#### Signal Definitions: EE228 - Common Signals

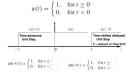
Unit Step	$u(t) = \begin{cases} 0, t < 0 \\ 1, t > 0 \end{cases}, \text{ undefined for } t = 0.$ Heaviside unit step: $u(0) = 0.5$	
Unit Ramp	$r(t) = \begin{cases} 0, t < 0 \\ t, t >= 0 \end{cases}$	
Signum (sign of argument)	$sgn(t) = \begin{cases} +1, & t > 0 \\ 0, & t = 0 \\ -1, & t < 0 \end{cases}$	
Rectangle or pulse	$rect(t) = \begin{cases} 1,  t  < 0.5 \\ 0, & else \end{cases}$ , width = 1	1 ns ns
Rectangle or pulse	$rect\left(\frac{t-\beta}{\alpha}\right) = \begin{cases} 1, \beta - (\alpha/2) < t < \beta + (\alpha/2) \\ 0, & else \end{cases}$ width = $\alpha$	- <del> </del>
Triangle	$tri(t) = \begin{cases} 1 -  t , &  t  < 1 \\ 0, & else \end{cases}, \text{ width} = 2$	

Sinc	$\begin{aligned} & \operatorname{sinc}(t) = \sin(\pi t) / \pi t \\ & \operatorname{sinc}(t) = 0, \ t = \pm 1, \pm 2, \pm 3, \dots \text{(zero crossings)} \\ & \operatorname{sinc}(0) = 1 \end{aligned}$	, A.
Impulse	$\delta(t) = \begin{cases} 0, & t \neq 0 \\ \infty, & t = 0 \end{cases} \text{ Area} = 1$	
Weighted Impulse	$A \delta(t) = \begin{cases} 0, & t \neq 0 \\ \infty, & t = 0 \end{cases} Area = A$	(1)

$$\begin{aligned} e^{j\theta} &= \cos \theta + j \sin \theta \\ \cos \theta &= \mathcal{R} \mathcal{E} \left\{ e^{j\theta} \right\} = \frac{e^{j\theta} + e^{-j\theta}}{2} \\ \sin \theta &= \mathcal{I} \mathcal{M} \left\{ e^{j\theta} \right\} = \frac{e^{j\theta} - e^{-j\theta}}{2j} \end{aligned}$$

#### UNIT - STEP:

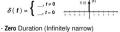
#### **Unit-Step Function**



#### **Unit-Ramp Function**

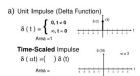
$$r(t) = \begin{cases} t, & \text{for } t \ge 0 \\ 0, & \text{for } t < 0 \end{cases}$$

#### Unit Impulse (Delta Function)



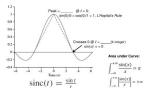
- Infinite Amplitude (height) @ t = 0
- -<u>Finite</u> Area =  $Area = \int_{-\infty}^{\infty} \delta(t)dt =$
- -Infinite Energy  $= \int_{-\infty}^{\infty} |\delta^2(t)| dt =$

# **OPERATIONS ON IMPULSES**

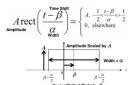


# IMPULSE PRODUCT/Siefting PROP:

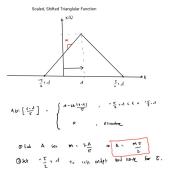
#### **Unit-Sinc Function**



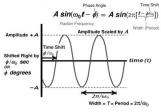
#### Scaled, Shifted Rectangular **Pulse Function**



How to find is setting the  $\beta - \alpha/2$  to end point of

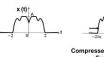


# Scaled, Shifted Sinusoids



# Operations on signals:

# **Operations on Signals:** Time Scaled: $S\{x(t)\}=x(\alpha t)$



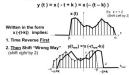
Original Signal

 $y(t) = x(\alpha t)$ 

Compressed in time for  $|\alpha|$  \_ 1 Ex: y(t) = x()Expanded in time for  $|\alpha| = 1$ Ex: y(t) = x( ) = x( )

$$-2 < t < 2 \text{ => } -2/\alpha < t < 2/\alpha$$

# Operations on Signals: Time Shift With Reversal



# Periodicity, Energy and Power:

# **Signal Measures**

# Signal Areas & Integrals:

Signal Area = 
$$\int_{-\infty}^{\infty} x(t)dt$$
  
Absolute Area =  $\int_{-\infty}^{\infty} |x(t)|dt$   
Absolutely Integrable:  $\int_{-\infty}^{\infty} |x(t)|dt$  (finite area)

$$Time\ Average = \frac{1}{-}\int x(t)dt$$

Signal Energy = Integral of Instantaneous Power over all time Signal Energy =  $\int_{-\infty}^{\infty} P_i(t)dt =$ 

Avg. Signal Power = Time average of Instant. Power over all time Total Signal Energy / Time Duration (i.e. energy per unit of time) Avg. Signal Power =  $\lim_{t\to\infty} \frac{1}{t} \int_{-t/t}^{t/2} |x^2(t)| dt$  If nonperiodic signal

 $= \frac{\text{Energy in 1 Period}}{\text{Time in 1 Period}} \qquad \text{If periodic signa} \\ \text{with period "T"}$ 

# Energy vs. Power Signal:

Signal is a power signal if: (example - periodic fns)

If the signal is not time-limited (infinite time duration.

