



DSP - Fisika UI

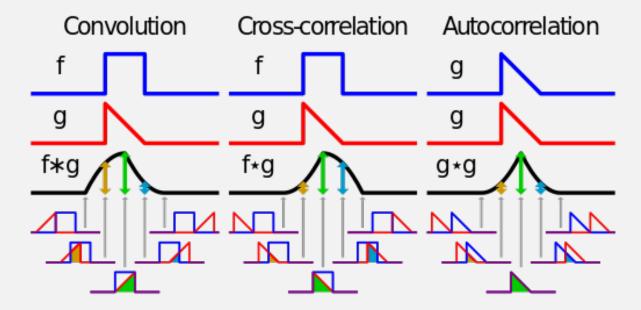


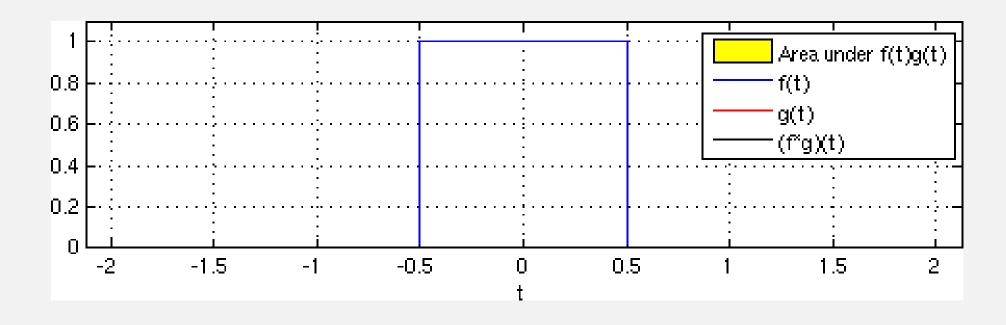
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- Convolution is a mathematical operation on two functions (f and g)
- Produces a third function, that is typically viewed as a modified version of one of the original functions, giving the integral of the pointwise multiplication of the two functions as a function of the amount that one of the original functions is translated.





• The impulse response of an LTI system is given by h[n]

$$y[n] = LTI[x[n]] = \sum_{k=-\infty}^{\infty} x[k]h[n-k]$$

• The mathematical operation is called a *linear convolution sum* and is denoted by

$$y[n] \triangleq x[n] * h[n]$$

• If the sequences are mathematical functions (of finite or infinite duration), then we can analytically evaluate for all n to obtain a functional form of y[n].

- The "n" dependency of y[n] deserves some care: for each value of "n" the convolution sum must be computed **separately** over all values of a dummy variable "m". So, for each "n"
 - Rename the independent variable as m. You now have x[m] and h[m]. Flip h[m] over the origin. This is h[-m]
 - 2. Shift **h[-m]** as far left as possible to a point "**n**", where the two signals barely touch. This is **h[n-m]**
 - 3. Multiply the two signals and sum over all values of **m**. This is the convolution sum for the specific "**n**" picked above.
 - 4. Shift / move h[-m] to the right by one sample, and obtain a new h[n-m]. Multiply and sum over all m.
 - 5. Repeat 2~4 until **h[n-m]** no longer overlaps with **x[m]**, i.e., shifted out of the **x[m]** zone.

- The "n" dependency of y[n] deserves some care: for each value of "n" the convolution sum must be computed **separately** over all values of a dummy variable "m". So, for each "n"
 - 1. Rename the independent variable as **m**. You now have **x**[**m**] and **h**[**m**]. Flip **h**[**m**] over the origin. This is **h**[-**m**]
 - 2. Shift **h[-m]** as far left as possible to a point "**n**", where the two signals barely touch. This is **h[n-m]**
 - 3. Multiply the two signals and sum over all values of **m**. This is the convolution sum for the specific "**n**" picked above.
 - 4. Shift / move h[-m] to the right by one sample, and obtain a new h[n-m]. Multiply and sum over all m.
 - 5. Repeat $2\sim4$ until h[n-m] no longer overlaps with x[m], i.e., shifted out of the x[m] zone.

$$y[n] = x[n] * h[n] = \sum_{m=-\infty}^{\infty} x[m] \cdot h[n-m] = \sum_{m=-\infty}^{\infty} h[m] \cdot x[n-m]$$

USEFUL EXPRESSIONS

 The following expressions are often useful in calculating convolutions of analytical discrete signals

