

impulse: $\delta(t) = \frac{d}{dt}u(t)$
 $\int_{-\infty}^{\infty} \delta(t-t_0) dt = 1$
 $\int_{-\infty}^{\infty} \delta(t-t_0) f(t) dt = f(t_0)$

$$\delta(t) = \frac{d}{dt}u(t)$$

$$\delta(t) = \frac{d}{dt}u(t)$$

$$\int_{-\infty}^{\infty} \delta(t-t_0) dt = 1$$

convolution:

$$\delta \rightarrow LTI \rightarrow h$$

$$y[n] = x[n-3]$$

$$x[n] \rightarrow LTI \rightarrow x[n-3]$$

$$h_1[n] = \delta[n-3]$$

$$a(2x[n-1]) + b(2x[n-1])$$

$$x[n] = \sum_{k=-\infty}^{\infty} x[k]\delta[n-k]$$

$$\int x(\tau)\delta(t-\tau) d\tau = x(t)$$

$$x[n] * \delta[n-n_0] = x[n-n_0]$$

- Commutative** - can flip either signal
 $x(t) * h(t) = h(t) * x(t)$
- Associative** - order doesn't matter
 $x(t) * (h_1(t) * h_2(t)) = (x(t) * h_1(t)) * h_2(t)$
- Distributive** - divide and conquer
 $x(t) * (h_1(t) + h_2(t)) = (x(t) * h_1(t)) + (x(t) * h_2(t))$

Results so far...

$$u(t) * u(t) = \begin{cases} t & t < 0 \\ 0 & t > 0 \end{cases}$$

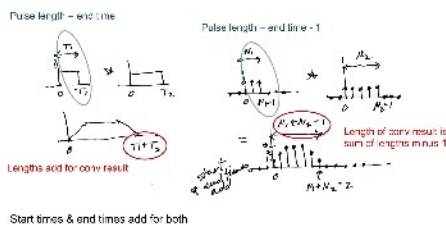
$$u(t) * e^{-at}u(t) = \frac{1}{a}(1 - e^{-at})u(t)$$

$$u(t) * e^{-at} = \frac{1}{a}e^{-at}u(t)$$

$$p(t) * p(t) = \begin{cases} 0 & t < 0 \\ 1 & 0 < t < 1 \\ 2-t & 1 < t < 2 \end{cases}$$

start and end time:

DT vs. CT convolution (cont.)

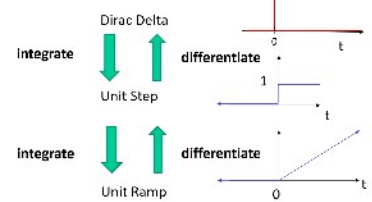


properties:

System Property Testing using Impulse Response $h(t)$

- Since the impulse response $h(t)$ fully specifies an LTI system, we can use it in property testing
- Gives additional tools to test system properties (easy tests for LTI systems)
- Key results:
 - Memoryless $\leftrightarrow h(t) = T[\delta(t)] = A\delta(t)$
 - Causal system $\leftrightarrow h(t) = 0$ for $t < 0$
 - Stable system $\leftrightarrow \int_{-\infty}^{\infty} |h(t)| dt < \infty$ (absolutely integrable)
 - Invertible \leftrightarrow there exists $h_2(t)$ such that $h(t) * h_2(t) = \delta(t)$

Signal Relationships

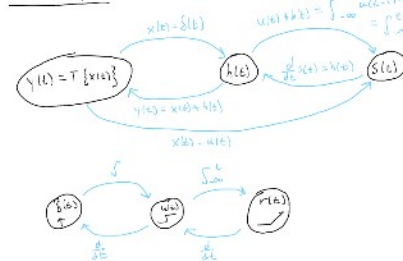


For DT, we used sum instead of integral, and first difference instead of differentiating

$$s(t) = r(t+2) - 2r(t) + 2r(t-1) - r(t-2)$$

$$\text{Answer: } h(t) = \frac{d}{dt}s(t) = u(t+2) - 2u(t) + 2u(t-1) - u(t-2)$$

Step Response



$$r(t) = u(t) * u(t)$$

$$h_2(t) = \int_{-\infty}^t h_1(\tau) d\tau$$

$$= \int_{-\infty}^t \delta(\tau) d\tau = u(t)$$

$$= \int_{-\infty}^t \delta(\tau) d\tau$$

$$= u(t) - u(t-1)$$

Discrete-time is Tricky (III)

