Signal Processing

Reminder

N	continue	discret
1	$ f _1 = \int_{\mathbf{R}} f(t) dt$	$\sum_{n=-\infty}^{+\infty} u_n $
2	$ f _2 = \sqrt{\int_{\mathbf{R}} f(t) ^2 dt}$	$\sqrt{\sum_{n=\infty}^{\infty} u_n ^2}$
∞	$ f _{\infty} = \sup_{t} f(t) $	$sup_n u_n $
Р	$ f _1 = \int_{\mathbf{R}} f(t) ^p dt^{1/p}$	$\left(\sum_{n=\infty}^{\infty} u_n ^p\right)^{1/p}$

- The uniform convergence, with the ∞ -Norm.
- The absolute convergence, with the 1-Norm
- The convergence in energy, with the 2-Norm

Convolution

Continus time: $y(t)=(h*x)(t)=\int_{\mathbf{R}}h(u)x(t-u)du$ Discrete time: $y[t]=(h*x)[t]=\sum_{k\in\mathbf{Z}}x[t-k]$ The convolution is commutative and the neutral element is Dirac distribution $\delta(t)$

The Cross and auto correlation

 $(x \star y)(t) = \int_{\mathbf{R}} \overline{x(u)} y(t+u) du = \int_{\mathbf{R}} x(t-u) \overline{y(u)} du$ The cross correlation is not a commutative operation. Let $\tilde{x} = x(-t)$, then $(x \star y)(t) = (\tilde{x} \star y)(t)$

The same think of discret time.

Digital signals are obtained from analog signals by sampling and quantization.

Filter: A filter is a linear time-invariant system. It is perfectly determined by its impulse response h.

Numerical filter with impulse response h[t]. The output y[t] of an analog filter with input (also called excitation) x[t] is obtained by :

$$y[y] = (h * x)[t] = \sum_{k \in \mathbf{R}} h[t - k]x[k]$$

Causality: A signal is causal iff its start at (or after) the date t=0.

Anti-causality A signal is anti-causal iff its stop a before the date t=0.

A signal, or system, which is not causal or anti-causal is acausal.

For systems, causality implies that the output does not depends on the futur of the input.

Stability A signal a stable iff it is summable.

Energy The energy of a signal is its squared 2-norm.

Elementary signals

Numerical Dirac impulse

$$\delta_k[n] = \begin{cases} 1 & \text{if } n = k \\ 0 & \text{otherwise} \end{cases}$$

Heaviside

$$\Theta[n] = \begin{cases} 1 & \text{if } n \ge 0\\ 0 & \text{otherwise} \end{cases}$$

Rectangular window

$$\Pi(t) = \begin{cases} 1 & \text{if } -\frac{1}{2} \le t \le \frac{1}{2} \\ 0 & \text{otherwise} \end{cases}$$

Fourier Analysis

Fourier Coefficients (Fourier transform of periodic functions)

$$\hat{f}[n] = c_n(f) = \frac{1}{T} \int_0^T f(t) \exp^{-i\frac{2\pi}{T}nt} dt$$

 $\hat{f}[n]$ measure how f is close the pure frequency

Spectrum of a periodic signal The spectrum is the set of the squared modulus of the fourier coefficients: $spectrum(f) = |\hat{f}[n]|^2$

Propreties

Time shift:

for $f_a(t) = f(t+a)$ then $\hat{f}_a[n] = \exp^{i\frac{2\pi}{T}na} \hat{f}[n]$ Derivation:

 $f^{(k)}[n] = (\exp^{i\frac{2\pi}{T}n})^k \hat{f}[n]$

Fourier Serie

$$S(f)(t) = \sum_{n=-\infty}^{+\infty} \hat{f}[n] \exp^{i\frac{2\pi}{T}nt}$$

Theorem Let $f,g \in L^2([0,T])$ and $\{\hat{f}[n]\}_{n\in\mathbb{Z}}$ and $\{\hat{g}[n]\}_{n\in\mathbb{Z}}$ their respective Fourier coefficients. Then

$$||f||^2 = \frac{1}{T} \int_{-T/2}^{T/2} |f(t)|^2 dt = \sum_{n=-\infty}^{+\infty} |\hat{f}[n]|^2$$

$$\langle f, g \rangle = \frac{1}{T} \int_{-T/2}^{T/2} f(t) \bar{g}(t) dt = \sum_{n=-\infty}^{+\infty} \hat{f}[n] \overline{\hat{g}}_n$$

Fourier Analysis

Coefficients de Fourier trigonométriques

$$a_0(f) = \frac{1}{T} \int_0^T f(t) dt$$

$$a_n(f) = \frac{2}{T} \int_0^T f(t) \cos\left(\frac{2\pi}{T} nt\right) dt \quad \forall n \ge 1$$

$$b_n(f) = \frac{2}{T} \int_0^T f(t) \sin\left(\frac{2\pi}{T} nt\right) dt \quad \forall n \ge 0$$

$$a_0(f) + \sum_{n=1}^{+\infty} a_n(f) \cos\left(\frac{2\pi}{T}nt\right) + \sum_{n=1}^{+\infty} b_n(f) \sin\left(\frac{2\pi}{T}nt\right)$$

Plancherel-Parseval's Theorem
$$|f|^2 = |a_0(f)|^2 + \frac{1}{2} \sum_{n=1}^{+\infty} |a_n(f)|^2 + \frac{1}{2} \sum_{n=1}^{+\infty} |b_n(f)|^2$$

continuous time

Fourier Transform

$$\hat{f}(\nu) = \int_{-\infty}^{+\infty} f(t)e^{-i2\pi\nu t} dt$$

with ν frequency in Herz and the pulsation $\omega = 2\pi\nu$

Inverse Fourier Transform

$$f(t) = \int_{\mathbb{R}} \hat{f}(v)e^{i2\pi vt} dv$$

Properties

- 1. Convolution:
- 5. Scaling: let $q_s(t) = f(t/s)$

$$\widehat{f * q}(\nu) = \widehat{f}(\nu)\widehat{g}(\nu)$$

- $\hat{q}_s(\nu) = |s| \hat{f}(s\nu)$
- 2. Multiplication:
- 6. Time derivation:

$$\widehat{f \cdot g}(\nu) = (\widehat{f} * \widehat{g})(\nu)$$

- $\widehat{f(p)}(\nu) = (i2\pi\nu)^p \widehat{f}(\nu)$
- 3. Translation:
- let $g_a(t) = f(t-a)$
- 7. Frequency derivation: let $q_n(t) = (-i2\pi t)^p f(t)$

$$\hat{g}_a(\nu) = e^{-i2\pi a\nu} \hat{f}(\nu)$$

$$\hat{q}_n(t) = \hat{f}^{(p)}(\nu)$$

- 4. Modulation:
- let $q_{\theta}(t) = e^{i2\pi\theta t} f(t)$
- 8. Complexe conjugation

$$\hat{g}_{\theta}(\nu) = \hat{f}(\nu - \theta)$$

$$\widehat{\bar{f}}(\nu) = \overline{\widehat{f}(-\nu)}$$

Fourier Analysis

Finite Fourier Transform

Analog Filtering

Bilateral Laplace transform

$$\hat{x}(p) = \int_{-\infty}^{+\infty} x(t)e^{pt}dt$$

Reference

References