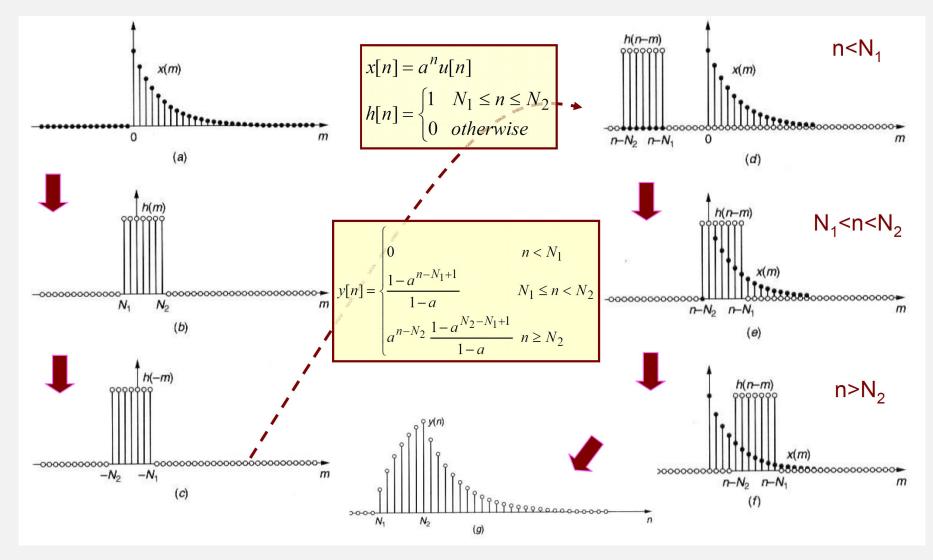
$$\sum_{n=0}^{\infty} a^n = \frac{1}{1-a}, \quad |a| < 1$$

$$\sum_{n=k}^{\infty} a^n = \frac{a^k}{1-a}, \quad |a| < 1$$

$$\sum_{n=m}^{N} a^n = \frac{a^m - a^{N+1}}{1-a}, \quad a \neq 1$$

$$\sum_{n=0}^{N-1} a^n = \begin{cases} \frac{1-a^N}{1-a}, & |a| \neq 1 \\ N, & a = 1 \end{cases}$$

CONVOLUTION EXAMPLE



• Let the rectangular pulse x(n) = u(n) - u(n - 10) be an input to an LTI system with impulse response

$$h(n) = (0.9)^n u(n)$$

• Determine the output y(n).

From the convolution equation

$$y(n) = x(n) * h(n)$$

$$y(n) = \sum_{k=0}^{9} (1)(0.9)^{(n-k)} u(n-k) = (0.9)^n \sum_{k=0}^{9} (0.9)^{-k} u(n-k)$$

- The sum in equation is almost a geometric series sum except that the term u(n-k) takes different values depending on n and k.
- There are three possible conditions under which u(n-k) can be evaluated.

- Case 1
 - n < 0
 - Then $u(n k) = 0, 0 \le k \le 9$

$$y(n) = 0$$

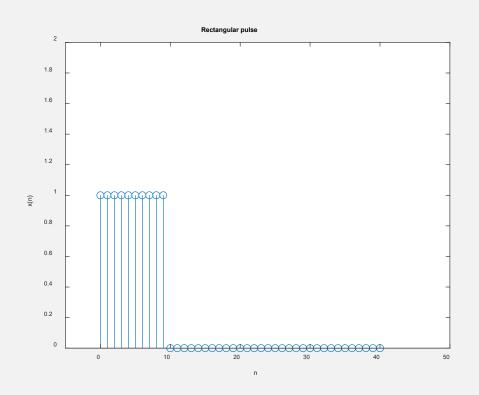
- Case 2
 - The nonzero values of x(n) and h(n) do not overlap.
 - $0 \le n < 9$: Then $u(n k) = 1, 0 \le k \le n$.

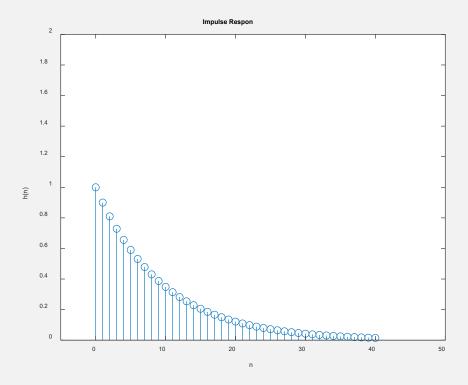
$$y(n) = (0.9)^{n} \sum_{k=0}^{n} (0.9)^{-k} = (0.9)^{n} \sum_{k=0}^{n} \left[(0.9)^{-1} \right]^{k}$$
$$= (0.9)^{n} \frac{1 - (0.9)^{-(n+1)}}{1 - (0.9)^{-1}} = 10 \left[1 - (0.9)^{(n+1)} \right]$$