- Keeping track of frequencies/periods for discrete-time
 $x[n] = A \cos(\omega_s n + \phi)$
 - $\omega_s, N_s = \frac{2\pi}{\omega_s}$: frequency and period of sinusoid if it were in CT
 - N_s : the number of samples before $x[n]$ repeats itself
- If N_s is rational $\leftrightarrow x[n]$ is periodic
 - What is the fundamental period N_0 of the DT signal?
 - We need to find some $N_0 \in \mathbb{Z}^+$ such that $N_0 \omega_s = 2\pi k$ for some integer k
 - The smallest such N_0 is the fundamental frequency, N_0

math solve convolution:

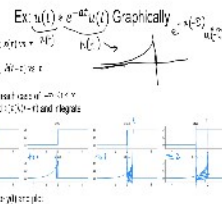
$$h(t) = \text{simple} \rightarrow h(t-\tau)$$

$$x(t) = \text{complex}$$

Graph:

$$x(t) \text{ stay}$$

$$h(t) \rightarrow \text{inverse and shift}$$



Complex number:

$z = a + jb$
 $j = \sqrt{-1}$
 $a = \text{Re}\{z\}$
 $b = \text{Im}\{z\}$

$|z| = r = \sqrt{a^2 + b^2}$
 $\angle z = \phi = \tan^{-1} \frac{b}{a}$

polar: $z = r e^{j\phi} = r(\cos(\phi) + j\sin(\phi))$

draw the circle:

Signal in Freq domain:

CT:

$x(t) = C e^{st} = C e^{\sigma t} e^{j\omega t}$
 $\sigma = 0$ period to aperiodic
 $\cos(\omega t) = \frac{1}{2} (e^{j\omega t} + e^{-j\omega t})$
 $\sin(\omega t) = \frac{1}{2j} (e^{j\omega t} - e^{-j\omega t})$

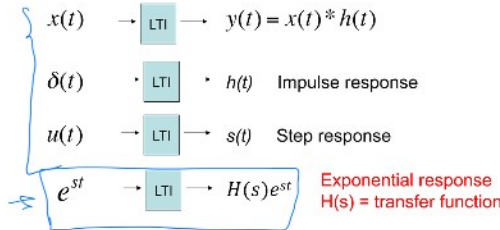
$s = \sigma + j\omega$

$\cos(\omega t) = \frac{1}{2} (e^{j\omega t} + e^{-j\omega t})$
 $\sin(\omega t) = \frac{1}{2j} (e^{j\omega t} - e^{-j\omega t})$

$e^{j\omega t} \rightarrow \text{LTI} \rightarrow A e^{j\omega t}$
 $e^{st} \rightarrow \text{LTI} \rightarrow H(s) e^{st}$

Checking Signal Properties: $x(t) = C e^{st}$

- When is a complex exponential bounded?
 $x(t) \leq B < \infty \forall t$ $|C e^{st}| \leq B$
 $\Rightarrow \frac{1}{T} \int_{-T/2}^{T/2} |x(t)|^2 dt < \infty$
- When is a complex exponential a power signal?
 $\frac{1}{T} \int_{-T/2}^{T/2} |x(t)|^2 dt < \infty$
- When does a complex exponential have finite energy?
 $\int_{-\infty}^{\infty} |x(t)|^2 dt < \infty$



s is a complex number $s = \sigma + j\omega$

Continuous-time Fourier Series (for Periodic Signals only)

Signal Synthesis: building signals with CEs

$x(t) = \sum_{k=-\infty}^{\infty} c_k e^{jk\omega_0 t}$ Sum and multiple CEs

Signal Analysis

$c_k = \frac{1}{T_0} \int_{T_0} x(t) e^{-jk\omega_0 t} dt$ take apart $x(t)$

The $< T_0 >$ means integrate over one period

For simple sinusoids, just use Euler's formula

Analysis: Square wave: $c_0 = \frac{1}{2}$ $c_k = \frac{1}{k\pi} \sin(\frac{k\pi}{2})$ $\omega_0 = \frac{2\pi}{T_0}$ if complex: $c_k = \frac{1}{k}$