Has Infinite total signal energy Has finite non zero Power (Energy/Period)

Signal is an Energy signal if: (example - non periodic fns, half cycle sinusoid)

Has finite signal Energy

Has zero signal power(  $E/\infty=0$  )

Power of a Sinusoid:  $P_{avg} = \frac{A^2}{2}$ 

Power of periodic signal made up of sinusoids (  $X_0 is const$  ):

$$P_{avg} = P_0 + \sum_{k=1}^{\infty} P_k = X_0^2 + \sum_{k=1}^{\infty} \frac{A_k^2}{2}$$

Energy of half cycle sinusoid:  $E = \frac{A^2b}{2}$ 

#### Signal Measures - Periodic Signals Power Signals: Finite Power / Energy = oc

Average Value 
$$\overline{x} = \frac{1}{T} \int_{t_0}^{t_0+T} x(t) dt$$
 (DC value)

Effective Value 
$$x_{eff} = \int_{t_b}^{t_b+T} dt = x_{ema}$$
 (rms value)

Effective Value = DC value having same average power as 
$$x(t)$$
 = (square) ROOT of the MEAN (average) of  $x(t)$  SQUARED =  $\sqrt{P}$ 

Signal Power = 
$$\frac{1}{T} \int_{T} |x^{2}(t)| dt$$

The "fundamental frequency" f<sub>0</sub> of the sum signal will be the " " of the individual frequencies.

#### Even and Odd Parts of f(t)

Any function f(t) can be represented by the sum of an even symmetric function  $f_c(t)$  and an odd symmetric function  $f_0(t)$ ,

$$f(t) = f_e(t) + f_0(t)$$

letermined by:

determined by: 
$$\begin{aligned} & \text{Even Part of } f(t): & f_e(t) = \frac{f(t) + f(-t)}{2} \\ & \text{Odd Part of } f(t): & f_o(t) = \frac{f(t) - f(-t)}{2} \end{aligned}$$

## System Classifications:

#### **Checking Differential Equation For** Linearity:

System is <u>Nonlinear</u> if the Diff. Equation describing it contains:

# Checking Differential Equation For Time-Invariance:

System is Time <u>Varying</u> if the Diff. Equ. describing its behavior contains:

$$y(t) = 2tx'(t) + ...$$

• Time-scaled inputs or output terms 
$$y(t) = b_0 x(2t) + ...$$

### Static vs. Dynamic

#### Static Systems (instantaneous)

- Determining the present output of the system  $y(t_0)$  at any particular time " $t_0$ " requires \_\_\_\_\_ of the input  $x(t_0)$
- All x(t) and y(t) terms have the \_\_\_\_ arguments ( ).
- Effects of previous inputs do not linger to affect the output later -
- No derivatives or integrals in system equation.  $y(t) = S\{x(t)\} = Cx(t) + A$

# Other Classes / Properties

#### Causal (non-anticipating)

- At any time "t<sub>0</sub>" (the "present"), the output of the system y(t<sub>0</sub>) is determined completely by values of the input x(t) that arrived at the <u>same</u> or <u>previous</u> times ... i.e. x(t) for t \_\_\_ t<sub>0</sub>
- Causal systems do not use \_\_\_\_\_ values of the input  $x(t+\tau)$  to determine the present output y(t)

#### Causal Systems (non-anticipating)

#### Determining Causality from the system differential equation:

- If the equation is in terms of y(t), y'(t), y''(t),... there are no \_\_\_\_\_terms in the equation
- 2. There are no \_\_\_\_

# SOLVING DIFFERENTIAL EQNS:

#### Alternate Solution Linear Diff. Equ.

Total Response y<sub>T</sub>(t) has 2 Alternate Components



γ<sub>zi</sub>(t)



 $\begin{array}{ll} - & \text{Includes the part of the Natural Response due to IC's:} \\ & \text{Total Natural Response: } \gamma_{N_i}(t) = \gamma_{N_1}(t) + \gamma_{N_2}(t) \\ - & \text{If system is truly "relaxed" (all IC's=0),} & \gamma_{Zi}(t) = 0 \end{array}$ 

#### **Solution Method** Method of Undetermined Coefficients - General Case

Solution Steps:

- 1. Convert the Differential Equation to a Single Input Case
- 2. Find ZIR, the zero-input response,  $y_{ZIR}(t) = y_{N_1}(t)$
- 3. Find the  $\boldsymbol{forced\ response},\,y_F(t)$  of the Single Input Case
- 4. Find ZS1, the zero-state response of the Single Input Case
- 5. Find ZSR, the zero-state response of the General Case 6. Find General Case total response,  $y_T(t) = y_{ZSR}(t) + y_{ZIR}(t)$

