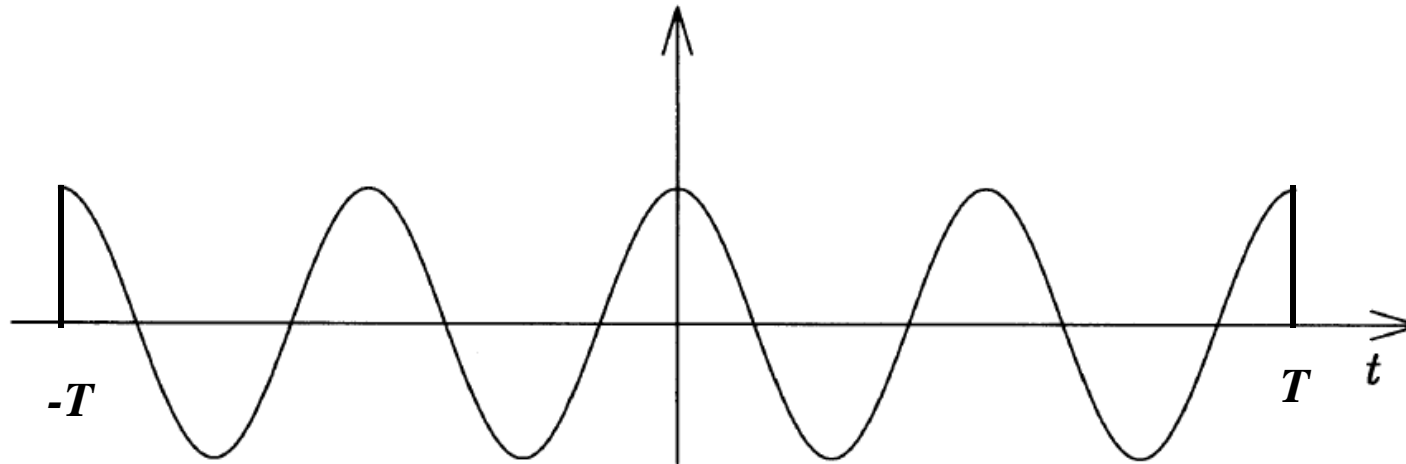
The effect of truncation: multiplying signal by a boxcar function