Results So Far: Easy Cases for $x(t), T_0 = \frac{2\pi}{\omega_0}$

- Constant term $\rightarrow c_0$ $3e^{j\omega_0 t} \rightarrow c_k = 3$
- Lonely CE $(A e^{jk\omega_0 t}) \rightarrow c_k = A$
- Sinusoids (Euler)
 $\cos(k\omega_0 t) = \frac{1}{2} (e^{jk\omega_0 t} + e^{-jk\omega_0 t})$ $c_k = \frac{1}{2}$ $c_{-k} = \frac{1}{2}$
 $\sin(k\omega_0 t) = \frac{1}{2j} (e^{jk\omega_0 t} - e^{-jk\omega_0 t})$ $c_k = \frac{1}{2j}$ $c_{-k} = -\frac{1}{2j}$
- More generally
 $x(t) = A + \sum_{k=1}^{\infty} r_k \cos(k\omega_0 t + \theta_k)$
 $c_0 = A$ $c_k = \frac{r_k}{2} e^{j\theta_k}$ $c_{-k} = \frac{r_k}{2} e^{-j\theta_k}$ $k = 1, 2, \dots$

Fourier Series for Sinusoids: Analysis by Synthesis

$x(t) = \sum_{k=-\infty}^{\infty} c_k e^{jk\omega_0 t}$ $\omega_0 = \frac{2\pi}{T_0}$

Steps:

- Find fundamental frequency, ω_0
 $T_0 = \text{LCM}(T_1, T_2, \dots)$ $\omega_0 = \frac{2\pi}{T_0} = \text{GCD}(\omega_1, \omega_2, \dots)$
- Expand sinusoids into CEs via Euler's formula:
 $\cos(\omega t) = \frac{1}{2} (e^{j\omega t} + e^{-j\omega t})$
- Write CEs in terms of ω_0
- Read off Fourier coefficients, c_k (complex numbers)

Fourier Series Properties

$x(t) \leftrightarrow c_k$
 $\text{Linear ops: } y(t) = ax(t) + b \leftrightarrow c_k^y = ac_k^x + b \delta_{k,0}$ (Added constant only affects DC term)
 $\text{Time scale: } y(t) = x(at) \leftrightarrow c_k^y = c_{ka}^x$ (Scale a , scale c_k)
 $\text{Reverse: } y(t) = x(-t) \leftrightarrow c_k^y = c_{-k}^x$ (Reverse k)
 $\text{Shift in time: } y(t) = x(t - t_0) \leftrightarrow c_k^y = c_k^x e^{-jk\omega_0 t_0}$ (Add linear phase term $-jk\omega_0 t_0$)
 $ax_1(t) + bx_2(t) \leftrightarrow c_k^y = ac_k^{x_1} + bc_k^{x_2}$ (Adjusting k if ω_0 changes)

Ex 2: Impulse Train

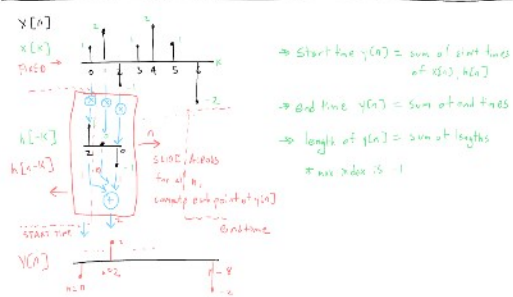
$x(t) = \sum_{n=-\infty}^{\infty} \delta(t - nT_0)$
 $c_k = \frac{1}{T_0}$

$\sum_{k=0}^n ar^k = a \left(\frac{1-r^{n+1}}{1-r} \right)$

$1 + r + r^2 + \dots + r^n = \text{SUM}(k=0 \dots n) r^k = (1 - r^{n+1}) / (1 - r)$
 $r + r^2 + r^3 + \dots + r^n = \text{SUM}(k=1 \dots n) r^k = (r - r^{n+1}) / (1 - r)$

$e^{-j\omega_0 t} = (-1)^k$
 $h[n] = (h_1 + h_2 - h_3) \cdot h_4$
 $h[n] = h_1 + h_2 - h_3$
 $h_1 = h_2$
 $= e^{-j\omega_0 n} u[n] = e^{-j\omega_0 n} u[n-1]$
 $= \sum_{k=0}^n e^{-j\omega_0 k} u[k] = u[n-k-1]$
 $= \sum_{k=0}^{n-1} e^{-j\omega_0 k} = u[n-1]$

$y(t) = x(t - t_0) = \sum_{n=-\infty}^{\infty} \delta(t - t_0 - nT_0)$
 $c_k^y = c_k^x e^{-jk\omega_0 t_0}$
 $c_k^y = \frac{1}{T_0} e^{-jk\omega_0 t_0}$
 $c_k^y = \frac{1}{T_0} e^{-jk\omega_0 t_0} \rightarrow \frac{1}{T_0} e^{-jk\omega_0 t_0}$



$\sum_{n=-\infty}^{\infty} c_n e^{-jn\omega_0 t} = \sum_{n=-\infty}^{\infty} c_n e^{-jn\omega_0 t}$
 $\sum_{n=-\infty}^{\infty} c_n e^{-jn\omega_0 t} = \sum_{n=-\infty}^{\infty} c_n e^{-jn\omega_0 t}$
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 $\sum_{n=-\infty}^{\infty} c_n e^{-jn\omega_0 t} = \sum_{n=-\infty}^{\infty} c_n e^{-jn\omega_0 t}$

For each system shown, state whether or not the given observed input/output pair $x(t), y(t)$ indicates the system is not time-invariant. Explain your reasoning either way.