 $y_{ZS1}(t) = y_F(t) + y_{N_2}(t)$ 

#### Solution Method

Method of Undeter

#### Solution Steps:

1. Find ZIR, the **zero-input response**,  $y_{ZI}(t) = y_{N1}(t)$ [Solve with \_ \_; using given I.C.'s.]

2. Find the forced response, y<sub>F</sub>(t)

[  $y_F(t)$  is a solution of the Single Input Case Diff Equ. ]

2. Find ZSR, the zero-state response,  $y_{7S}(t) = y_F(t) + y_{NS}(t)$ [Solve with actual x(t); using \_\_

#### Solution Method - General Case

When terms such as Ax(t) + B dx(t)/dt appear on the right side of a generalized differential equation, a similar approach is used.

$$y^{(n)}(t) + a_{n-1}y^{(n-1)}(t) + \ldots + a_1y'(t) + a_0y(t) = A \, x(t) + B \, dx(t)/dt$$

We first find the <u>zero state response</u> of the <u>single input case</u> y<sub>zs1</sub>(t), the response due to <u>just the x(1)</u> forcing function, for the modified (single input case) differential equation:

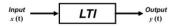
$$y^{(n)}(t) + a_{n-1}y^{(n-1)}(t) + ... + a_1y'(t) + a_0y(t) = x(t)$$

2. Then linearity and superposition are used to find the <u>overall ZSR</u> due to all the input terms [ x(t) and its derivatives ] on the right side of the equation, *i.e.*:

If input  $x(t) \rightarrow yields S.I.C. ZS response <math>y_{xst}(t)$  from the system, en input of [A x(t) + B dx(t)/dt]  $\Rightarrow$  yields Gen. Zero State Response [Ay<sub>zs1</sub>(t) + B $dy_{zs1}(t)/dt$ ]

#### STABILITY FROM LOOKING AT DE:

# BIBO Stability Requirements



- . Every root of the characteristic equation must have a
- Constrains natural response to be bounded

#### **BIBO Stability Requirements**



2. The degree of the highest derivative of x(t) in the differential equation (RHS) \_the *order* of the must differential equation (the highest derivative of y(t) on Left Hand Side)

 $y'(t) + a_0y(t) = b_0 x(t) + b_1 x'(t) + b_2 x''(t)$ 

# Impulse response and step response:

### Methods for Determining h(t)

- Find the Impulse Response h(t) as the derivative of the Step Response s(t)
- First find the Step Response by solving the relaxed system differential equation [ICs=0] for: x(t) = u(t)
- 2. Solve the relaxed system differential equation for:  $x(t) = \delta(t)$

#### Method for Finding Impulse Response h(t)

- 1. Set all SIC Initial Conditions to zero (relaxed) y(0)=0, y'(0)=0,...
- 2. Set the highest order initial condition to 1  $y^{n-1}(0)=1$

Ex: n=2: y(0)=0, y'(0)=1

- $\begin{array}{ll} \textbf{3. Solve Single-Input Diff Equ. for } x(t) = 0 \\ & \text{Solve for the natural response only } (y_r(t) = 0) \\ & \text{Use the above IC's to find unknown coef's} \end{array}$
- 4. Determine generalized h(t) same way as generalizing the ZSR.

# Other System Properties from h(t)

# Stability

For a stable system: h(t) must be "

# Causality

• For a system to be causal:

h(t) = , for t < 0

Convolution:

# **Properties of Convolution**

 $y(t) = \int_{-\infty}^{\infty} x(\lambda)h(t-\lambda)d\lambda$ 

Time Shift: x(t) \* h(t) = y(t)

 $x(t-\tau)*h(t) = y(t-\tau)$ 

 $x(t-\tau) * h(t-\alpha) =$ 

x(t) \* h(t) = y(t)Derivative:

x'(t) \* h(t) =

# **Properties of Convolution**

 $y(t) = \int x(\lambda)h(t-\lambda)d\lambda$ 

Convolution with Impulse

 $\delta(t) * h(t) =$ 

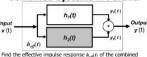
Convolution with a Shifted Impulse  $\delta(t-\tau)*h(t) =$  $\delta(t-\tau)h(t) =$ 

## **Combined Systems in Cascade**



Find the effective impulse response  $h_{eff}(t)$  of the combined CASCADED Filters:

#### **Combined Systems in Parallel**



#### **Evaluation of Convolution Integral** Two Methods:

#### 1. Analytical Evaluation

PARALLEL Filter

- Break up the integral based on similar pairings of u(t)'s in x() and h() functions

#### 2. Graphical Evaluation (Convolution By Ranges)

- Break up problem into different regions based on where signal transitions and overlaps occur
   In each region, find a unique solution using the appropriate analytical functions for x() and h()

# **Convolution by Ranges:**

First determine both x(t) and h(t) 1.)