Cosine signal and its Fourier Transform

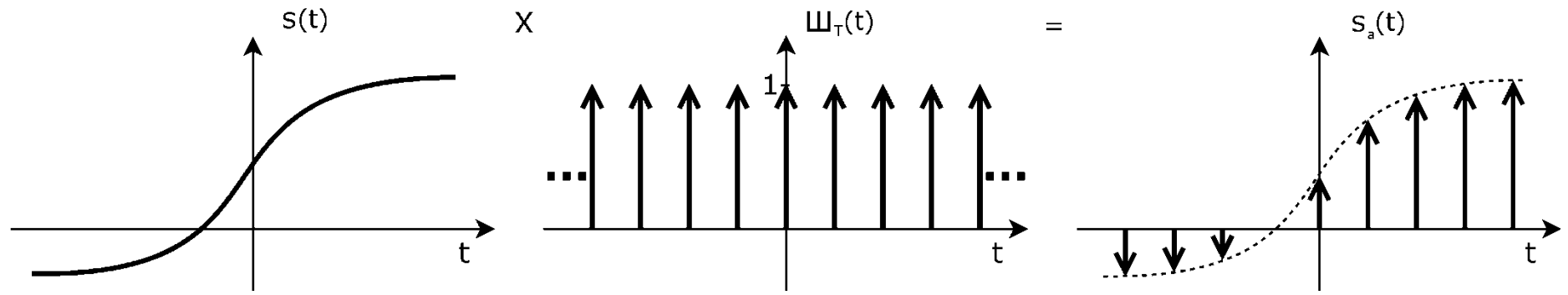


A truncated cosine and its Fourier transform



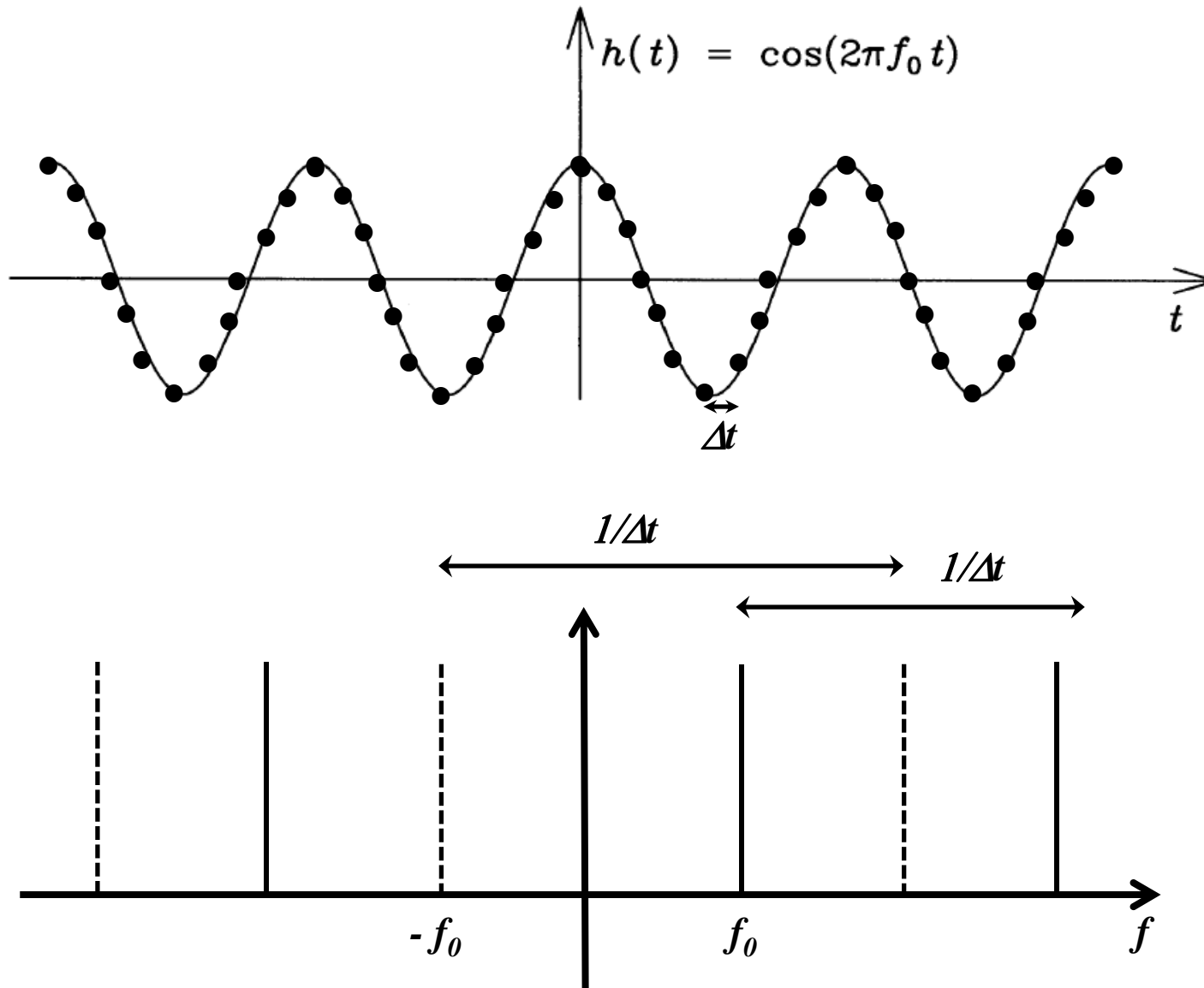
The effect of sampling at rate $1/\Delta t$:

Multiplying the actual signal by a Dirac comb, an infinite series of delta function