- $\cos(\pi t) \xrightarrow{T_2} 3 \sin(\pi(t+1))$
 Answer: No.
 A linear gain and a constant delay are not inconsistent with an LTI system. This in/out pair does not break LTI behavior rules.
- $e^{-2t} \cos(\pi t) \xrightarrow{T_2} e^{-2t} \sin(\pi t) u(t)$
 Answer: Yes.
 The appearance of $u(t)$ in the output indicates that this is not LTI.
- $a e^{-2t} \cos(\pi t) \xrightarrow{T_2} a e^{-2t} \cos(\pi t)$
 Answer: No.
 $y(t) = a e^{-2t} \cos(\pi t) = b e^{-2t} a e^{-2t}$, so this is just a linear gain. The in/out pair does not indicate the system isn't LTI.
- $\cos(\frac{\pi}{2} n) + e^{-\frac{1}{2} |n|} \xrightarrow{T_2} \sin(\frac{\pi}{2} n)$
 Answer: Yes.
 No LTI system can selectively zero-out the $e^{-\frac{1}{2} |n|}$ term. This would be possible if there was no $|\cdot|$ operation in the exponent and if there was a phase lag introduced in the output signal.

CT vs. DT LCCDE

Assume we have an LTI system described a LCCDE:

$$\sum_{k=-\infty}^{\infty} a_k y[n-k] = \sum_{k=-\infty}^{\infty} b_k x[n-k]$$

Use the differentiation [or time shift] property of the FT:

$$\frac{d}{dt} x(t) \leftrightarrow j\omega X(\omega) \quad x(t-k) \leftrightarrow e^{-j\omega k} X(\omega)$$

Take FT of both sides of the LCCDE:

$$\sum_{k=-\infty}^{\infty} a_k (j\omega)^k Y(\omega) = \sum_{k=-\infty}^{\infty} b_k (j\omega)^k X(\omega)$$

Pull out common terms and get:

$$\frac{Y(\omega)}{X(\omega)} = \frac{\sum_{k=-\infty}^{\infty} b_k (j\omega)^k}{\sum_{k=-\infty}^{\infty} a_k (j\omega)^k}$$

Example: Freq derivative

- Find the FT of $v(t) = te^{-2t}u(t)$
- From the formulas, $x(t) = e^{-2t}u(t)$ has FT $X(\omega) = \frac{1}{2+j\omega}$
- Property: $tx(t) \leftrightarrow j \frac{d}{d\omega} X(\omega)$

Useful Result: Parseval's Relation

- Power of a CT periodic signal

$$P = \frac{1}{T} \int_{-T/2}^{T/2} |x(t)|^2 dt = \sum_{k=-\infty}^{\infty} |c_k|^2$$

- Power of a DT periodic signal

$$P = \frac{1}{N} \sum_{n=-N/2}^{N/2} |x[n]|^2 = \sum_{k=-\infty}^{\infty} |c_k|^2$$

Fourier Series - CT vs. DT

Signal Synthesis: building signals with CEs

$$x(t) = \sum_{k=-\infty}^{\infty} c_k e^{j\omega_k t} \quad x[n] = \sum_{k=-\infty}^{\infty} c_k e^{j\omega_k n}$$

Signal Analysis

$$c_k = \frac{1}{T} \int_{-T/2}^{T/2} x(t) e^{-j\omega_k t} dt \quad c_k = \frac{1}{N} \sum_{n=-N/2}^{N/2} x[n] e^{-j\omega_k n}$$

T_0 and N_0 indicate integrate/sum over length of one period

$$T_0 = \frac{2\pi}{\omega_0}, N_0 = \frac{2\pi}{\omega_0}$$

Recall: Discrete-time Periodicity is Tricky (II)

- There is a finite unique range of frequencies
 - For periodic $x[n] = e^{j\omega_0 n}$, $\omega_0 = \omega_0 + 2\pi m$
 - In general, frequency $\omega_k = \omega_0 + 2\pi k$
 - Often the range is given as $(-\pi, \pi]$, but it could be $[0, 2\pi)$
- Period of 1 gives a constant, so the smallest period is 2 ($\omega_0 = \pi$)
 - the highest frequency is π !
- For periodic $x[n] = e^{j\omega_0 n}$ with $N_0 = \frac{2\pi}{\omega_0}$
 - $(k + N_0)\omega_0 = (k + \frac{2\pi}{\omega_0})\omega_0 = k\omega_0 + 2\pi \Rightarrow k\omega_0$
 - There are N_0 unique harmonics!