Range End Point.

2.) Then find pairwise sum

REPT(y(t)) = SORT(set(PWS(REPT(h(t)), REPT(x(t))))

Then write y(t) as a piecewise where y\_1,.., y\_k is the value

between the endpoints.

# **GRAPHICAL CONVOLUTION**

For each range of "t" determined above:

- Draw x(t-λ) relative to h(λ) {or h(t-λ) relative to x(λ)} on a single set of λ axes.
- Observe the endpoints of the regions where the two functions overlap. The beginning and endpoint of the overlapping regions (in terms of  $\lambda$ ) determine the lower and upper limits of integration for  $\lambda$  in the convolution integral in this range this range.
- Solve the integral and evaluate over the upper and lower limits.
- REPEAT Steps above for each range of t

CHECKing answers for solution:

$$\begin{split} Lenth_y &= Lenth_x + Lenth_h \\ T_{start,y} &= T_{start,x} + T_{start,h} \\ T_{end,y} &= T_{end,x} + T_{end,h} \end{split}$$

#### Correlation:

#### Correlation

 $r_{xh}(t) = x(t) **h(t) = \int x(t)h(t-\tau)dt = x(t) *h(-t)$ 

- · Measure of similarity between x, h
- Like Convolution, But Shift h(t) without Flipping it
- Not Commutative:  $r_{xh}(t) = r_{hx}(-t)$
- "Cross-Correlation" if  $x(t) \neq h(t)$
- "Auto-Correlation" if x(t) = h(t)

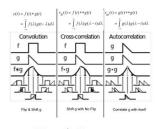
#### Cross-Correlation

• For non-periodic functions:

$$r_{12}(\tau) = \int_{-\infty}^{+\infty} x_1(t)x_2(t-\tau)dt$$
,  $-\infty < \tau < +\infty$ 

For **periodic** functions, with period T, the correlation function is given by:

$$r_{12}(\tau) = \frac{1}{T} \int_{\frac{T}{2}}^{\frac{T}{2}} v_1(t) v_2(t-\tau) dt$$



# **Summary** - Correlation

· Correlation is computed similarly to Convolution, but without the signal flip

$$r_{sh}(\tau) = x(t) **h(t) = \int_{-\infty}^{\infty} x(t)h(t-\tau)dt = x(t) *h(-t)$$

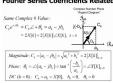
- Cross-Correlation  $r_{vb}(\tau)$  reveals the degree of similarity between two different signals, and how the similarity changes with time alignment of the signals
- **Auto-Correlation**  $r_{xx}(\tau)$  (correlating a signal with itself) reveals the **randomness or periodicity** of a signal

$$r_{xx}(\tau) = x(t) * *x(t) = \int_{-\infty}^{\infty} x(t)x(t-\tau)dt = x(t) *x(-t)$$

Filtering (convolution) a noisy signal with a time-reversed clean version of a signal implements correlation detection (also called "matched" filtering") by cross correlating the noisy and clean signals to find if the signal is detected and at what time position.

#### FS: Coefficients Relationship:

#### **Fourier Series Coefficients Related**



#### **Exponential Fourier Series** Coefficient Symmetry

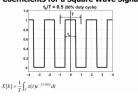
$$\begin{split} & f \circ A + b \cdot E \\ & X[k] = \frac{1}{2} C_1 e^{-j h} = \frac{1}{2} C \angle + \theta, \qquad X[-k] = \frac{1}{2} C_1 e^{-j h} = \frac{1}{2} C \angle - \theta_1 \\ & [X[k]] = |X[-k]| = \frac{1}{2} C_1 \qquad \text{[DD] is "Even Symmetric"} \\ & \angle X[k] = -\angle X[-k] = \theta, \qquad \text{ZMy Is "Coolly gate Symmetry"} \\ & X^+[k] = X[-k] \qquad \text{XN} \text{ Is "Conjugate Symmetry"} \\ & \mathcal{H}(X[k]) = \mathcal{H}(X[k]) = \mathcal{H}(X[k]) + m[X[-k]) \end{split}$$

# FS: Finding Fourier Coefficients:

#### **Finding Exponential Fourier Series** Coefficients: a<sub>k</sub>, b<sub>k</sub>



#### Find the Exponential Fourier Series Coefficients for a Square Wave Signal



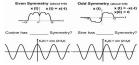
# **Exponential Fourier Coefficients**



# **Exponential Fourier Series**

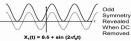
# FS: Shortcut (Signal Symmetrys)

