

Nyquist–Shannon sampling theorem

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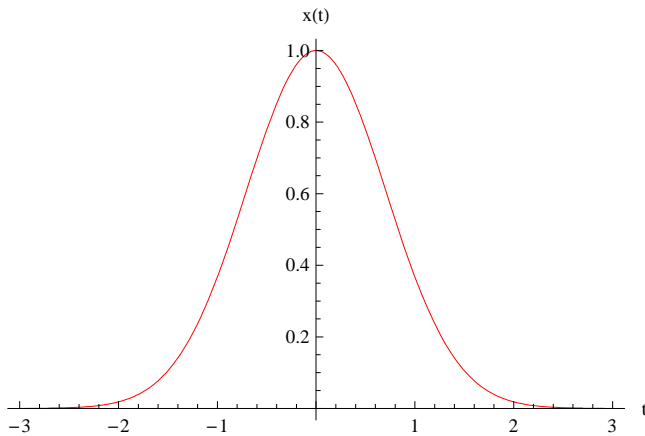
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Consider the signal x as function of time t in the form of an exponential function $x(t) = \text{Exp}[-t^2]$. Let's plot this function.

We see that for $|x| > 3$, the function is very small.

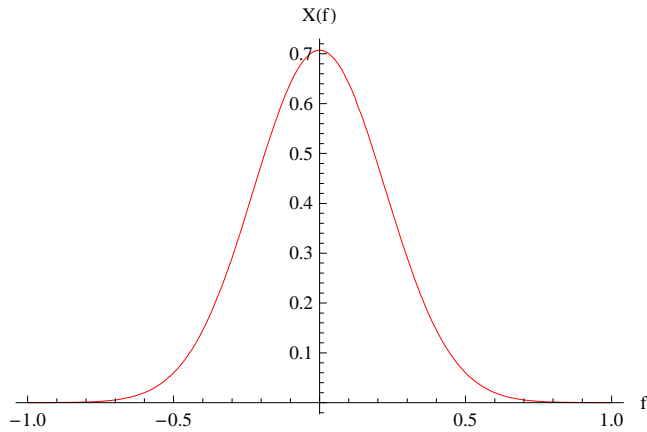
```
Clear[x, t];  
x[t_] = Exp[-t^2];  
plot1 = Plot[x[t], {t, -3, 3}, PlotRange -> All, PlotStyle -> Red, AxesLabel -> {"t", "x(t)"}]
```



The Fourier Transform of this function can be easily evaluated and is $X(f) = \frac{e^{-f^2}}{\sqrt{2}}$. If we plot the Fourier transform, we see that it becomes very small for $|f| > 1$. This indicates that the one-sided bandwidth of this function B must be around $B=1.0$.

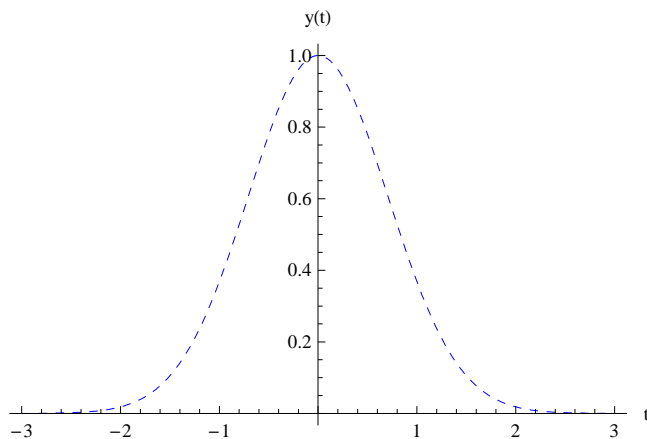
```
X[f_] = FourierTransform[x[t], t, 2 π f]
Plot[X[f], {f, -1, 1}, PlotRange → All, PlotStyle → Red, AxesLabel → {"f", "X(f)"}]
```

$$\frac{e^{-f^2 \pi^2}}{\sqrt{2}}$$



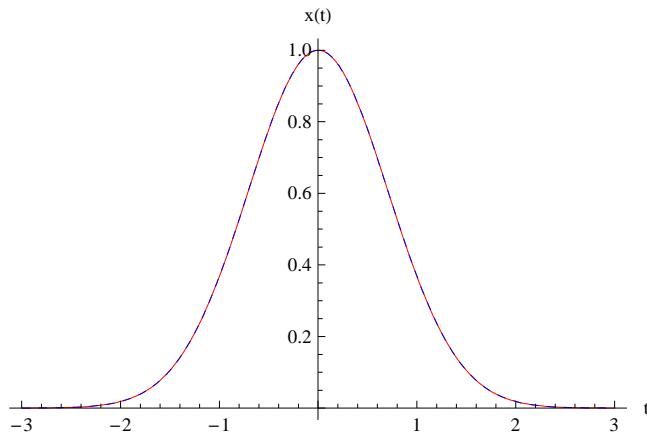
The Nyquist–Shannon sampling theorem tells us to choose a sampling rate f_s at least equal to twice the bandwidth, i.e. $f_s=2B$. The sampled signal is $x(nT)$ for all values of integer n . In practice, a finite number of n is sufficient in this case since $x(nT)$ is vanishingly small for large n . We chose $n_{\text{Max}}=10$ for the maximum value of n . Now, from the digital signal $x(nT)$, we try to reconstruct the analog signal $y(t)$ which should ideally be very close to $x(t)$ if the sampling rate is properly chosen. We plot $y(t)$

```
B = 1;
fs = 2 B;
T = 1 / fs;
nMax = 10;
y[t_] = Sum[x[n T] Sinc[π (t - n T) / T], {n, -nMax, nMax}];
plot2 = Plot[y[t], {t, -3, 3}, PlotRange → All,
  PlotStyle → {Dashed, Blue}, AxesLabel → {"t", "Y(t)"}]
```