Eigenfunctions & LTI Systems

$H(\omega) = H(s)|_{s=j\omega}$

$x(t) = \sum_{k=-\infty}^{\infty} c_k e^{j\omega_k t} \rightarrow y(t) = \sum_{k=-\infty}^{\infty} c_k H(\omega_k) e^{j\omega_k t}$

$x[n] = \sum_{k=-\infty}^{\infty} c_k e^{j\omega_k n} \rightarrow y[n] = \sum_{k=-\infty}^{\infty} c_k H(\omega_k) e^{j\omega_k n}$

$H(\omega)$ specifies frequency scaling

CT Fourier Transform

Synthesis (Inverse FT):

$$x(t) = \frac{1}{2\pi} \int_{-\infty}^{\infty} X(\omega) e^{j\omega t} d\omega$$

Analysis (FT):

$$X(\omega) = \int_{-\infty}^{\infty} x(t) e^{-j\omega t} dt$$

DT Fourier Transform

Synthesis (Inverse FT):

$$x[n] = \frac{1}{2\pi} \int_{-\pi}^{\pi} X(\omega) e^{j\omega n} d\omega$$

Analysis (FT):

$$X(\omega) = \sum_{n=-\infty}^{\infty} x[n] e^{-j\omega n}$$

DT frequency domain is less intuitive, so we'll start with CT FTs, except we need to know a little for the lab

Signals that have a FT

Condition 1: If the signal has finite energy, then it has a FT. $E\{x(t)\} < \infty$

Alternate Conditions for Existence of FT:

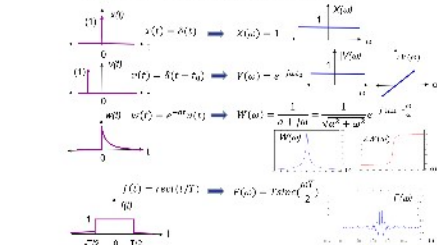
Dirichlet Conditions: If the following conditions hold:

- The signal $f(t)$ is absolutely integrable
 - Over any finite interval, $f(t)$ has a finite number of maxima and minima
 - In any finite interval, $f(t)$ has a finite number of discontinuities, each of which is finite
- Then the FT representation converges to $f(t)$ except at the points of discontinuity, where it converges to the average value on either side

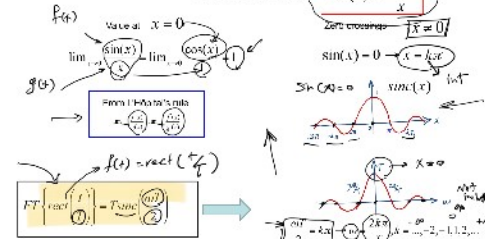
A Few FT Properties

| | $f(t)$ - time domain | $F(\omega)$ - Fourier Transform | Reason/Intuition |
|--------------|----------------------|-------------------------------------|---|
| Scaling | $af(t)$ | $aF(\omega)$ | linearity |
| Addition | $f(t) + g(t)$ | $F(\omega) + G(\omega)$ | linearity |
| Time shift | $f(t - t_0)$ | $e^{-j\omega t_0} F(\omega)$ | Shift in time \rightarrow linear phase term in frequency |
| Time scaling | $f(at)$ | $\frac{1}{ a } F(\frac{\omega}{a})$ | inverse relationship between time/frequency |
| Conjugation | $f^*(t)$ | $F^*(-\omega)$ | Frequency reversal, and $\omega \rightarrow -\omega$ |
| Convolution | $(f * g)(t)$ | $F(\omega) G(\omega)$ | Convolution in time \rightarrow multiplication in frequency |

Fourier Transforms so far...