# Signal Symmetries



#### Odd Symmetry



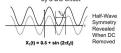


#### **Half-Wave Symmetry**



#### **Half-Wave Symmetry**

Half-Wave Symmetry can also be **hidden** by a DC Offset



# **Signal Symmetries**

$$a_k = \frac{2}{T} \int_0^T x_p(t) \cos(2\pi k f_0 t) dt \qquad b_k = \frac{2}{T} \int_0^T x_p(t) \sin(2\pi k f_0 t) dt$$

$$a_k = \frac{1}{T} \int_0^{\infty} x_p(t) \cos(2\pi k f_0 t) dt \qquad b_k = 0$$

• 
$$x_p(t)$$
 has Odd Symmetry:

$$b_k = \frac{1}{T} \int_0^{\infty} x_p(t) \sin(2\pi k f_0 t) dt \qquad a_k = 0$$

#### **Signal Symmetries**

#### • x<sub>p</sub>(t) has Half-Wave Symmetry:

$$a_k = \frac{4}{T} \int_0^{T/2} x_p(t) \cos(2\pi k f_0 t) dt$$
  $k = odd$   
 $b_k = \frac{4}{T} \int_0^{T/2} x_p(t) \sin(2\pi k f_0 t) dt$   $k = odd$ 

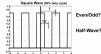
# **Signal Symmetries**

# • x<sub>p</sub>(t) has Even and Half-Wave Symmetry: $a_k = \frac{8}{T} \int_0^{T/4} x_p(t) \cos(2\pi k f_0 t) dt$ k = odd

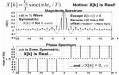
 $b_k = \frac{8}{T} \int_0^{T/4} x_p(t) \sin(2\pi k f_0 t) dt \qquad k = odd$ te: You can not use the half-wave symmetry integrals above if the half-wave symmetry is "hidden" (i.e. if there is a DC offset).

-The  $k_{\rm size} = 0$  result for the Fourier Coefficients DOES apply for  $k \ge 2$  if there is





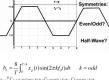
#### Symmetry Seen in Spectrum



#### Other Symmetries in Spectrum



#### Example: Trapezoidal Signal



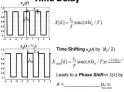


#### FS: properties

# Properties of Fourier Series

$x(t) = x^*(t)$	← X <sub>c</sub> = X <sup>s</sup> Conjugate
x(t) = x(-t)	$\iff X_e = X_{-e}$ Real
x(t) = -x(-t)	$\iff X_c = -X_{-c}$ imaginan
x(t) = -x(t + T/2)	$\iff X_e = 0 \text{ for } n \text{ even}$
$\underline{x(t)+x(-t)}$	$\iff \mathcal{RE}\{X_n\}$
$\frac{x(t)-x(-t)}{2}$	$\iff jIM\{X_n\}$
ax(t) + by(t)	$\iff aX_n + bY_n$
x(-t)	$\iff X_n^*$
$x(t - t_0)$	$\iff X_n e^{-j2\pi n t_0/T}$
	x(t) = x(-t) x(t) = -x(-t) x(t) = -x(t + T/2) x(t) = -x(t + T/2) $\frac{x(t) + x(-t)}{2}$ $\frac{x(t) + x(-t)}{2}$ x(-t)

# **Time Delay**



#### FS: POWER

#### **Power in Harmonic Signals**

So we can compute the power in a harmonic signal from the **time-domain** waveform itself: Signal Power =  $\frac{1}{T} \int_{0}^{\infty} |x_p|^2(t) dt =$ 

OR from its frequency spectrum & Fourier Series coef And we should get the same answer!!

| X[k] |2 - "Power Spectral Deposits

#### Power in Harmonic Signals

$$\begin{split} P &= \begin{bmatrix} c_i^2 + 0.5\sum_{k=1}^{\infty}c_k^2 \end{bmatrix} & \text{Polar} \\ &= \begin{bmatrix} a_k^2 + 0.5\sum_{k=1}^{\infty}(a_k^2 + b_k^2) \end{bmatrix} & \text{Trigonometric} \\ &= \sum_{k=1}^{\infty}|X[k]|^2 = |X[0]|^2 + 2\sum_{k=1}^{\infty}|X[k]|^2 \\ &= \text{Exponential (2-sided)} \end{split}$$

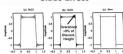
#### **Power in Few Harmonics**

The Power in a subset of the harmonic frequencies in a signal can also be found using the Exponential Fourier Coefficients of those harmonics:



# FS: GIBBS EFFECT

# **Gibbs Effect**



# FS: System Function

# $x(t) = \sum_{k} c_k \cos(2\pi k f_0 t + \theta_k)$

# What is H(f)?



ransfer) Function: of output signal y(t) to input signal x(t)cy-dependent (changes with f) scale factor

H(f) = |H(f)| ∠H(f)

Magnitude Scale Factor (Gain) Phase Shift (Time Shift) y (t)= $\sum_{k}$ |H(kf<sub>0</sub>)| c<sub>k</sub> cos( $2\pi kf_0 t + \theta_k + \angle H(kf_0)$ )

# **Fourier Series Concepts**

- Any (almost) periodic signal can be represented by a combination of harmonically-related (f=kf<sub>0</sub>, k integer) sinusoids (sin / cos / e<sup>j2,nt</sup>)

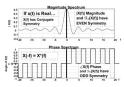
  - Requirements: x(t) absolutely integrable, finite # of finite discontinuities, finite # max/mins
- Reconstruction from Fourier Series will be imperfect at discontinuities (overshoot)

#### FT: Integral Definiti

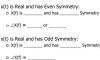
#### What IS the Fourier Transform?



