# **İşaret İşleme**Frekans Cevabı-H11CD1

Dr. Meriç Çetin versiyon231120

Tablo 5.1 Fourier Dönüşümünün Özellikleri

Özellik	x(t)	X(w)
	$x(t)$ $x_1(t)$ $x_2(t)$	$X(w)$ $X_1(w)$ $X_2(w)$
Doğrusallık	$a_1x_1(t) + a_2x_2(t)$	$a_1 X_1(w) + a_2 X_2(w)$
Zamanda Öteleme	$x(t-t_0)$	$e^{-jwt_0}X(w)$
Frekans -domeninde Öteleme	$e^{jw_0t}x(t)$	$X(w-w_0)$
Zamanda Ölçekleme	x(at)	$\frac{1}{ a }X\left(\frac{w}{a}\right)$
Zamanda Geri Dönüş	x(-t)	X(-w)
Zamanda Türev	$\frac{d}{dt}x(t)$	jwX(w)
Frekans-domeninde Türev	-jtx(t)	$\frac{d}{dw}X(w)$
Çifteşlik	X(t)	$2\pi x(-w)$
Konvolüsyon	$x_1(t) * x_2(t)$	$X_1(w)X_2(w)$
Çarpma	$x_1(t)x_2(t)$	$\frac{1}{2\pi}X_1(w) * X_2(w)$
Parseval Bağıntısı	$\int_{-\infty}^{+\infty}  x(t) ^2 dt =$	$\frac{1}{2\pi} \int_{-\infty}^{+\infty}  X(w) ^2 dw$

Tablo 5.2 Bazı Fourier Dönüşüm Çiftleri

x(t)	X(w)	X(s)
$\delta(t)$	1	1
u(t)	$\pi\delta(w) + \frac{1}{jw}$	$\frac{1}{s}$
− <i>u</i> (− <i>t</i> )	$\pi\delta(w) - \frac{1}{jw}$	$\frac{1}{s}$
1	$2\pi\delta(w)$	
sgn(t)	2 jw	
tu(t)		$\frac{1}{s^2}$
$e^{-at}u(t)$	$\frac{1}{jw+a}$	$\frac{1}{s+a}$
$-e^{-at}u(-t)$	$\frac{1}{jw+a}$	$\frac{1}{s+a}$
$te^{-at}u(t)$	$\frac{1}{(jw+a)^2}$	$\frac{1}{(s+a)^2}$
$-te^{-at}u(-t)$	$\frac{1}{(jw+a)^2}$	$\frac{1}{(s+a)^2}$

x(t)	X(w)	X(s)
$e^{-at}\cos(w_0t)u(t)$	$\frac{jw+a}{(jw+a)^2+w_0^2}$	$\frac{s+a}{(s+a)^2+w_0^2}$
$e^{-at}\sin(w_0t)u(t)$	$\frac{w_0}{(jw+a)^2 + w_0^2}$	$\frac{w_0}{(s+a)^2+w_0^2}$
$e^{\mp jw_0t}$	$2\pi\delta(w\pm w_0)$	
$\cos(w_0t)$	$\pi\delta(w-w_0)+\pi\delta(w+w_0)$	
$\sin(w_0t)$	$-j\pi\delta(w-w_0)+j\pi\delta(w+w_0)$	
e-a t	$\frac{2a}{a^2 + w^2}$	
$\frac{1}{a^2+t^2}$	$\frac{\pi}{a}e^{-a w }$	
$P_a(t)$	$2\frac{\sin(aw)}{w}$	
$\frac{\sin(at)}{\pi t}$	$P_a(w)$	
e-at2	$\sqrt{\frac{\pi}{a}}e^{-\frac{w^2}{4a}}$	
$\sum_{k=-\infty}^{\infty} \delta(t - kT_0)$	$w_0 \sum_{k=-\infty}^{\infty} \delta(w - kw_0)$	

## Frekans Cevabi

h(t) darbe cevabı bilinen bir sistemin girişine belli bir x(t) sinyali uygulandığında y(t) çıkışının

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

konvolüsyonu ile bulunabileceği ve bu nedenle de h(t) darbe cevabının sistemin giriş-çıkış ilişkisini ifade etmede kullanılabileceği ifade edilmişti. Fourier dönüşümünün özelliklerinden olan

$$x_1(t) * x_2(t) \leftrightarrow X_1(w)X_2(w)$$
 şeklindeki konvolüsyon özelliği kullanılırsa,