- Case 3
 - The impulse response h(n) partially overlaps the input x(n).
 - $n \ge 9$: Then $u(n k) = 1, 0 \le k \le 9$

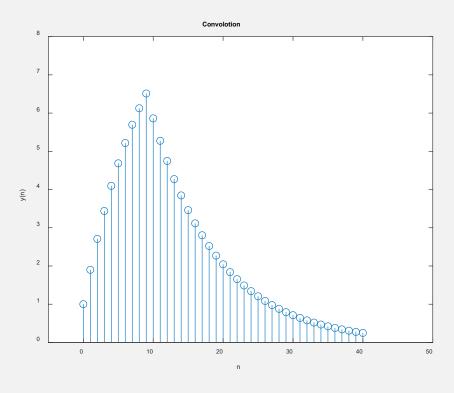
$$y(n) = (0.9)^{n} \sum_{k=0}^{9} (0.9)^{-k}$$
$$= (0.9)^{n} \frac{1 - (0.9)^{-10}}{1 - (0.9)^{-1}} = 10(0.9)^{n-9} \left[1 - (0.9)^{10}\right]$$

• The input sequence and the impulse response





• The output sequence

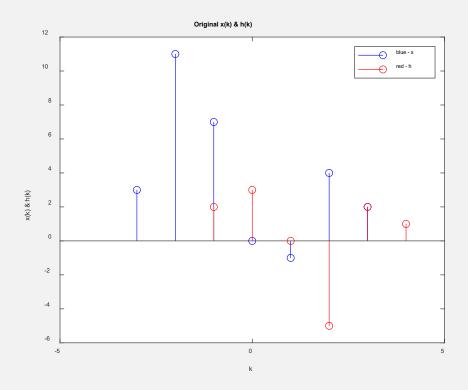


Given the following two sequences

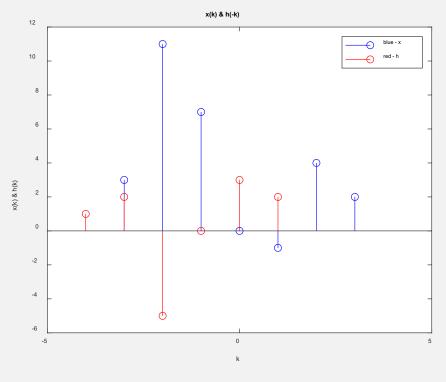
$$x(n) = [3,11,7,0,-1,4,2], -3 \le n \le 3$$

$$h(n) = [2, 3, 0, -5, 2, 1], -1 \le n \le 4$$

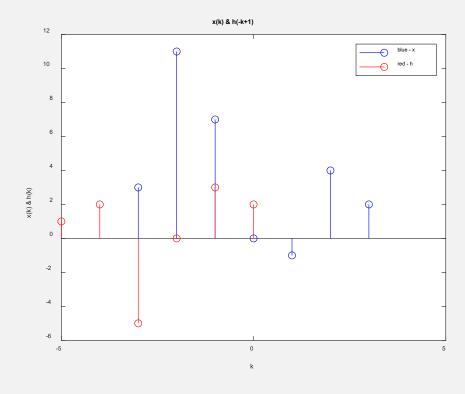
• Determine the convolution y(n) = x(n) * h(n).



The original sequences

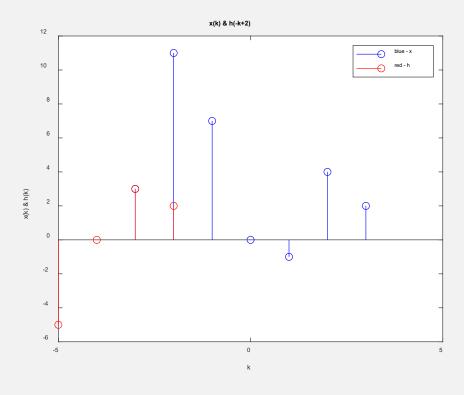


x(k) and h(-k), the folded version of h(k)



x(k) and h(-1-k), the folded-and-shifted by -1 version of h(k).

$$\sum_{k} x(k)h(-1-k) = 3 \times (-5) + 11 \times 0 + 7 \times 3 + 0 \times 2 = 6 = y(-1)$$



x(k) and h(2 - k), the folded-and-shifted-by-2 version of h(k)

$$\sum_{k} x(k)h(2-k) = 11 \times 1 + 7 \times 2 + 0 \times (-5) + (-1) \times 0 + 4 \times 3 + 2 \times 2 = 41 = y(2)$$

