

Fundamentals of Information Theory II

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Reading

- <http://mathworld.wolfram.com/SamplingTheorem.html>
- <http://cwx.prenhall.com/bookbind/pubbooks/cyganski/chapter0/deluxe.html> - demo applets

Representing signals: frequency, time, amplitude

- Baseband signals: $F(t)$ limited in frequency in the interval $[-W, W]$.
- Broadband signals (carrier based):

Moved in the interval $[W_c - W, W_c + W]$

Example: $F(t) = F_s(t) \sin(W_c t)$

Representing signals by sample points

- Theoretically, infinite number of points are required.
- Locating a point in space: coordinate systems.
- Locating a function $F(t)$:

vector $F(t_1), F(t_2), \dots, F(t_n)$

t_n - sample points

- Orthonormal sets of functions: $\{G_n(X)\}$, where:

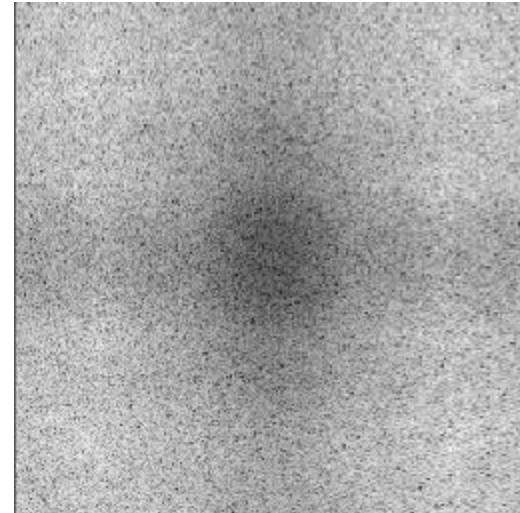
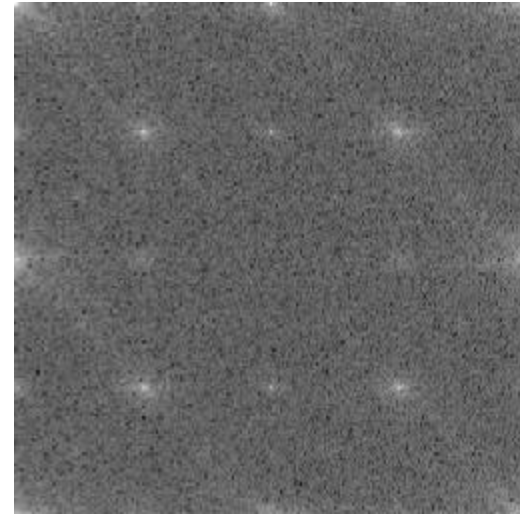
$$\int_a^b G_n(x)G_m(x)dx = \begin{cases} 1 & \text{if } n = m; \\ 0 & \text{if } n \neq m. \end{cases}$$

- Expansion of a function (Fourier series):

$$F(x) = \sum_{n=-\text{inf}}^{n=+\text{inf}} C_n G_n(x)$$

$$C_m = \int_{-\text{inf}}^{+\text{inf}} F(x)G_m(x)dx$$

Displaying 2D Magnitude Spectra of Quantization Noise



Sampling theorem

- Orthonormal set: $\{sinc_n(t)\}$, $t \in [-T, +T]$, $T \rightarrow \text{inf}$.

$$sinc_n(t) = \frac{\sin(2\pi W(t - \frac{n}{2W}))}{2\pi W(t - \frac{n}{2W})}$$

- Fourier expansion of $F(t)$:

$$F(t) = \sum_{n=-\text{inf}}^{n=+\text{inf}} C_n \frac{\sin(2\pi W(t - \frac{n}{2W}))}{2\pi W(t - \frac{n}{2W})}$$

- Sample points: $\dots, t_1, t_2, \dots, t_n, \dots$, where $t_n = \frac{n}{2W}$, $n \in [-\text{inf}, +\text{inf}]$ and $[-W, +W]$ is the frequency interval.
- Let $t = t_n$, for $n \in [-\text{inf}, +\text{inf}]$. Then all terms in the Fourier expansion of $F(t)$ above are 0 except one, $\frac{\sin(0)}{0} = 1$. That is, $F(t_n) = C_n$.
- Hence (sampling theorem):

$$F(t) = \sum_{n=-\text{inf}}^{n=+\text{inf}} f(t_n) \frac{\sin(2\pi W(t - \frac{n}{2W}))}{2\pi W(t - \frac{n}{2W})}$$

- Constraints:

$F(t)$ is limited in frequency in $[-W, W]$,

There are infinite number of sample points (infinite time duration)

AD and DA conversion

- quantization (thresholds)
- amplifiers vs. repeaters
- generally very low error rates (regenerating digital values at relay points)

Error detection and correction

- Binary case: 000,111 for 0,1 - one error correcting code
- Arbitrary waveform: three instead of one sample point - widening the bandwidth (2TW dimensions)