How to Plot $\sin(x) = \frac{\sin(x)}{x}$



FT of General Periodic Signals

We know that

$$FT\{e^{j\omega_0 t}\} = 2\pi \delta(\omega - \omega_0)$$

Let $x(t)$ be a periodic signal with fundamental frequency ω_0 and FS representation

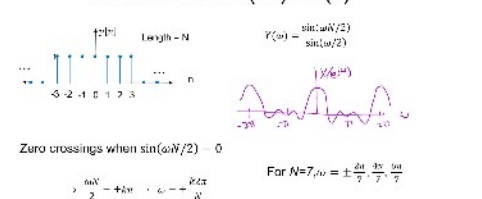
$$x(t) = \sum_{k=-\infty}^{\infty} a_k e^{j\omega_0 k t}$$

Because the FT is a linear operator, it follows that

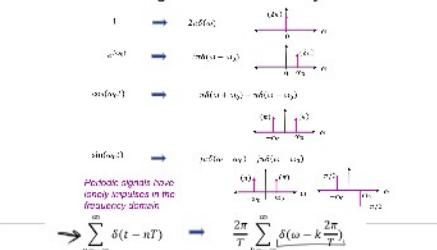
$$X(\omega) = FT\{x(t)\} = \sum_{k=-\infty}^{\infty} a_k FT\{e^{j\omega_0 k t}\} = \sum_{k=-\infty}^{\infty} a_k 2\pi \delta(\omega - k\omega_0)$$

The FT of a periodic signal is a sum of lonely impulses

How to Plot $\sin(ax)/\sin(x)$



Periodic signal FTs: Summary so far...



Digression:

Sinusoid in Freq \leftrightarrow Lonely Impulses in Time

$$x(t) = \frac{1}{2} \delta(t-5) + \frac{1}{2} \delta(t+5)$$

$$X(\omega) = \frac{1}{2} e^{-j5\omega} + \frac{1}{2} e^{j5\omega} = \cos(5\omega)$$

$$y(t) = \frac{1}{2} \delta(t+5) + \frac{1}{2} \delta(t-5)$$

$$Y(\omega) = \frac{1}{2} e^{j5\omega} + \frac{1}{2} e^{-j5\omega} = \cos(5\omega)$$

Find the Fourier Transform of $f(t) = e^{-a|t|}$ ($a > 0$)

Break the signal into two pieces: $f(t) = e^{-a(-t)}u(-t) + e^{-at}u(t)$

Use additivity & time reverse (scaling) property

Inverse Transform Example

$$X(\omega) = \frac{1}{2\pi} \int_{-\infty}^{\infty} x(t) e^{j\omega t} dt$$

$$x(t) = \frac{1}{2\pi} \int_{-\infty}^{\infty} X(\omega) e^{-j\omega t} d\omega$$

$$x(t) = \frac{1}{2\pi} \int_{-B}^B \frac{1}{\omega} e^{-j\omega t} d\omega = \frac{1}{2\pi} \left[\frac{1}{-j} \text{Si}(\omega t) \right]_{-B}^B = \frac{1}{\pi} \text{Si}(Bt)$$

Fourier Series vs. Fourier Transform

$$x(t) = \sum_{k=-\infty}^{\infty} c_k e^{jk\omega_0 t}$$

$$T_k = 2\pi / \omega_0$$

$$y(t) = \sum_{k=-\infty}^{\infty} c_k H(jk\omega_0) e^{jk\omega_0 t} = \sum_{k=-\infty}^{\infty} b_k e^{jk\omega_0 t}$$

$$X(j\omega) = \frac{1}{2\pi} \int_{-\infty}^{\infty} x(t) e^{-j\omega t} dt$$

$$Y(j\omega) = \frac{1}{2\pi} \int_{-\infty}^{\infty} y(t) e^{-j\omega t} dt$$

$$H(j\omega) = \frac{Y(j\omega)}{X(j\omega)}$$

$$H(j\omega)$$
 specifies frequency scaling

Simple Linear Phase System

$$y(t) = Mx(t - \tau)$$

System that scales the input by M and delays it by τ has the transfer function:

$$H(\omega) = Me^{-j\omega\tau}$$

and the impulse response:

$$h(t) = M\delta(t - \tau)$$

Describe these systems:

- 1) $H(\omega) = e^{j\omega\tau}$ scales by 1, delays by τ
- 2) $H(\omega) = e^{j\omega\tau}$ scales by 1, advances by τ
- 3) $H(\omega) = 1/3$ scales by 1/3, no delay
- 4) $H(\omega) = 5e^{-j\omega\tau}$ scales by 5, delays by τ

Constant magnitude for all frequencies \rightarrow all pass filter

Example 3

$$x(t) = 1 + \sin(20\pi t)$$

$$y(t) = H(j\omega)x(t)$$

$$H(j\omega) = \begin{cases} 1 & |\omega| < 3\pi \\ 0 & |\omega| > 3\pi \end{cases}$$

$$h(t) = 3\text{sinc}(3\pi t)$$

$$y_2(t) = H(j\omega)x(t) = 1 + \sin(20\pi t)$$

Why Linear Phase?