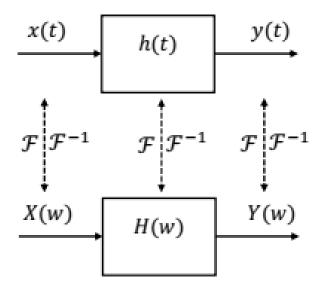
$$y(t) = x(t) * h(t) \leftrightarrow X(w)H(w) = Y(w)$$

ifadesi elde edilir. Burada X(w), H(w) ve Y(w) sırasıyla x(t), h(t) ve y(t)'nin Fourier dönüşümleridir. Aynı eşitlik

$$H(w) = \frac{Y(w)}{X(w)}$$

### Frekans Cevabi - devam

ile ifade edilebilir. h(t)'nin Fourier dönüşümü H(w)'ye frekans cevabı denir. Artık, darbe cevabı bilinen bir sisteme belli bir sinyal uygulandığında çıkış sinyali frekans cevabı yardımıyla bulunabilir. Bu nedenle de DZD bir sistemin frekans cevabı sistemin giriş-çıkış ilişkisini ifade etmede kullanılabilir. Bunu aşağıdaki şekille görmek mümkündür.



## Frekans Cevabi - devam

Frekans cevabı H(w), bağımsız frekans değişkenine bağlı bir fonksiyon olup karmaşık değerler alabilmektedir. Bu nedenle H(w) kutupsal olarak

$$H(w) = |H(w)|e^{j\langle H(w)}$$
 
$$Y(w) = X(w)H(w)$$
 
$$|Y(w)|e^{j\langle Y(w)} = |X(w)|e^{j\langle X(w)}|H(w)|e^{j\langle H(w)}$$

şeklinde yazılabileceği için çıkış sinyalinin genliği

$$|Y(w)| = |X(w)||H(w)|$$

fazı da

$$\langle Y(w) = \langle X(w) + \langle H(w) \rangle$$

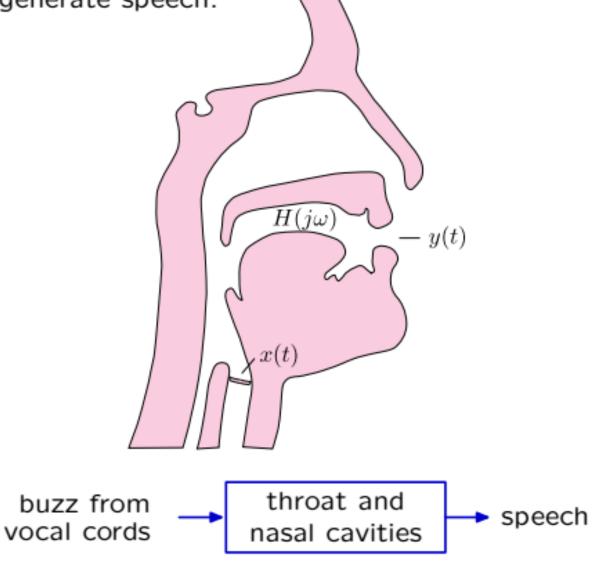
biçiminde ifade edilebilir.

## Filtre Kavramı

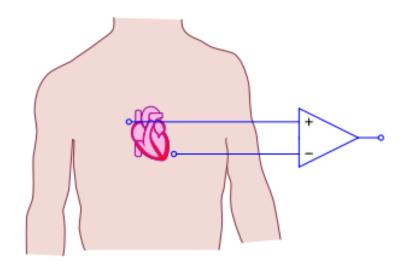
Önceki altkısımda da görüldüğü gibi DZD bir sistemin girişinden periyodik bir sinyal uygulandığında, sistem bu giriş sinyalinin frekans bileşenlerinden bazılarını geçirmekte bazılarını da süzmektedir. Bu açıdan bakıldığında her DZD sistem aslında frekans domeninde bir filtre gibi görülebilir. DZD sistem giriş sinyalinin frekans bileşenlerinden bazılarının genliklerini sıfırlamakta ve dolayısıyla bu bileşenleri süzmektedir. Bileşenlerin faz ve genliklerinin nasıl etkileneceği tamamen DZD sistemin frekans cevabı tarafından belirlemektedir. Literatürde pek çok filtre tipi mevcuttur. Bunlardan belli başlı olanları şunlardır:

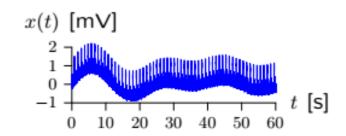
#### Source-Filter Model of Speech Production

Vibrations of the vocal cords are "filtered" by the mouth and nasal cavities to generate speech.



An electrocardiogram is a record of electrical potentials that are generated by the heart and measured on the surface of the chest.