#### **Symmetries in Fourier Spectrum**



# Signal Symmetries



# **Existence of Fourier Transforms**

- A non-periodic signal x(t) can be represented by a unique Fourier Transform IF:
- 1. x(t) is "absolutely integrable"
- $\int_{-\infty}^{\infty} |x(t)| dt < \infty$  Integral over ALL time to
- x(t) has a finite number of maxima & minima.
   x(t) has a finite number of finite discontinuities.

# **Existence of Fourier Transforms**

However, some signals that don't meet these strict requirements still have valid Fourier Transforms. (Ex: x(t)=u(t) is not abs. integrable  $\int \!\!\! -\infty$ )

- . F.T. of an "absolutely integrable" signal
- F.T. of a "not absolutely integrable" signal
  \_\_\_\_, if it does not grow exponentially (e<sup>+at</sup>).
- F.T. of a signal that grows exponentially or

#### FT AND FS Relationship:

# Methods of Computing Fourier Transforms and Inverse Transforms

 Converting known exponential Fourier Series coefficients of a periodic signal version x<sub>s</sub>(t)  $X_p[k] \Rightarrow X(f) = T X_p[k]_{kf_0 \to f}$ Fourier Series of A <u>Periodic Signal</u> A <u>Non-periodic Signal</u> X<sub>P</sub>(t) made up of 1 Period of made up of 1 Period of the periodic signal X<sub>P</sub>(t)

# FT: Properties

## **Duality: Similarity Theorem**

Fourier Transform:  $X(f) = \int_{-\pi}^{+\infty} x(t)e^{-j2\pi \beta} dt$ Inverse Fourier Transform  $x(t) = \int_{-\pi}^{+\pi} X(f)e^{\tau/2\pi\beta}df$ 

If FT{ x(t)} = X(f)Then FT{ X(t)} = x(-f)

For functions with Even Symmetry in x(t):  $x(+t) = x(-t) \qquad X(+f) = X(-f)$ So FT{ X(t)} = x(f)

# • Duality ( Summarry, . • Let $\chi(t) \Leftrightarrow \chi(\omega)$ **₼**--**₼** -A--

#### Time / Frequency Scaling

me Scaling 
$$x(\alpha t) \xleftarrow{FT} \frac{1}{\ell m FT} X(f/\alpha)$$
 $\xrightarrow{FT} X(\beta f)$ 
Frequency Scaling  $X(\beta f)$ 
Frequency Scaling  $X(\beta f)$ 

## Time / Frequency Scaling

An extension of this concept is the Time/Frequency Limit Theorem: A signal can <u>not</u> have <u>BOTH</u> a finite tir duration ("time-limited") <u>AND</u> a finite len frequency spectrum ("band-limited"). A signal can be either "time-limited" or "band-limited",...or neither,...\_ sime Infinite freq. spect Finite in time

#### FT: ENERGY

# **Energy in Non-Periodic Signals**

We can compute the energy in an aperiodic signal from the time-domain waveform itself: Total Energy =  $\int_{-\infty}^{\infty} x^2(t)dt =$ \_ OR from its Fourier Transform frequency spectrum: And we should get the same answer!!
"Parseval's Theorem"

ode: You can not use the half-wave symmetry integrals above if the half-wave symmetry is "hidden" (i.e. if there is a DC offset).

-The  $k_{max} = 0$  result for the Fourier Coefficients DOES apply for  $k \ge 2$  if t

# FT: input a sinusoid

#### **System Response to Periodic Input Signals**



nse to the Fourier Series Term of Input Signal  $y(t) = |H(f_0)|c_1\cos(2\pi f_0t + \theta_1 + \angle H(f_0))$ 

#### LT: Definition

#### Limitations of the Fourier Transform

 Fourier Transform is \_\_\_\_\_
 a number of interesting signals
 Fourier Transform does not convert 2. Fourier System Analysis: Y(f)=X(f)H(f) only allows us to determine the \_\_\_\_\_ 

#### Region of Convergence (ROC)

For the Laplace Transform to exist, the Transform *integral* must converge, limiting the values of  $s\ (or\ \sigma)$  that we can use to only those for which:

$$\int_{-\infty}^{+\infty} |x(t)e^{-\sigma t}u(t)| dt < \infty \quad \text{(absolutely integrable)}$$

- Choosing σ too small could make it too weak to force convergence
- Choosing negative values for σ would take a converging function and make it go to +∞ (f)

How do we choose and specify the allowed range of  $\sigma$ ??