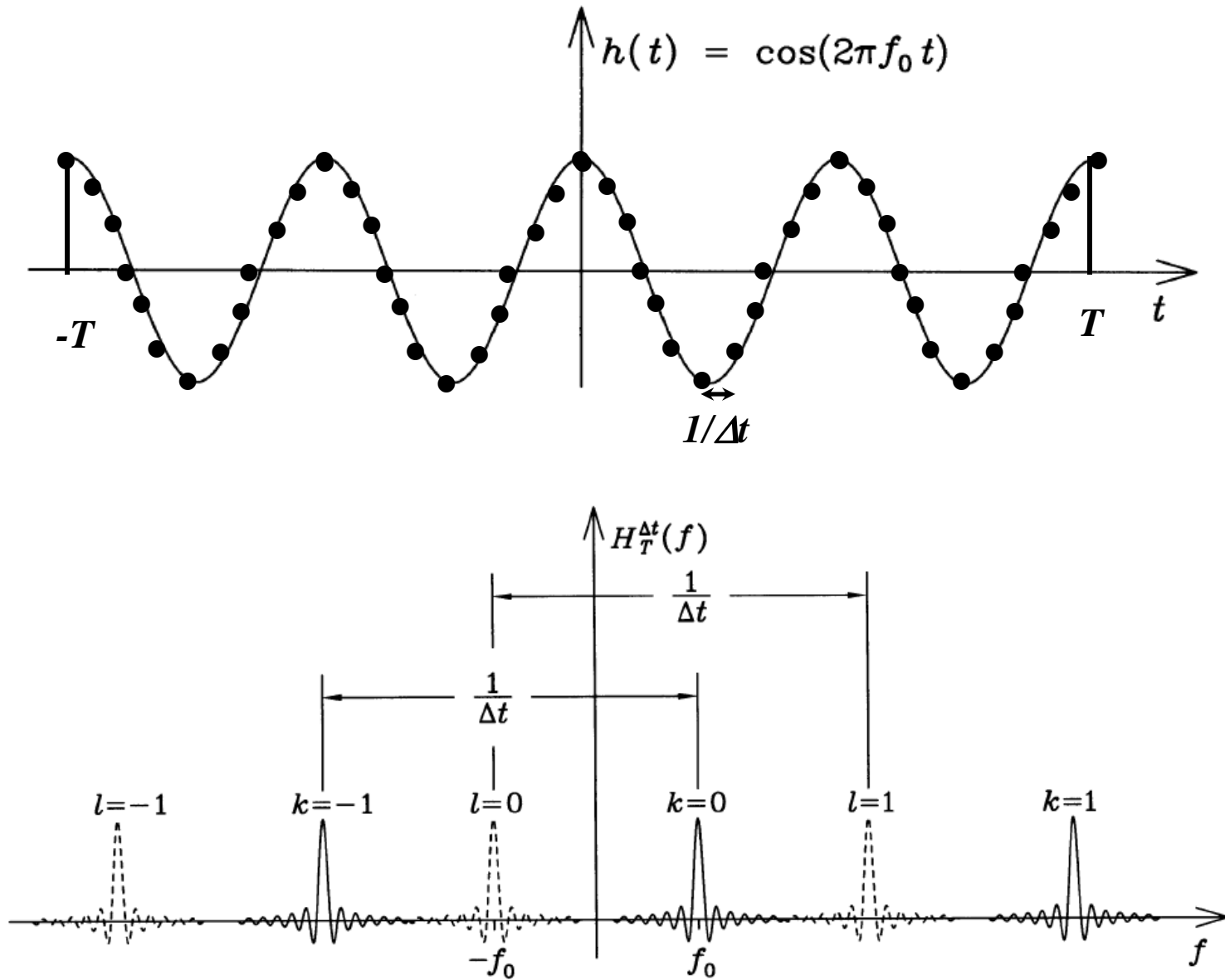


The Fourier transform of a Dirac comb times a signal will be the convolution of a Dirac comb spaced by $1/\Delta t$ and the Fourier transform of the signal