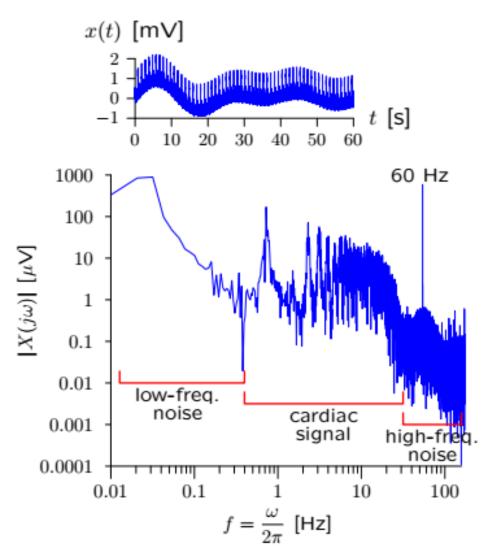
In addition to electrical responses of heart, electrodes on the skin also pick up other electrical signals that we regard as "noise."

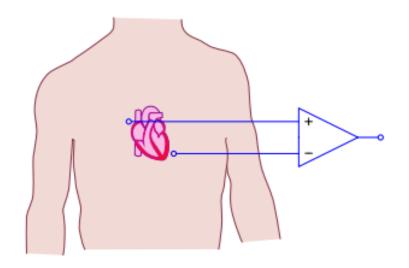
ECG and analysis by T. F. Weiss

We wish to design a filter to eliminate the noise.

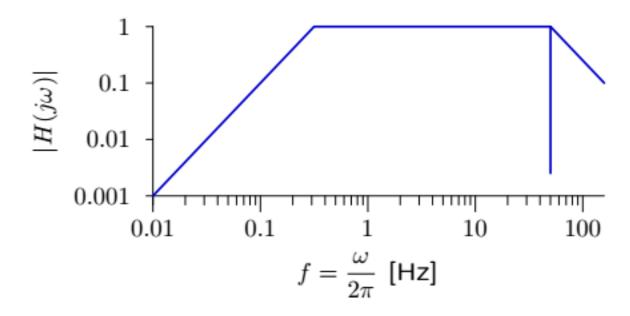
$$x(t) \longrightarrow$$
 filter  $\longrightarrow y(t)$ 

We can identify "noise" using the Fourier transform.





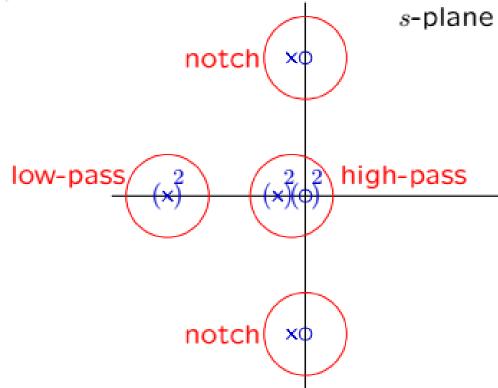
Filter design: low-pass flter + high-pass filter + notch.



#### Electrocardiogram: Check Yourself

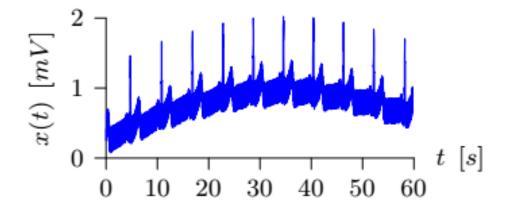
Which poles and zeros are associated with

- the high-pass filter?
- the low-pass filter?
- the notch filter?

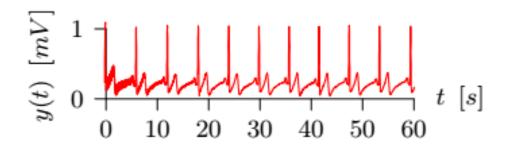


Filtering is a simple way to reduce unwanted noise.

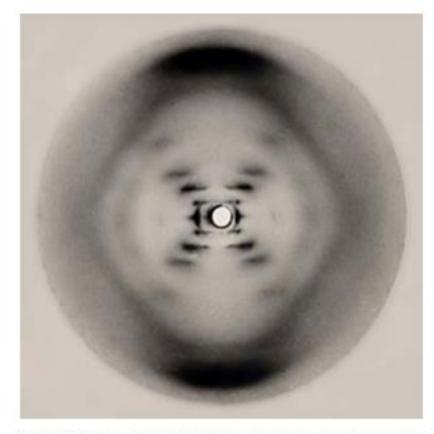
#### Unfiltered ECG



#### Filtered ECG



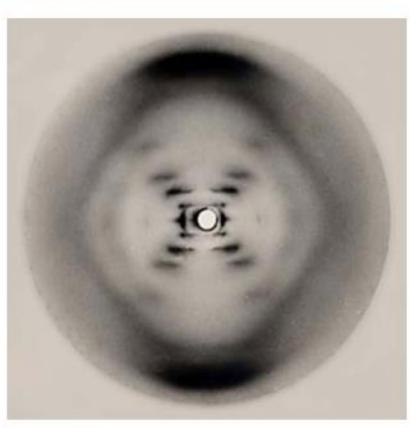
Taken by Rosalind Franklin, this image sparked Watson and Crick's insight into the double helix.