- Similar graphical calculations can be done for other remaining values of y(n).
- The beginning point (first nonzero sample) of y(n) is given by

$$n = -3 + (-1) = -4$$

The end point (the last nonzero sample) is given by

$$n = 3 + 4 = 7$$

The complete output is given by

$$y(n) = \{6,31,47,6,-51,-5,41,18,-22,-3,8,2\}$$

MATLAB CONVOLUTION

- If arbitrary sequences are of infinite duration, then MATLAB cannot be used directly to compute the convolution.
- MATLAB does provide a built-in function called conv that computes the convolution between two finite-duration sequences. The conv function assumes that the two sequences begin at n = 0 and is invoked by

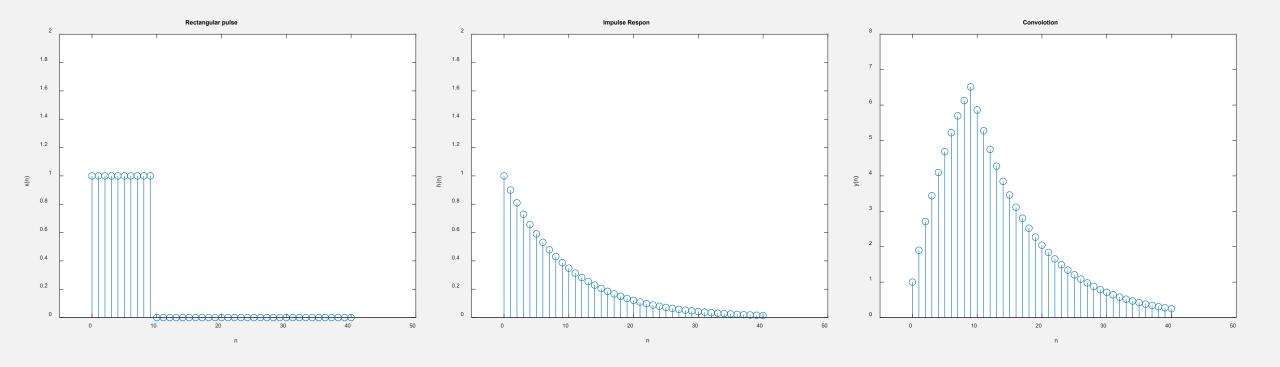
```
• y = conv(x, h);
```

• Let the rectangular pulse x(n) = u(n) - u(n - 10) be an input to an LTI system with impulse response

$$h(n) = (0.9)^n u(n)$$

• Determine the output y(n).

```
[x1,n] = stepseq(0,0,40);
[x2,n] = stepseq(10,0,40);
x = x1-x2;
h = (0.9.^n).*stepseq(0,0,40);
y = conv(x,h);
subplot(311); stem(n,x);
subplot(312); stem(n,h);
subplot(313); stem(n,y(1:length(n)));
```



MATLAB CONVOLUTION

 A simple modification of the conv function, which performs the convolution of arbitrary support sequences

```
function [y,ny] = conv m(x,nx,h,nh)
% Modified convolution routine for signal processing
% [y,ny] = conv m(x,nx,h,nh)
% [y,ny] = convolution result
% [x,nx] = first signal
% [h,nh] = second signal
nyb = nx(1) + nh(1);
nye = nx(length(x)) + nh(length(h));
ny = [nyb:nye]; y = conv(x,h);
```

Given the following two sequences

$$x(n) = [3,11,7,0,-1,4,2], -3 \le n \le 3$$

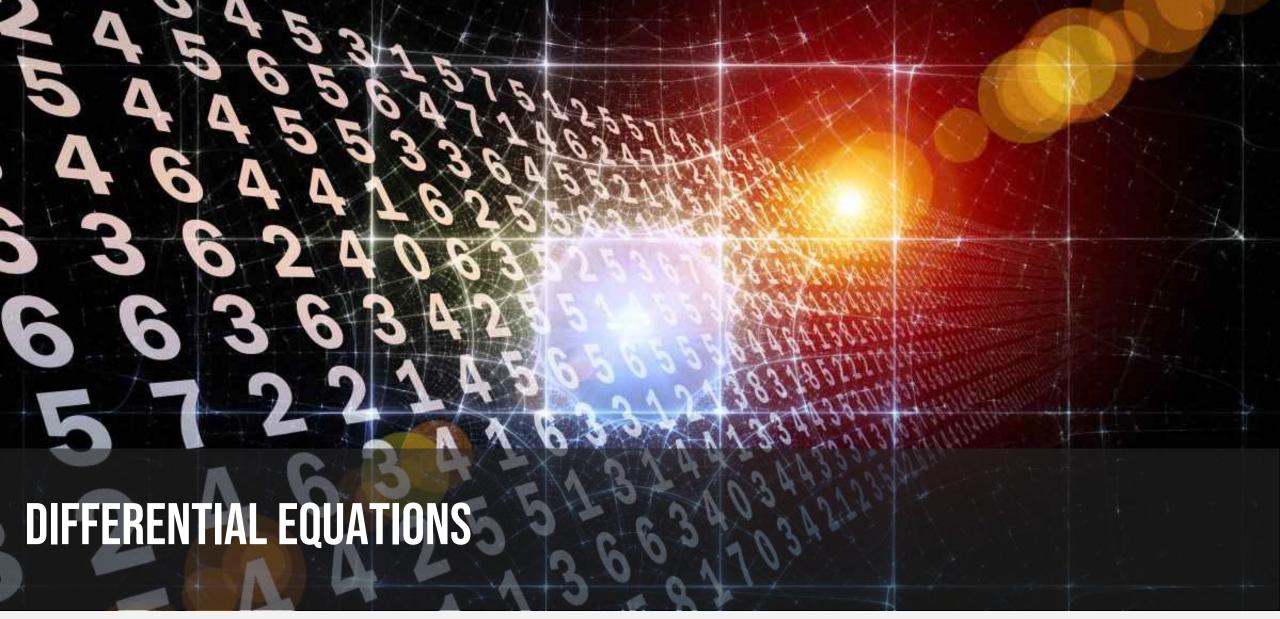
$$h(n) = [2, 3, 0, -5, 2, 1], -1 \le n \le 4$$