Let's compare $x(t)$ in red and $y(t)$ in blue. They are right on top of each other because we chose the sampling rate according to the sampling theorem.

```
Show[plot1, plot2]
```



Let's now sample the signal at a lower rate than what the sampling theorem suggests. For example, we choose our bandwidth B too small, say $B = 0.5$. The original signal $x(t)$ differs from the reconstructed signal $y(t)$.

```
B = 0.5;
```

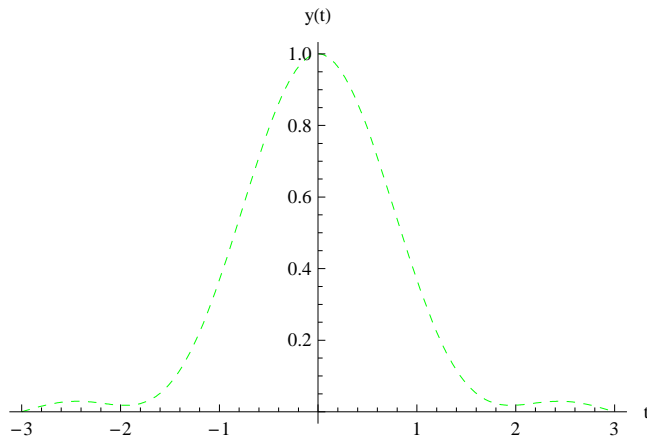
```
fs = 2 B;
```

```
T = 1 / fs;
```

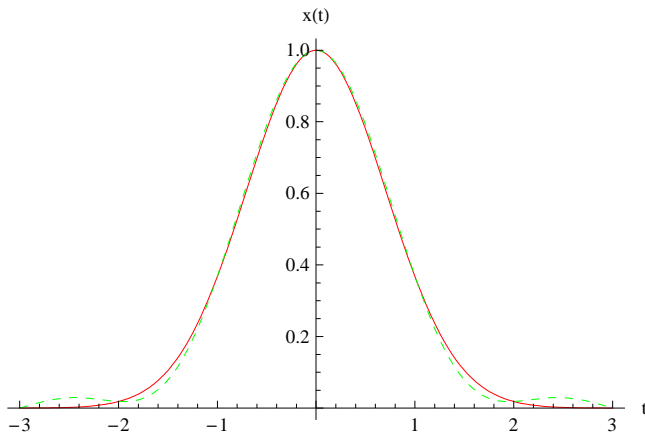
```
nMax = 10;
```

$$y[t_] = \sum_{n=-nMax}^{nMax} x[n T] \text{Sinc}\left[\pi \frac{t - n T}{T}\right];$$

```
plot3 = Plot[y[t], {t, -3, 3}, PlotRange -> All,
  PlotStyle -> {Dashed, Green}, AxesLabel -> {"t", "y(t)"}]
```

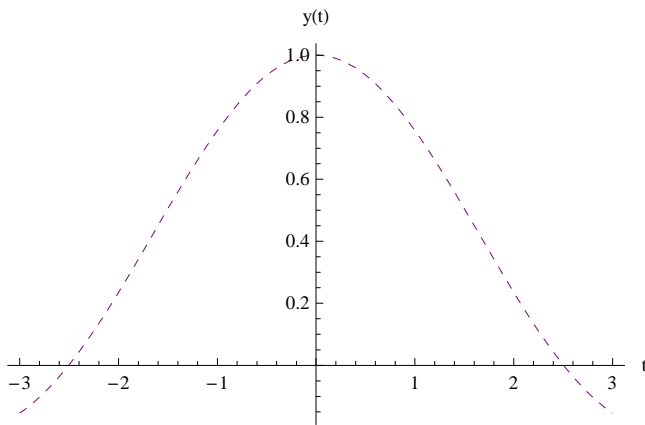


```
Show[plot1, plot3]
```



If we lower the sampling rate even more, the disagreement between $x(t)$ and $y(t)$ becomes more apparent.

```
B = 0.2;
fs = 2 B;
T = 1 / fs;
nMax = 10;
y[t_] = Sum[x[n T] Sinc[Pi (t - n T) / T], {n, -nMax, nMax}];
plot4 = Plot[y[t], {t, -3, 3}, PlotRange -> All,
  PlotStyle -> {Dashed, Purple}, AxesLabel -> {"t", "y(t)"}]
```



```
Show[plot1, plot4]
```