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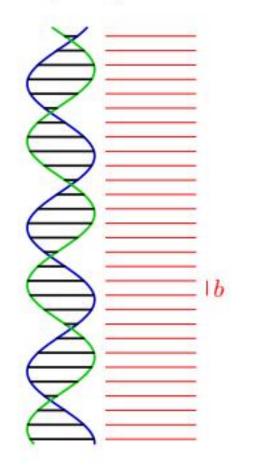
This is an x-ray crystallographic image of DNA, and it shows the Fourier transform of the structure of DNA.

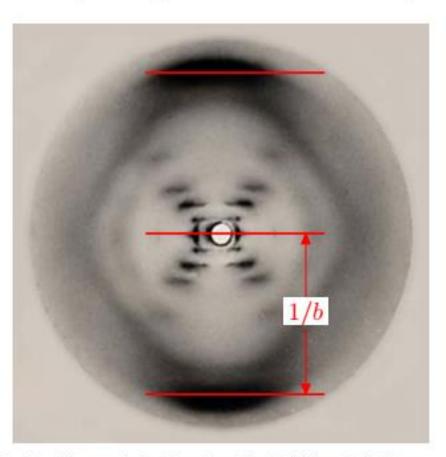




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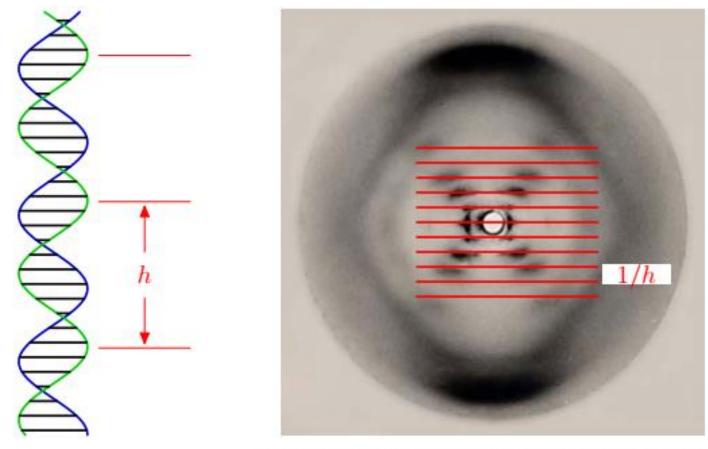
High-frequency bands indicate repeating structure of base pairs.





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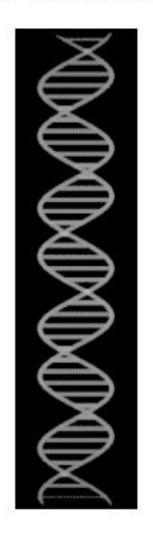
Low-frequency bands indicate a lower frequency repeating structure.

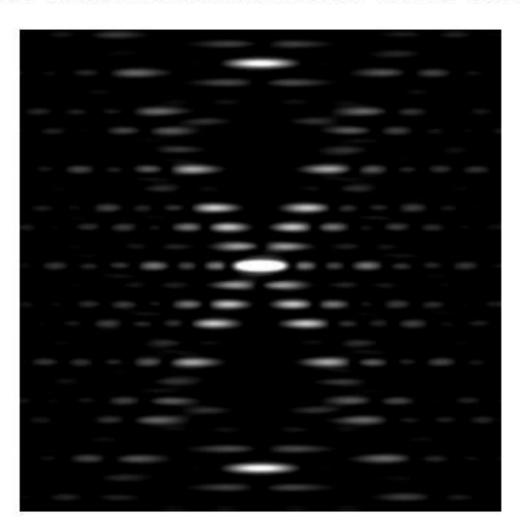


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#### Simulation

Easy to calculate relation between structure and Fourier transform.