• Determine the convolution y(n) = x(n) * h(n).

```
>> x = [3, 11, 7, 0, -1, 4, 2]; nx = [-3:3];
>> h = [2, 3, 0, -5, 2, 1]; nh = [-1:4];
```

```
>> [y,ny] = conv_m(x,nx,h,nh)
y =
6 31 47 6 -51 -5 41 18 -22 -3 8 2
ny =
-4 -3 -2 -1 0 1 2 3 4 5 6 7
```

$$y(n) = \{6,31,47,6,-51,-5,41,18,-22,-3,8,2\}$$

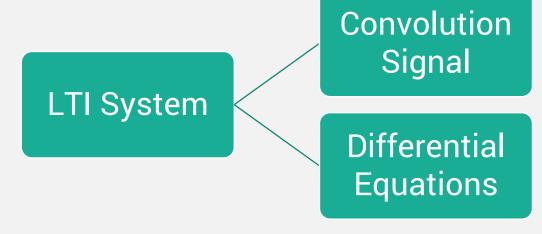


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LTI SYSTEM



$$y[n] \triangleq x[n] * h[n]$$

$$\sum_{k=0}^{N} a_{k} y [n-k] = \sum_{k=0}^{M} b_{k} x [n-k]$$

a recursive approach for computing the current output, given the input values and previously computed output values

LINEAR CONSTANT-COEFFICIENT DIFFERENCE EQUATIONS

 In general, an Nth-order linear constant coefficient difference equation has the form

$$\sum_{k=0}^{N} a_{k} y [n-k] = \sum_{k=0}^{M} b_{k} x [n-k]$$

- The output is not uniquely specified for a given input
 - The initial conditions are required
 - · Linearity, time invariance, and causality depend on the initial conditions
 - If initial conditions are assumed to be zero, system is linear, time invariant, and causal

LINEAR CONSTANT-COEFFICIENT DIFFERENCE EQUATIONS

- Example
 - Moving Average

$$y[n] = x[n] + x[n-1] + x[n-2] + x[n-3]$$

Difference Equation Representation

$$\sum_{k=0}^{0} a_k y [n-k] = \sum_{k=0}^{3} b_k x [n-k] \text{ where } a_k = b_k = 1$$

RECURSIVE SOLUTION

• In the discrete-time case, there are an alternative to find the Differential Equations

$$y[n] = \frac{1}{a_o} \left(\sum_{k=0}^{M} b_x x[n-k] - \sum_{k=1}^{N} a_k y[n-k] \right)$$

- This is often how digital filters are implemented on a computer or a digital signal processor board
- The response of a differential equation
 - The differential equation is discretized at a given sampling rate to obtain a difference equation
 - The response of the difference equation is computed recursively

GENERAL SOLUTION

 A general solution can be expressed as the sum of a homogeneous response (natural response), and a particular solution (forced response) of the system:

$$y[n] = y_h[n] + y_p[n]$$

• The concept of *initial rest* of the LTI causal system described by the difference equation here means that x[n] = 0 implies y[n] = 0

GENERAL SOLUTION

• The homogeneous part of the solution

$$y_h[n] = \sum_{k=1}^{N} c_k z_k^n$$

 z_k , k = 1, ..., N are N roots (also called *natural frequencies*) of the characteristic equation

$$\sum_{k=1}^{N} a_k z^k = 0$$

important in determining the stability of systems

GENERAL SOLUTION

• The particular part of the solution is determined from the right-hand side

x[n]	$y_p[n]$
A	K
$A.M^n$	K.M ⁿ
$A.n^{M}$	$K_0 n^M + K_1 n^{M-1} + K_2 n^{M-2} + \dots + K_M$
$A^n.n^M$	$A^{n} \left(K_{0} n^{M} + K_{1} n^{M-1} + K_{2} n^{M-2} + \ldots + K_{M} \right)$

Consider the difference equation

$$y[n] - 3y[n-1] - 4y[n-2] = 0$$

• Find system response y[n] for y[-1] = 5 and y[-2] = 0!