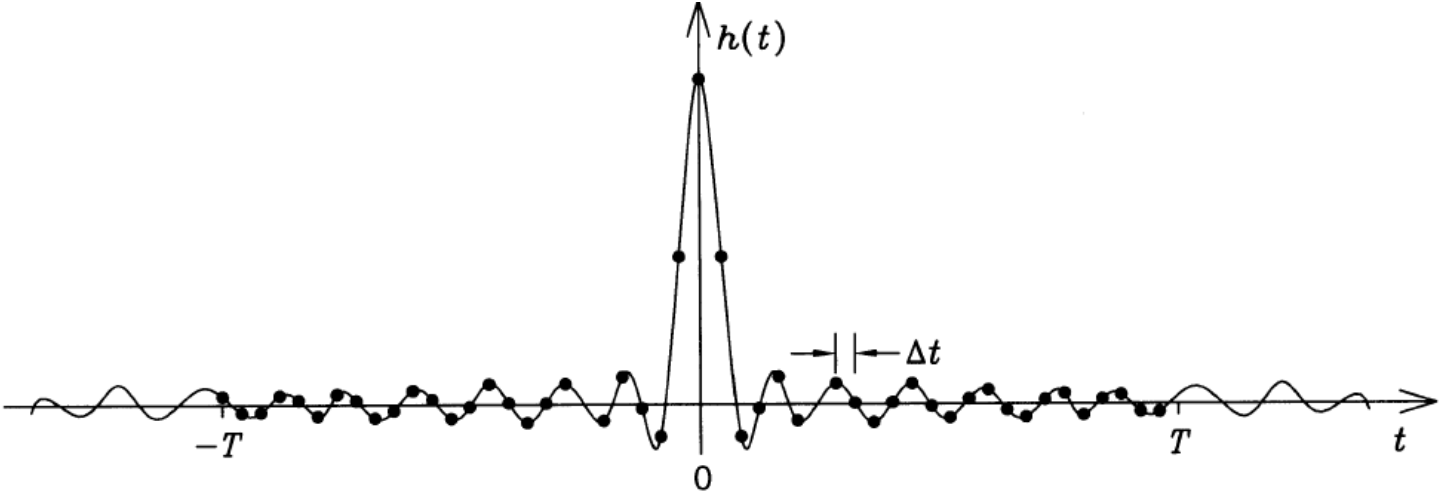
A sampled cosine and its Fourier Transform



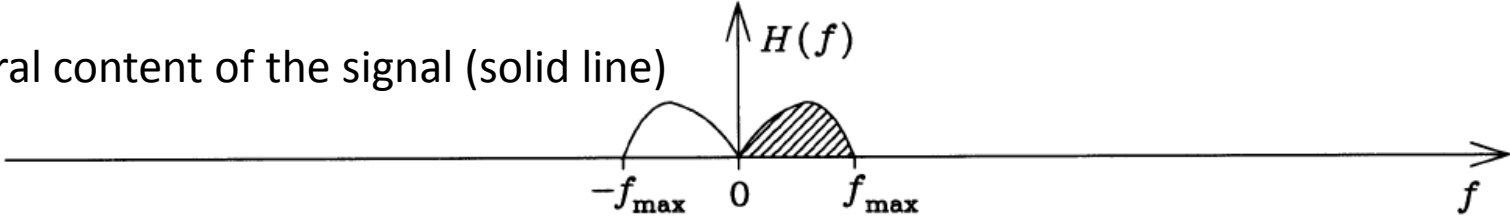
A truncated and sampled cosine and its Fourier Transform