#### **Methods of Computing Laplace Transforms**

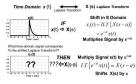
- 1. Using the integral Transform definitions  $X(s) = \int_{0^{-}}^{\infty} x(t)e^{-st}dt$
- 2. Converting known Fourier Transforms for causal signals:  $X(f)\Rightarrow X(s)$  a) Drop any  $S(\pm f_0)s$  in the Fourier Transform b) Change  $j2\pi j\Rightarrow s$   $f/j0\Rightarrow s$ )
- 3. Using known Laplace Transform pairs (from Tables) and modifications from L.T. <u>Properties</u>

## Region of Convergence (ROC)

The ROC of a unilateral Laplace transform (causal signal) is a right side half-plane.

#### LT: Properties

# S Domain Shift: The S Domain "Dual" of the Time-Shift Property



#### **Convolution Property**

The most important property of the LAPLACE TRANSFORM for System Analysis:

$$x(t)*h(t) \xrightarrow{LT}$$

The CONVOLUTION of two causal Functions in the Time Domain...

(Product of Complex Functions or perty holds for BOTH the Unitateral AND Bilateral Laplace Transforms

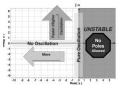
Can use the 1-sided LTs Only IF both x(t) and h(t) are

### LT: Transfer Function - poles and zeros

$$\begin{split} H(s) &\text{ is a "rational function"} \\ &\text{-ratio of functions of the same variable} \\ \\ H(s) &= \frac{b_s + b_s^{3!} + b_s^{3!} + \dots + b_y s^{N!}}{s^N + a_{N:1} s^{N!} + \dots + a_j s^{3!} + a_s + a_o} = \frac{Num(s)}{Denom(s)} \end{split}$$

It is a "rational function" if the order of the denominator polynomial (N) is equal to c greater than the order of the numerator polynomial (M)  $N \ge M$ " "rational function" if the order of the denominator polynomial (M) is greater than the order of the numerator polynomial (M) is greater than the order of the numerator polynomial (M)

# Poles in the S Plane



#### Stability Requirements



# Stability and the ROC of H(s)

- The Region of Convergence for a System Function H(s) can not include any \_\_\_\_\_\_.
- \* The ROC of a right-sided time function (incl. Causal) takes the form of  $\mathcal{R}e\{s\} = \sigma > \sigma_0$
- . ROC is to the right of some of
- The ROC of a Causal System [ causal h(t) ] is  $\mathcal{R}e(s) = \sigma > \mathcal{R}e(p_t)$  where  $p_t$  is the ROC is to the right of the rightmost pole For a Stable, Causal System, the ROC must include • The ROC of a Causal System [ causal h(r) ] is

# LLT: Partial Fraction Expansion

# Partial Fraction Expansion:

Special Cases:

Repeated Roots (Poles)
 The coefficients of these pole terms are determined by multiplication of H(s) only by the highest power of the repeated pole term, and evaluating at the pole value.

$$\begin{split} H(s)(s+p_1)^2 &= \frac{N(s)(s+p_1)^2}{(s+p_1)(s+p_2)^2} = \frac{K_1(s+p_1)^2}{(s+p_1)(s+p_2)^2} + \frac{d_1(s+p_2)^2}{(s+p_1)^2} + \frac{d_1(s+p_2)^2}{(s+p_2)^2} + \frac{d_1(s+p_2)^2}{(s+p_2)^2} \\ H(s)(s+p_2)^2 \Big|_{s=p_1} &= \frac{N(s)}{(s+p_1)^2} \Big|_{s=p_2} + \frac{K_1(s,p_2)^2}{(s+p_1)^2} \Big|_{s=p_2} + d_1 + d_2 \left[\frac{s^2}{s^2}\right]_{s=p_2} \end{split}$$

$$H(s)(s+p_1)^2 = \frac{N(s)(s+p_1)^2}{(s+p_1)^2} + \frac{d_1(s+p_2)^2}{(s+p_1)^2} \Big|_{s=p_2} + d_1 + d_2 \left[\frac{s^2}{s^2}\right]_{s=p_2} + \frac{d_1(s+p_2)^2}{(s+p_1)^2} \Big|_{s=p_2} + d_2 + d_2 \left[\frac{s^2}{s^2}\right]_{s=p_2} + d_3 + d_3 +$$

#### Partial Fraction Expansion:

Special Cases:

Repeated Roots (Poles)

The coefficients for decreasing powers of the repeated pole term are determined from derivatives of the previous function, again evaluated at the pole value.