Linear phase means that all frequency components shift by the same amount, so they combine in the same way

$$\cos(2t) + 0.5\cos(4t) \rightarrow \cos(2(t - \tau)) + 0.5\cos(4(t - \tau))$$

$$\cos(2t) + 0.5\cos(4t) \rightarrow \cos(2(t - \tau)) + 0.5\cos(4(t - \tau))$$

Human hearing is not sensitive to minor phase differences, but large distortions sound bad

Compare CT & DT: Practical 1st-order LPF

Continuous Time: $y(t) + a \frac{dy(t)}{dt} = x(t)$

Discrete Time: $y[n] + ay[n-1] = (1+a)x[n]$

$$h(t) = \frac{1}{a} e^{-t/a} u(t)$$

$$h[n] = (1+a)(-a)^n u[n]$$

$$H(\omega) = \frac{1}{1 + j\omega a}$$

$$H(\omega) = \frac{1+a}{1 + ae^{-j\omega}}$$

$$a < 0$$

Aperiodic over ω

Periodic over ω

Let's Focus on Digital Filters to Start

General equation: $\sum_{k=0}^N a_k y[n-k] = \sum_{m=0}^M b_m x[n-m]$

Two main categories of digital filters:

- Finite impulse response (FIR): $N=0$
- Infinite impulse response (IIR): everything else

The $y[n-k]$ terms introduce feedback, which makes the impulse response infinite length

Ex1: Ideal Low Pass Filter

Consider the periodic signal: $x(t) = 10 + \cos(2t) + \cos(10t) + \cos(20t)$

Find the outputs when $x(t)$ is input to these two filters

$$H_1(\omega) = \begin{cases} 2e^{j\omega} & |\omega| < 4\pi \\ 0 & |\omega| > 4\pi \end{cases}$$

All frequencies $|\omega| > 4\pi$ are removed

$$y_2(t) = 20 + 2\cos(t + 1) + 2\cos(4t + 1) + 2\cos(10t + 1)$$

Ex 2: Different Ideal Filters

Consider another periodic signal: $x(t) = \text{square wave (50% on/off), frequency } \omega_0 = 2\pi, T_0 = 1, T_1 = 1/4$

$$a_0 = \frac{1}{2} \text{sinc}(\pi/2), k \neq 0, a_0 = \frac{1}{2}$$

Find the outputs when $x(t)$ is input to these filters

$$H_1(\omega) = \begin{cases} 1 & |\omega| < 3\pi \\ 0 & |\omega| > 3\pi \end{cases}$$

$$H_2(\omega) = \begin{cases} 2e^{j\omega} & 3\pi < |\omega| < 7\pi \\ 0 & \text{otherwise} \end{cases}$$

$$H_3(\omega) = \begin{cases} 1 & |\omega| > \pi \\ 0 & |\omega| < \pi \end{cases}$$

$$a_0 = \frac{1}{2}, a_1 = \frac{1}{2} \text{sinc}(\pi/2) = \frac{1}{2} \text{sinc}(\pi/2) = \frac{1}{2}$$

$$a_{-1} = \frac{1}{2} \text{sinc}(-\pi/2) = \frac{1}{2} \text{sinc}(\pi/2) = \frac{1}{2}$$

Digression

FT property: frequency shifting

CT: $e^{j\omega_0 t} x(t) \leftrightarrow X(\omega - \omega_0)$

DT: $e^{j\omega_0 n} x[n] \leftrightarrow X(\omega - \omega_0)$

For DT, note that: $h_2[n] = (-1)^n h_1[n] = e^{-j\pi n} h_1[n]$

$$H_2(\omega) = H_1(\omega - \pi)$$

so multiplying by $(-1)^n$ shifts the center from 0 to π , changing an LPF to an HPF or vice versa