The effect of truncation and sampling a signal



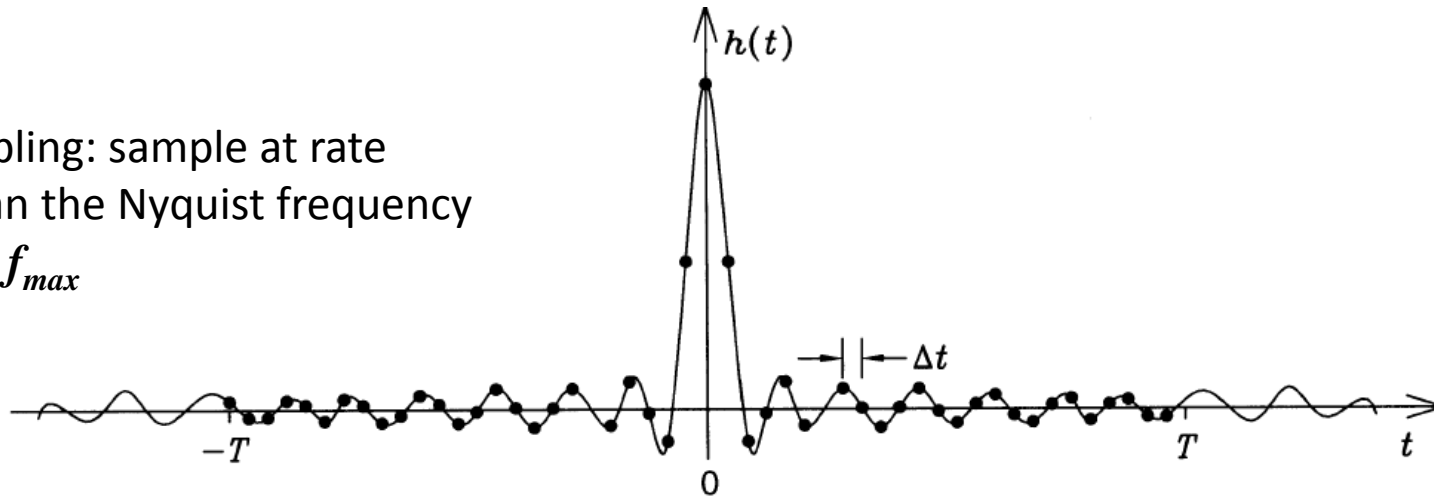
Spectral content of the signal (solid line)



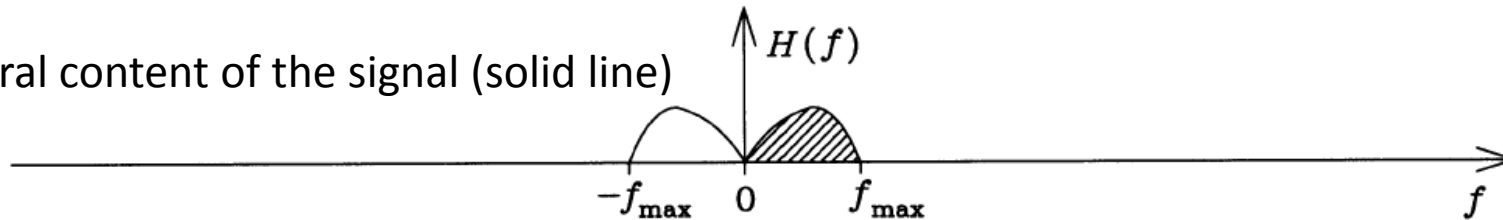
The effect of truncation and oversampling a signal