$$\begin{split} \frac{d}{dt} \left[ H(s)(s+p_{2})^{2} \right]_{\text{new}_{p}} &= \frac{d}{dt} \left[ \frac{N(s)}{(s+p_{1})} \right]_{\text{new}_{p}} \\ &= \frac{d}{dt} \left[ \frac{K_{1}(s+p_{1})}{(s+p_{1})} \right]_{\text{new}_{p}} + \frac{d}{dt} A_{1} + \frac{d}{dt} \left[ A_{1}(s+p_{1}) \right]_{\text{new}_{p}} \\ &= \left[ \frac{2K_{1}(s-p_{1})}{(s+p_{1})} - \frac{K_{1}(s-p_{1})}{(s+p_{1})} \right]_{\text{new}_{p}} + 0 + A_{1} \\ \frac{d}{dt} \left[ H(s)(s+p_{1})^{2} \right]_{\text{new}_{p}} - A_{1} \end{split}$$

#### Partial Fraction Expansion:

2. Repeated Roots (Poles)

$$A_n = \frac{1}{n!} \frac{d^n}{ds^n} \left[ H(s) \left( s + p_2 \right)^k \right]_{s = -p}$$

where: k-n = the power (exponent) of the pole in the partial fraction term that goes with  $A_s$ 

#### **Inverse Laplace Transform**

Non-"Strictly Proper" Rational Functions:

#### **Inverse Laplace Transform**

Functions with Time Shifts (e-as):

# LT: Initial Value and Final Value theorems

#### **Initial Value Theorem**

# **Method:**

- 1) Convert X(s) to Strictly Proper  $X_p(s)$  (if needed)
- 2) Multiply  $X_p(s)$  by s.
- 3) Divide both numerator & denominator polynomials by the highest power of "s" in the denominator.
- 4) Evaluate  $x(0^+) = \lim_{n \to \infty} |sX_p(s)|$

# **Final Value Theorem**

Predicts the "final" value of the signal:  $x(t)|_{t\to\infty}$ 

from its Laplace Transform X(s):

$$x(\infty) =$$

<u>IF:</u> All Poles w/  $\sigma$ 's<0 (any  $\omega$ )  $x(t)=Ae^{-\sigma t}cos(\omega t + \theta)+...$ when common factors (s-a) in numerator & denominator cancelled (in case zero cancels bad pole). [OK: 1 pole @ s=0 ( $\sigma$ =0,  $\omega$ =0)] x(t)=Bu(t)+...

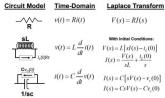
In final value theorem don't worry about if X(s) Is strictly proper or not

#### **Method for Finding Output of** Non-Relaxed Systems (ZIR, Total Resp)

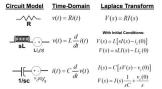
- **1.**  $x(t) \Rightarrow Laplace\ Transform \Rightarrow X(s)$
- 2. Laplace Transform Differential Equation Using Deriv. Property WITH Initial Conditions  $Y(s)P(s)-y(0^-)Q(s)-y'(0^-)R(s)-..=X(s)\bullet Z(s)$
- 3. Isolate Y(s) and group terms on RHS  $r(s) = X(s) \frac{Z(s)}{P(s)} + \frac{Y(0^*)Q(s) + y^*(0^*)R(s) + \dots}{P(s)}$ 4. Insert Y(s) expression and y(0), Y(0) values 5.  $Y_{Total}(t) = InverseLaplaceTr\{Y(s)\}$

#### LT: Circuit Analysis

#### **Non-Relaxed Laplace Circuit Models Current Source (Norton) IC Models**

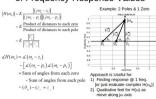


#### **Non-Relaxed Laplace Circuit Models** Voltage Source (Thevenin) IC Models



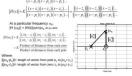
#### LT: Graphical Rep of Freq response

#### **Graphical Interpretation** of Frequency Response Values



#### Graphical Interpretation of Frequency Response Values

 $H(s) = K \frac{(s-z_1)(s-z_2)(s-z_3)...}{(s-p_1)(s-p_2)(s-p_3)...}$ 



# BODE PLOTS:

Pole or zero at s=0 (jw)^n:

Where n > 0 for zero Where n < 0 for pole

$$\begin{split} H(\omega) &= (j\omega)^n \\ H(\omega)_{dB} &= 20nlog|j\omega| \\ \angle H(\omega) &= (\pi/2)*n \end{split}$$

General Real zero or pole at s=-w\_c:

Where n > 0 for zero Where n < 0 for pole  $H(\omega) = (j\omega/\omega_c + 1)^n$  $H(\omega)_{dB} = 20*n*log|j\omega/\omega_c|$ 

$$\angle H_3(\omega) = n \angle \left(1 + j\frac{\omega}{\omega_c}\right) = \begin{cases} 0, & \omega < \omega_c \\ n\pi/4, & \omega = \omega_c \\ n\pi/2, & \omega > \omega \end{cases}$$

$$H(\omega) = (j\omega/\omega_c - 1)^n$$

Above shows a positive zero the only thing that as negative zero, difference is the phase is the phase is neg.

# Complex Poles at $s=-\sigma_1\pm\omega_1$