Solution: Assumption → homogen solution

$$y_h[n] = z^n$$

Substitution to original equation

$$z^{n} - 3z^{n-1} - 4z^{n-2} = 0$$

$$z^{n-2} (z^{2} - 3z - 4) = 0$$

$$(z+1)(z-4) = 0$$

$$z = -1 \text{ and } z = 4$$

Solution

$$y[n] = y_h[n] + y_p[n]$$

 $y[n] = a_1(-1)^n + a_2(4)^n$

Finding a₁ and a₂ to fulfil the solution

Finding a₁ and a₂ to fulfil the solution

$$y[n] - 3y[n-1] - 4y[n-2] = 0$$

 $y[n] = 3y[n-1] + 4y[n-2]$

$$y[0] = 3y[-1] + 4y[-2]$$

$$y[1] = 3y[0] + 4y[-1]$$

$$= 3y\{3y[-1] + 4y[-2]\} + 4y[-1]$$

$$= 13y[-1] + 12y[-2]$$

$$y_h[n] = a_1(-1)^n + a_2(4)^n$$

$$y_h[0] = a_1 + a_2$$

$$y_h[1] = -a_1 + 4a_2$$

Input the initial value

$$y[0] = 15$$

 $y[1] = 65$

$$15 = a_1 + a_2 a_1 = -1$$

$$65 = -a_1 + 4a_2 a_2 = 16$$

Solution

$$y[n] = -1(-1)^n + 16(4)^n \quad n \ge 0$$

Consider the difference equation

$$y[n] - \frac{5}{6}y[n-1] + \frac{1}{6}y[n-2] = 3x[n] - 2x[n-1]$$

- Find the recursive form of the difference equation
- Assuming initial rest and that the input is an impulse,

•
$$y[-2] = y[-1] = 0$$

$$y[n] = \frac{5}{6}y[n-1] - \frac{1}{6}y[n-2] + 3x[n] - 2x[n-1]$$

the recursion can be started

$$y[0] = \frac{5}{6}y[-1] - \frac{1}{6}y[-2] + 3x[0] - 2x[-1]$$

$$= \frac{5}{6}(0) - \frac{1}{6}(0) + 3(1) - 2(0) = 3$$

$$y[1] = \frac{5}{6}y[0] - \frac{1}{6}y[-1] + 3x[1] - 2x[0]$$

$$= \frac{5}{6}(3) - \frac{1}{6}(0) + 3(0) - 2(1) = \frac{1}{2}$$

$$y[2] = \frac{5}{6}y[1] - \frac{1}{6}y[0] + 3x[2] - 2x[1]$$
$$= \frac{5}{6}(\frac{1}{2}) - \frac{1}{6}(3) + 3(0) - 2(0) = -\frac{1}{12}$$

MATLAB IMPLEMANTATION

- A function called filter is available to solve difference equations numerically, given the input and the difference equation coefficients.
- In its simplest form this function is invoked by

$$y = filter(b,a,x)$$

- where
 - b = [b0, b1, ..., bM]; a = [a0, a1, ..., aN] are the coefficient arrays

$$\sum_{k=0}^{N} a_k y [n-k] = \sum_{k=0}^{M} b_k x [n-k]$$

MATLAB IMPLEMANTATION

- To compute and plot impulse response, MATLAB provides the function impz.
- When invoked by

$$h = impz(b,a,n)$$

• Computes samples of the impulse response of the filter at the sample indices given in n with numerator coefficients in b and denominator coefficients in a.