Oversampling: sample at rate
larger than the Nyquist frequency

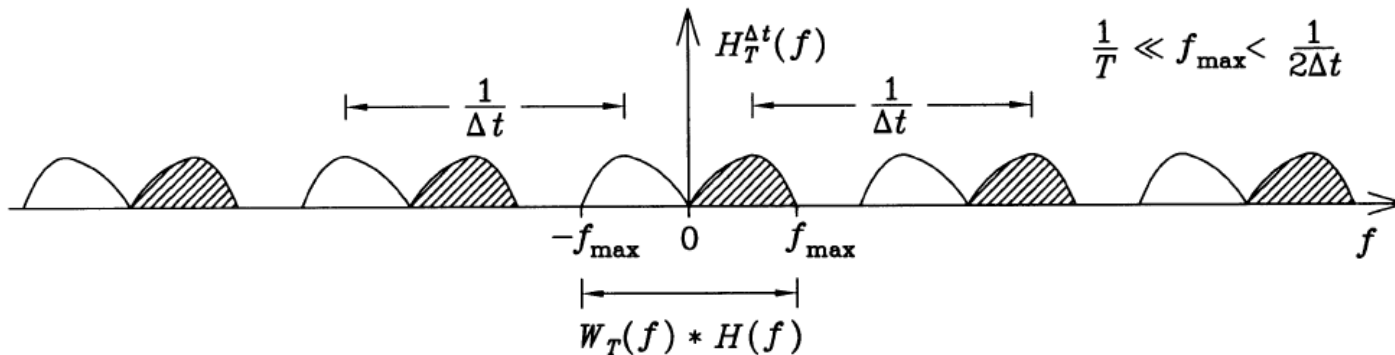
$$1/\Delta t > 2f_{\max}$$



Spectral content of the signal (solid line)



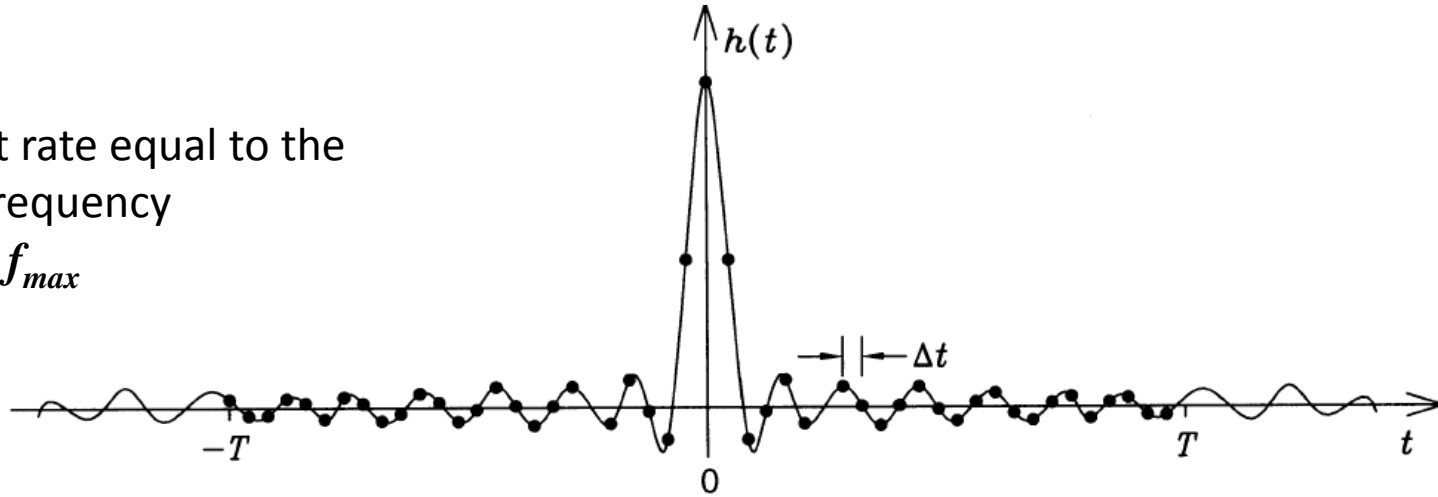
Truncated and oversampled sampled at $1/\Delta t > 2f_{\max}$



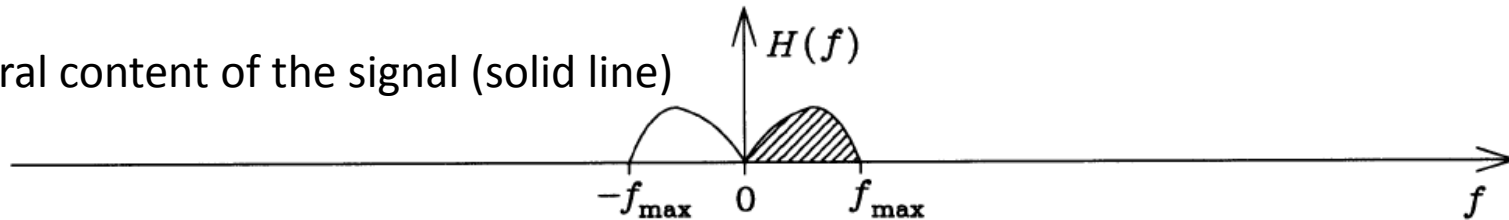
The effect of truncation and sampling at the Nyquist frequency