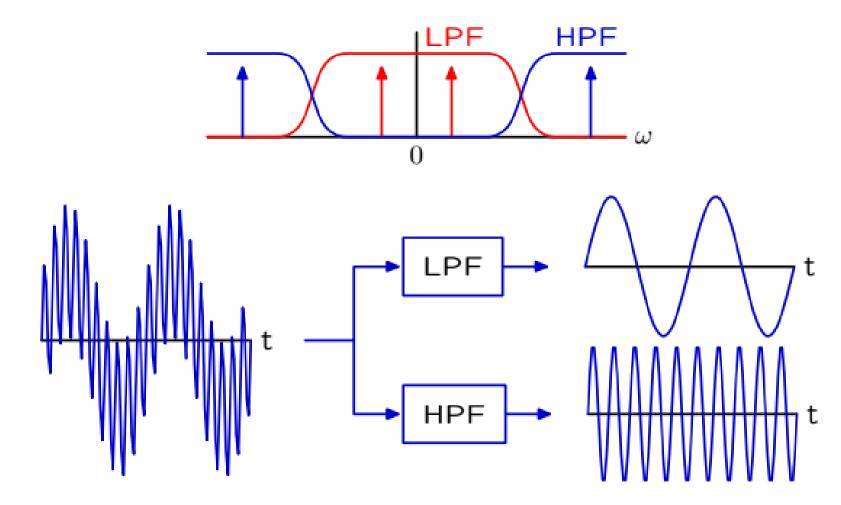




#### **Filtering**

Systems can be designed to selectively pass certain frequency bands.

Examples: low-pass filter (LPF) and high-pass filter (HPF).



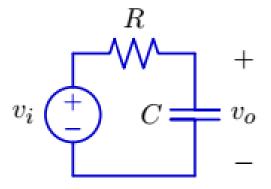
#### **Filtering**

Notion of a filter.

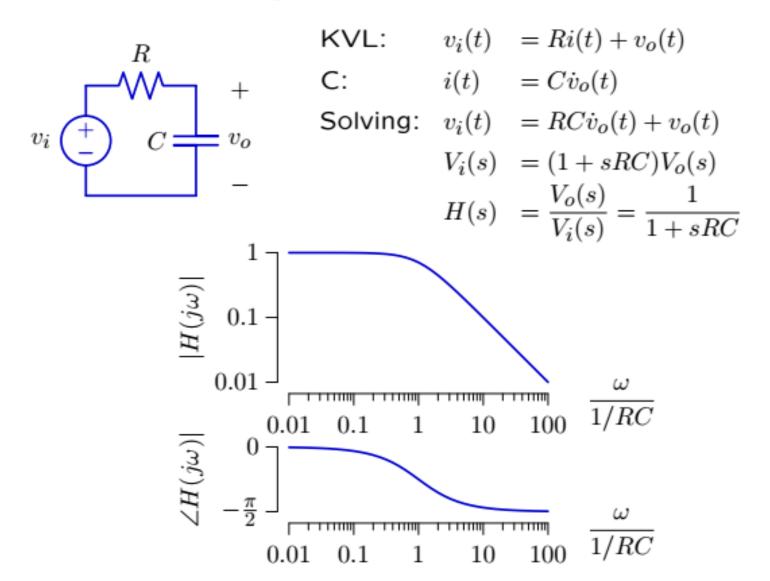
#### LTI systems

- cannot create new frequencies.
- can scale magnitudes and shift phases of existing components.

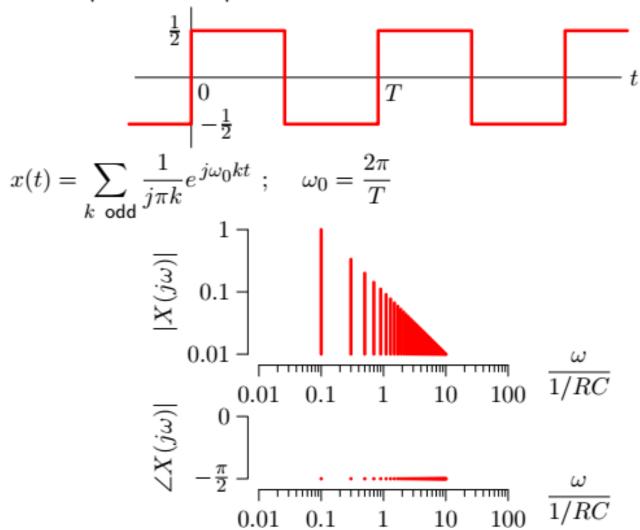
Example: Low-Pass Filtering with an RC circuit