Rewriting $(-a)^n u[n] = (-1)^n a^n u[n]$ explains the simple IIR results

Back to FT Properties

$f(t)$ -- Time Domain $F(\omega)$ -- Fourier Transform

Time shift: $f(t - t_0) \leftrightarrow F(\omega)e^{-j\omega t_0}$

Frequency shift: $f(t)e^{j\omega_0 t} \leftrightarrow F(\omega - \omega_0)$

Convolution: $f(t) * g(t) \leftrightarrow F(\omega)G(\omega)$

Multiplication: $f(t)g(t) \leftrightarrow \frac{1}{2\pi} F(\omega) * G(\omega)$

Frequency shift is a special case of multiplication with $g(t) = e^{j\omega_0 t}$

Aside: Duality & FT Properties

Duality: $f(t) \leftrightarrow F(\omega)$

Convolution: $f(t) * g(t) \leftrightarrow F(\omega)G(\omega)$

Multiplication: $f(t)g(t) \leftrightarrow \frac{1}{2\pi} F(\omega) * G(\omega)$

Time shift: $f(t - t_0) \leftrightarrow F(\omega)e^{-j\omega t_0}$

Frequency shift: $f(t)e^{j\omega_0 t} \leftrightarrow F(\omega - \omega_0)$

Time multiply: $tf(t) \leftrightarrow j \frac{d}{d\omega} F(\omega)$

Differentiation: $\frac{d}{dt} f(t) \leftrightarrow j\omega F(\omega)$

AM with Sinusoidal Carrier

also called AM-Double-Side-Band (DSB)

$$x(t)$$
 should be bandlimited, Low pass filter to ensure this

$$c(t) = \cos(\omega_c t)$$

$$C(\omega) = \frac{1}{2} [\delta(\omega - \omega_c) + \delta(\omega + \omega_c)]$$

$$y(t) = x(t) \cos(\omega_c t)$$

$$Y(\omega) = \frac{1}{2} [X(\omega - \omega_c) + X(\omega + \omega_c)]$$

Synchronous Demodulation

$$y(t) = x(t) \cos(\omega_c t)$$

$$Y(\omega) = \frac{1}{2} [X(\omega + \omega_c) + X(\omega - \omega_c)]$$

$$C(\omega) = \pi [\delta(\omega + \omega_c) + \delta(\omega - \omega_c)]$$

$$W(\omega) = Y(\omega) C(\omega)$$

$$W(\omega) = \frac{1}{4} [X(\omega + \omega_c) + X(\omega - \omega_c)]$$

Recover $x(t)$ from $y(t)$ via LPF $H(\omega)$ with gain 2 and cutoff frequency ω_{LPF}

Sampling in time: $x_s(t) = x(t)p(t)$

Multiply by an impulse train in time: $p(t) = \sum_{n=-\infty}^{\infty} \delta(t - nT_s)$

Convolves with an impulse train in frequency: $P(\omega) = \frac{2\pi}{T_s} \sum_{k=-\infty}^{\infty} \delta(\omega - k\omega_s)$

$$X_s(\omega) = \frac{1}{T_s} X(\omega) * P(\omega) = \frac{1}{T_s} \sum_{k=-\infty}^{\infty} X(\omega - k\omega_s)$$

Nyquist Sampling Theorem

For a band-limited signal $x(t)$ with $X(\omega) = 0$ for $|\omega| > \omega_B$, in order to uniquely reconstruct $x(t)$ after sampling, the sampling frequency ω_s must be greater than $2\omega_B$.

$$2\omega_B < \omega_s = \frac{2\pi}{T_s}$$

Also called Nyquist-Shannon Sampling Theorem

Terminology: $2\omega_B$ is the **Nyquist rate**, it is a property of a signal

$\frac{\omega_s}{2}$ is the **Nyquist frequency**, it is a property of the sampler

When $\omega_B > \frac{\omega_s}{2}$ then we say we have **aliasing**, or we are **undersampling**

Reconstructing Original Signal

$$X_s(\omega) = \frac{1}{T_s} \sum_{k=-\infty}^{\infty} X(\omega - k\omega_s)$$

Ideal low pass filter with gain T_s and cutoff frequency ω_c

$$H(\omega) = T_s \text{ for } |\omega| \leq \omega_c, 0 \text{ for } |\omega| > \omega_c$$

$$X(\omega) = \frac{1}{T_s} X_s(\omega) H(\omega)$$

Sampling in Frequency Domain

$$\omega_B = X(\omega)$$
 bandwidth

$$\omega_s = \frac{2\pi}{T_s}$$
 sampling frequency

$$X_s(\omega) = \frac{1}{T_s} \sum_{k=-\infty}^{\infty} X(\omega - k\omega_s)$$

$$X(\omega) = \frac{1}{T_s} X_s(\omega) H(\omega)$$