Given the following difference equation

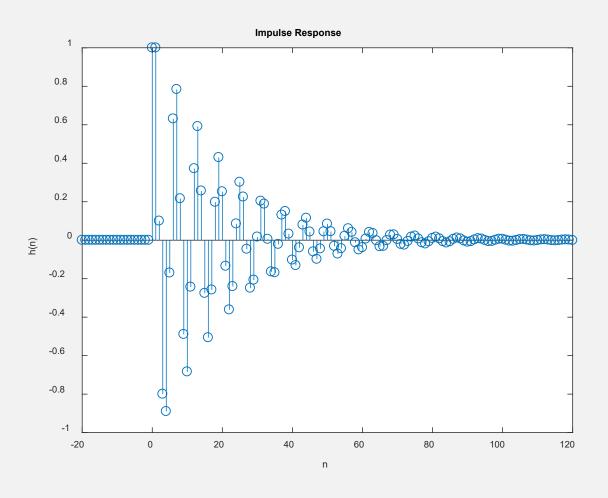
$$y(n) - y(n-1) + 0.9y(n-2) = x(n);$$

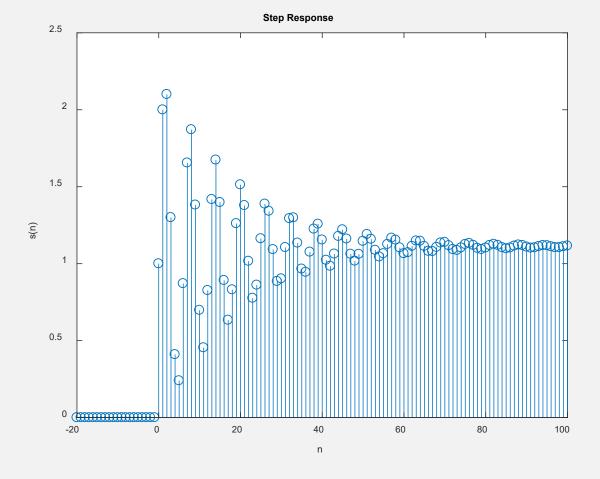
- Calculate and plot the impulse response h(n) at $n = -20, \dots, 100$.
- Calculate and plot the unit step response s(n) at $n = -20, \ldots, 100$.
- Is the system specified by h(n) stable?

- From the given difference equation the coefficient arrays are
 - b = [1]; a=[1, -1, 0.9];

```
>> b = [1]; a = [1, -1, 0.9]; n = [-20:100];
>> h = impz(b,a,n);
>> subplot(2,1,1); stem(n,h);
>> title('Impulse Response');
>> xlabel('n'); ylabel('h(n)')
```

```
>> n=-20:100;
>> x = (n-0) >= 0; % x = stepseq(0,-20,100);
>> s = filter(b,a,x);
>> subplot(2,1,2); stem(n,s)
>> title('Step Response');
>> xlabel('n'); ylabel('s(n)')
```





- To determine the stability of the system, we have to determine h(n) for all n.
- Although we have not described a method to solve the difference equation
- we can use the plot of the impulse response to observe that h(n) is practically zero for n > 120. Hence the sum

$$\sum |h[n]|$$

```
>> sum(abs(h)) ans = 14.8785
```

which implies that the system is stable.

- Let us consider the convolution given in previous example. The input sequence is of finite duration x(n) = u(n) u(n-10)
- while the impulse response is of infinite duration

$$h(n) = (0.9)^n u(n)$$

• Determine y(n) = x(n) * h(n).

- If the LTI system, given by the impulse response h(n), can be described by a difference equation, then y(n) can be obtained from the filter function.
- From the h(n) expression

$$(0.9)h[n-1] = (0.9)(0.9)^{n-1}u[n-1] = (0.9)^{n}u[n-1]$$

$$h[n] - (0.9)h[n-1] = (0.9)^{n}u[n] - (0.9)^{n}u[n-1]$$

$$= (0.9)^{n}(u[n] - u[n-1]) = (0.9)^{n}\delta[n]$$

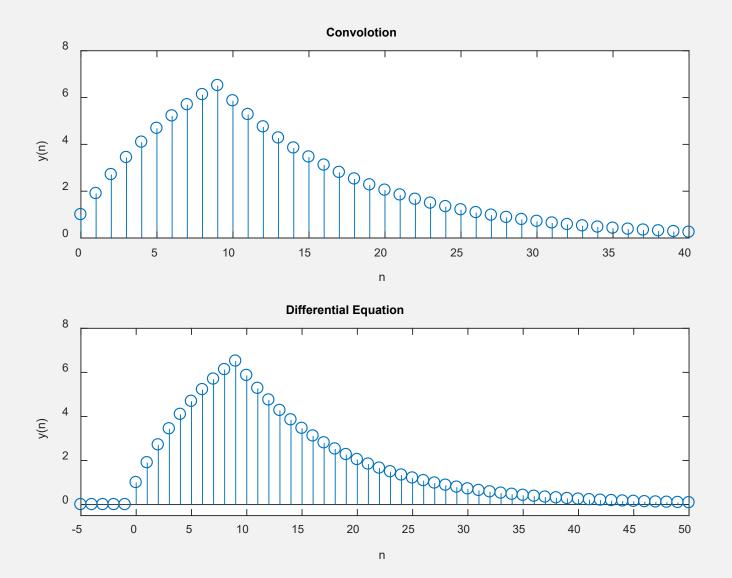
$$= \delta[n]$$

- The last step follows from the fact that $\delta(n)$ is nonzero only at n=0.
- By definition h(n) is the output of an LTI system when the input is $\delta(n)$.
- Hence substituting x(n) for $\delta(n)$ and y(n) for h(n), the difference equation is