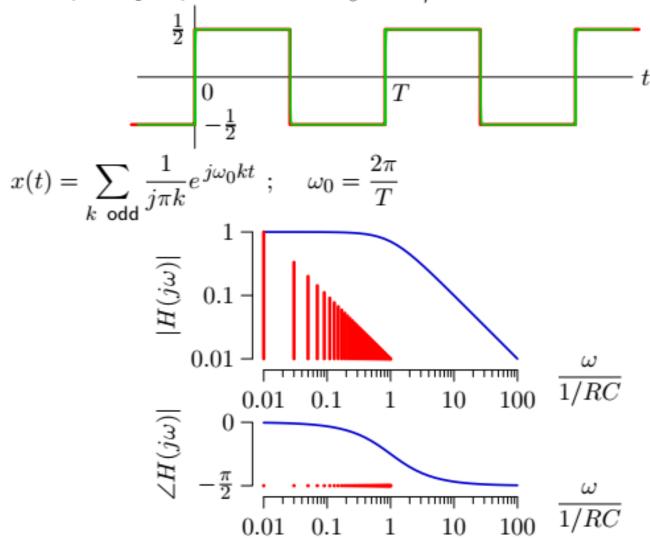
Calculate the frequency response of an RC circuit.



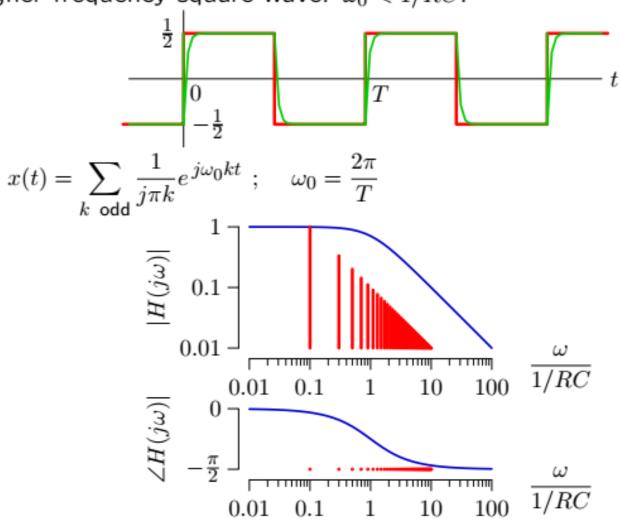
Let the input be a square wave.



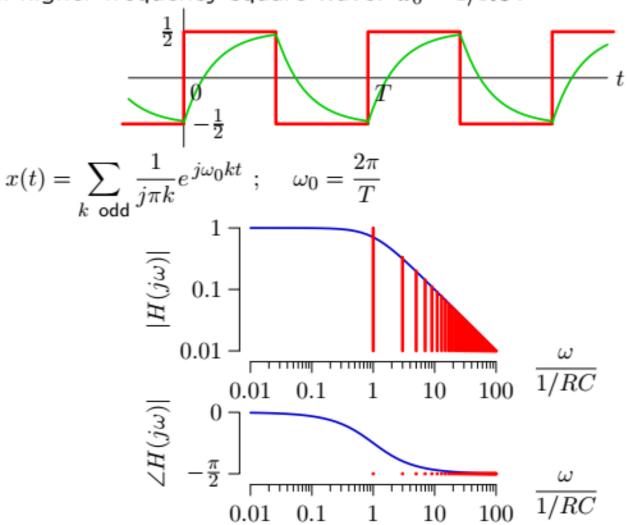
Low frequency square wave:  $\omega_0 << 1/RC$ .



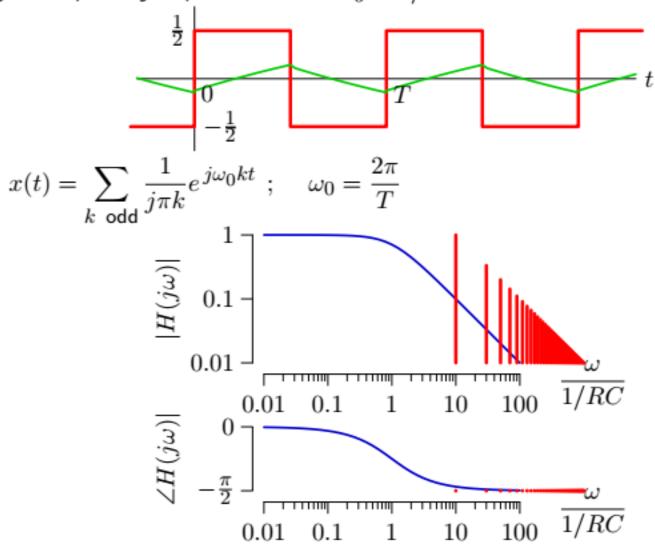
Higher frequency square wave:  $\omega_0 < 1/RC$ .



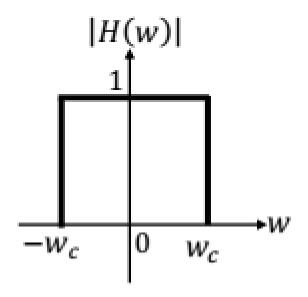
Still higher frequency square wave:  $\omega_0 = 1/RC$ .