Sample at rate equal to the Nyquist frequency

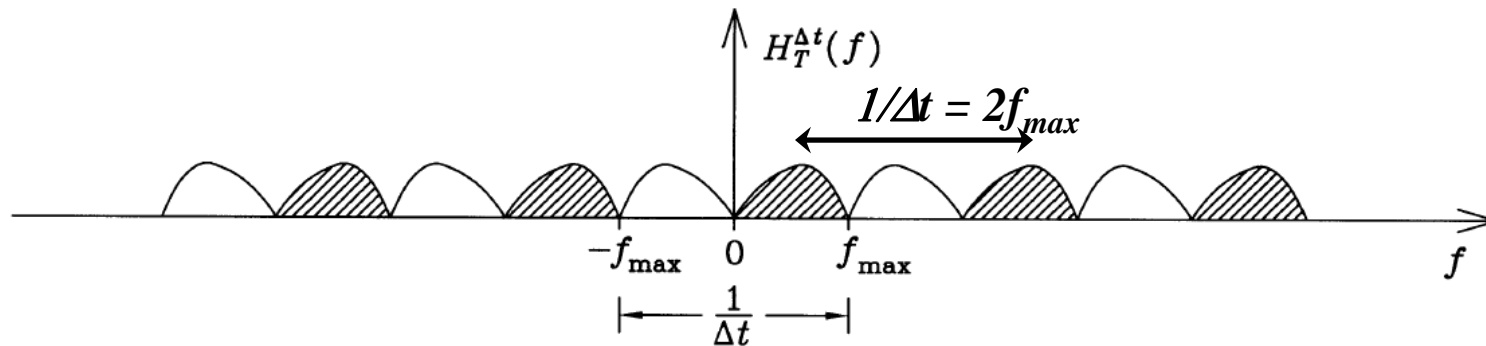
$$1/\Delta t = 2f_{max}$$



Spectral content of the signal (solid line)