$$y[n]-(0.9)y[n-1]=x[n]$$

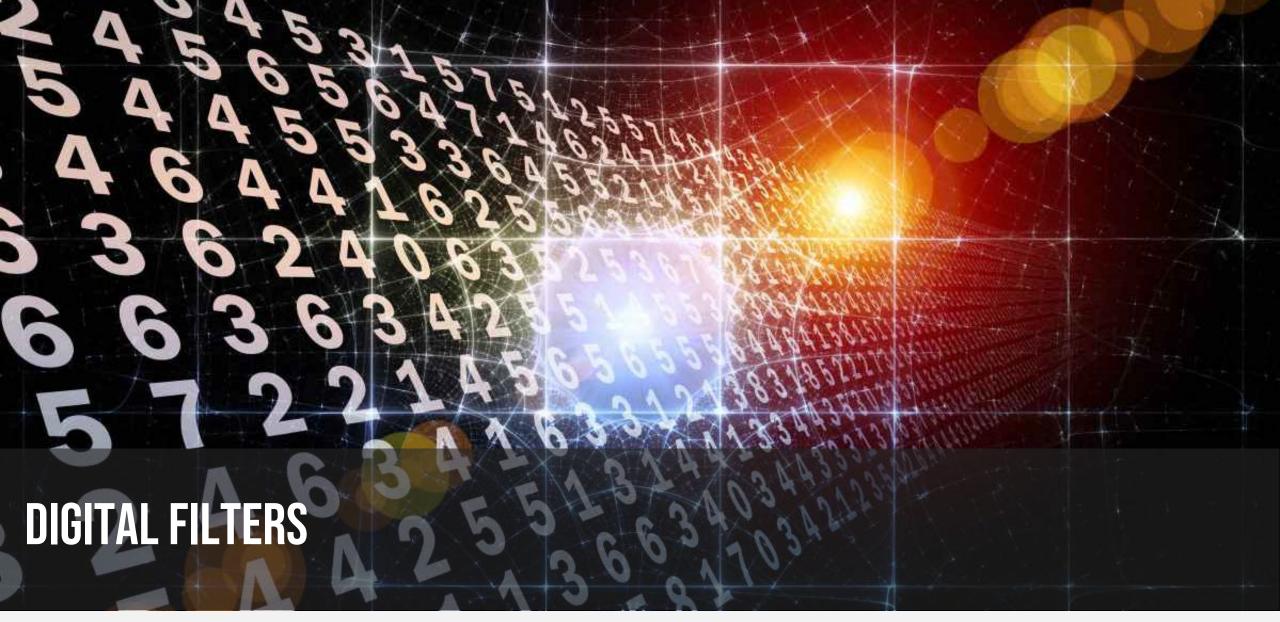
• MATLAB's filter function can be used to compute the convolution indirectly.

```
>> b = [1]; a = [1,-0.9];
>> n = -5:50;
>> x = stepseq(0,-5,50) - stepseq(10,-5,50);
>> y = filter(b,a,x);
>> subplot(2,1,2); stem(n,y);
>> title('Output sequence')
>> xlabel('n'); ylabel('y(n)'); axis([-5,50,-0.5,8])
```



```
err = y1(1:40) - y2(1:40);
mean(err);

disp(['Error ' num2str(mean(err))]);
```



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DIGITAL FILTERS

- Two types of digital filters
 - FIR filter If the unit impulse response of an LTI system is of finite duration, then the system is called a finite-duration impulse response (or FIR) filter.

$$y[n] = \sum_{m=0}^{M} b_m x[n-m]$$

- h(n) = 0 for n < n1 and for n > n2
- *IIR filter* If the impulse response of an LTI system is of infinite duration, then the system is called an *infinite-duration impulse response* (or IIR) filter.

$$\sum_{k=0}^{N} a_k y [n-k] = x[n]$$

MATLAB IMPLEMENTATION

- FIR filters are represented either as impulse response values $\{h(n)\}$ or as difference equation coefficients $\{b_m\}$ and $\{a_0 = 1\}$.
 - conv(x,h)
 - filter(b,1,x)
- IIR filters are described by the difference equation coefficients $\{b_m\}$ and $\{a_k\}$
 - filter(b,a,x)



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