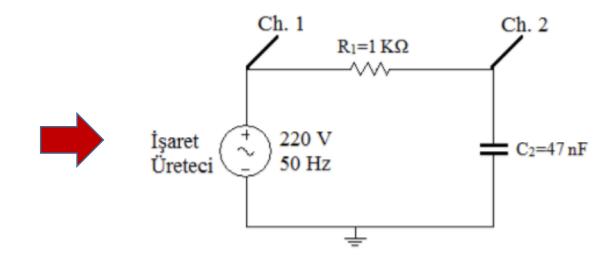
High frequency square wave:  $\omega_0 > 1/RC$ .



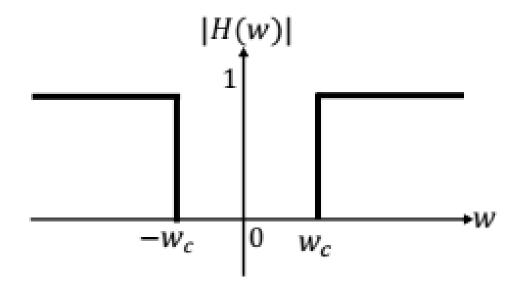
#### İdeal Alçak Geçiren Filtre



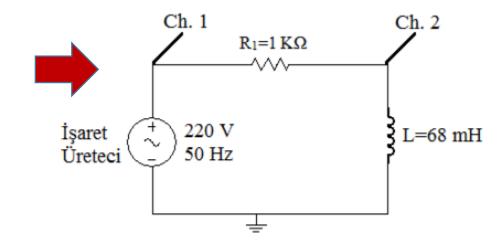
$$|H(w)| = P_a(w_c) = \begin{cases} 1 & |w| \le w_c \\ 0 & |w| > w_c \end{cases}$$



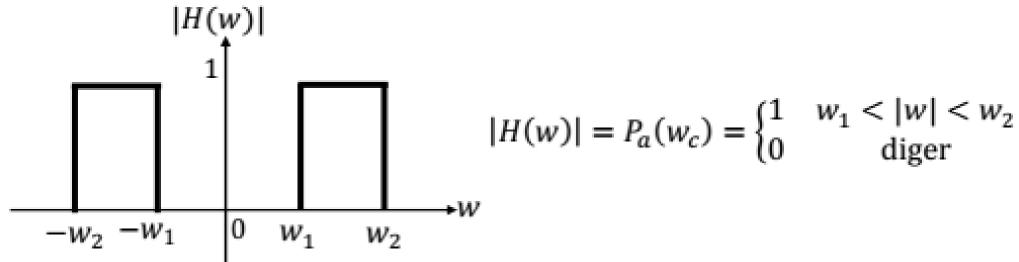
#### İdeal Yüksek Geçiren Filtre

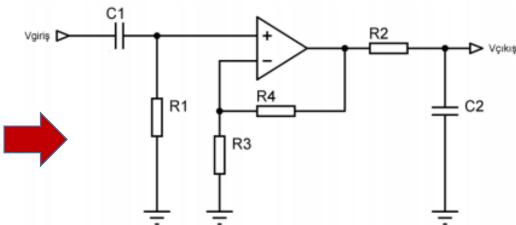


$$|H(w)| = P_a(w_c) = \begin{cases} 1 & |w| > w_c \\ 0 & |w| \le w_c \end{cases}$$