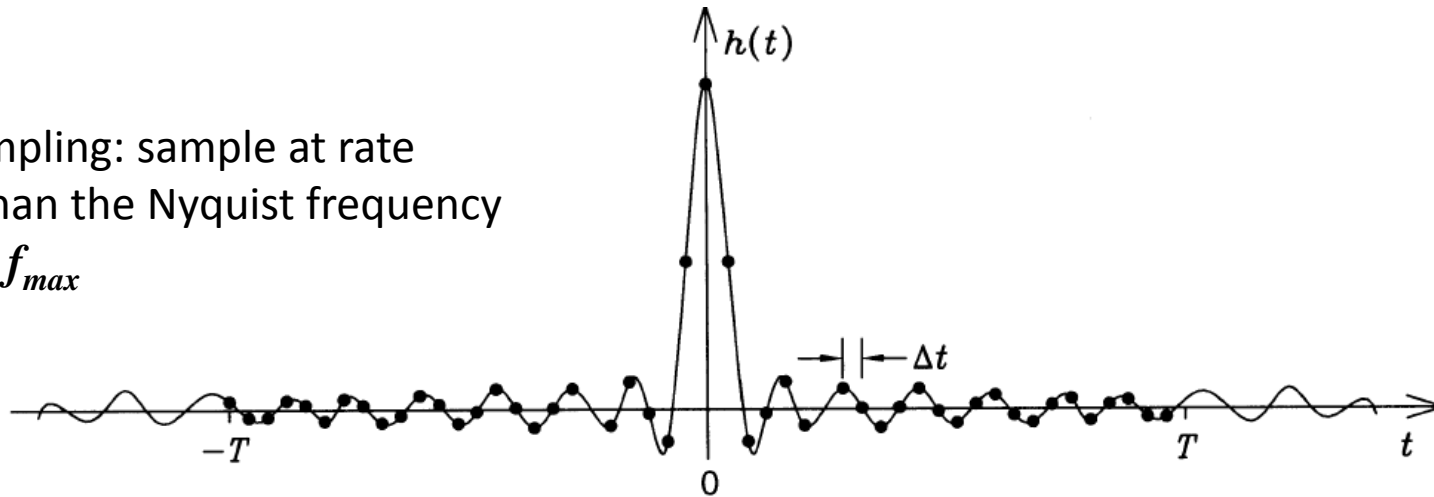
Truncated and sampled at $2f_{max}$



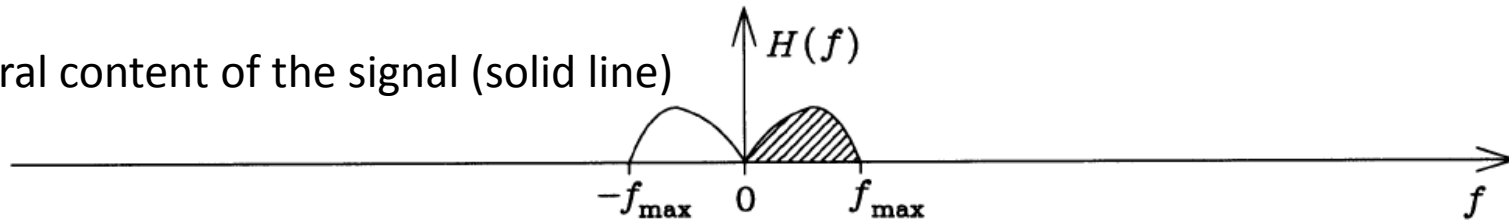
The effect of truncation and undersampling

Undersampling: sample at rate smaller than the Nyquist frequency

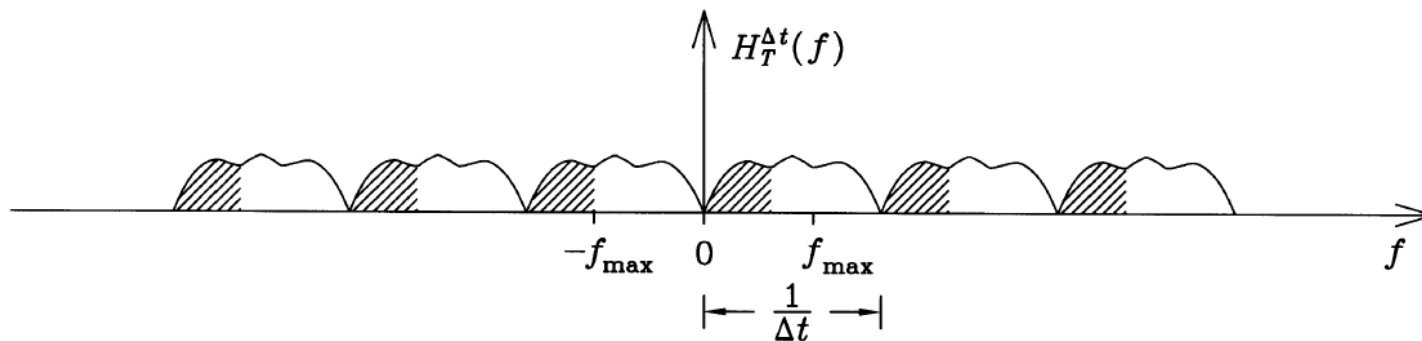
$$1/\Delta t < 2f_{max}$$



Spectral content of the signal (solid line)



Truncated and sampled sampled at $1/\Delta t < 2f_{max}$



General Rules

- For your signal determine highest frequency of interest f_{max}
- Make sure that the sample rate is larger than twice the highest frequency of interest: $1/\Delta t > 2f_{max}$
- Make sure to take data for a time much longer than the time between points: $1/T \ll 1/(2\Delta t)$

Exercise

Use a Techtronic sampling oscilloscope to measure a sinusoidal signal of frequency f . Observe aliasing by choosing a time scale, and thus a sample rate that undersamples the data.

Sample Rate = Record Length/Time Duration