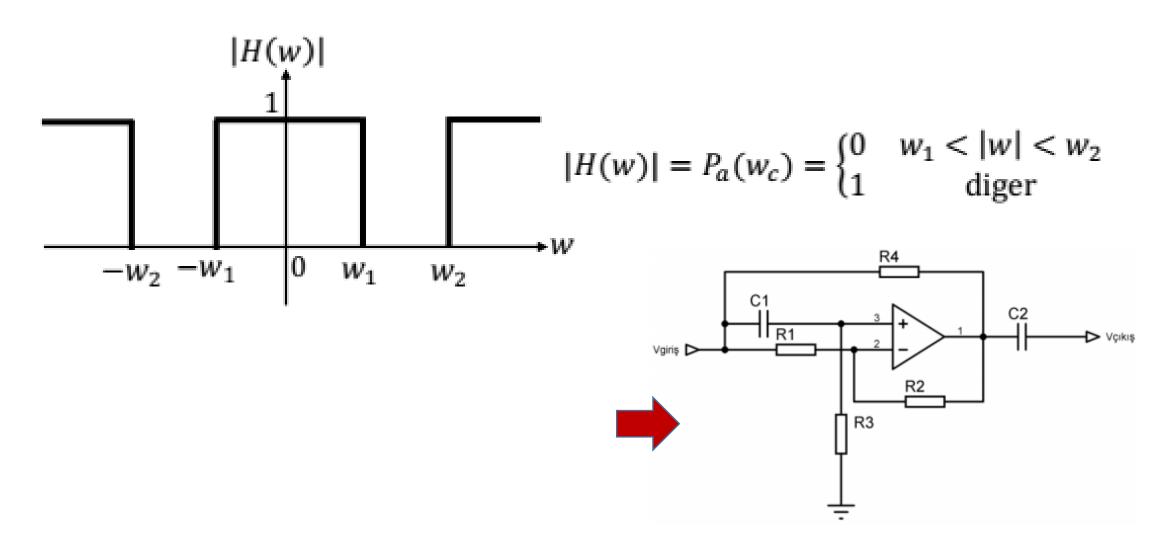


#### İdeal Bant Geçiren Filtre





#### İdeal Bant Söndüren Filtre



## Diferansiyel Denklemlerle İlişkisi

$$a_0y(t) + a_1\dot{y}(t) + a_2\ddot{y}(t) + \dots + a_Ny^{(N)}(t) = b_0x(t) + b_1\dot{x}(t) + b_2\ddot{x}(t) + \dots + b_Mx^{(M)}(t)$$

buradaki  $a_k$  ve  $b_k$  katsayıları reel ve sabit katsayılardır. Şimdi, giriş-çıkış ilişkisi bu şekildeki bir denklem ile ifade edilen DZD bir sistemin frekans cevabını bulalım: İlk olarak denklemin her iki tarafının Fourier dönüşümünü alalım:

$$a_0Y(w) + a_1jwY(w) + a_2(jw)^2Y(w) + \dots + a_N(jw)^NY(w) = b_0X(w) + b_1jwX(w) + \dots + b_M(jw)^MX(w)$$
$$(a_0 + a_1jw + a_2(jw)^2 + \dots + a_N(jw)^N)Y(w) = (b_0 + b_1jw + \dots + b_M(jw)^M)X(w)$$

Şimdi frekans cevabını yazabiliriz:

$$H(w) = \frac{Y(w)}{X(w)} = \frac{b_0 + b_1 j w + \dots + b_M (j w)^M}{a_0 + a_1 j w + a_2 (j w)^2 + \dots + a_N (j w)^N}$$

**Örnek**: Giriş-çıkış ilişkisi aşağıdaki gibi sabit katsayılı doğrusal bir denklem ile ifade edilen nedensel bir DZD sistemi ele alalım. Sistemin girişinden uygulanan  $x(t) = e^{-t}u(t)$  sinyaline karşı sistemin y(t) çıkışını bulunuz. Not: Sistemin başlangıç koşulları sıfırdır.

$$x(t) = e^{-t}u(t) \quad \to \quad \ddot{y}(t) + 1.6\dot{y}(t) + y(t) = 3\ddot{x}(t) + 6.2\dot{x}(t) + 4x(t) \quad \to \quad y(t) = ?$$

İlk olarak denklemin her iki tarafının Fourier dönüşümünü alalım:

$$(jw)^2Y(w) + 1.6jwY(w) + Y(w) = 3(jw)^2X(w) + 6.2jwX(w) + 4X(w).$$

Şimdi frekans cevabını yazabiliriz:

$$H(w) = \frac{Y(w)}{X(w)} = \frac{3(jw)^2 + 6.2jw + 4}{(jw)^2 + 1.6jw + 1}$$

Giriş sinyalinin Fourier dönüşümü

$$x(t) = e^{-t}u(t) \longleftrightarrow X(w) = \frac{1}{jw+1}$$

şeklindedir, böylece çıkışın Fourier dönüşümü şu şekilde bulunur:

$$Y(w) = H(w)X(w)$$

$$= \frac{3(jw)^2 + 6.2jw + 4}{(jw)^2 + 1.6jw + 1} \frac{1}{jw + 1}$$

Y(jw)'yi kısmi kesirlere ayıralım:

$$Y(jw) = \frac{3(jw)^2 + 6.2jw + 4}{(jw)^2 + 1.6jw + 1} \frac{1}{jw + 1}$$

$$= \frac{2}{jw + 1} + \frac{jw + 2}{(jw)^2 + 1.6jw + 1}$$

$$= \frac{2}{jw + 1} + \frac{s + 2}{(jw + 0.8)^2 + 0.36}$$

$$= \frac{2}{jw + 1} + \frac{jw + 0.8}{(jw + 0.8)^2 + (0.6)^2} + 2\frac{0.6}{(jw + 0.8)^2 + (0.6)^2}$$

Y(s)'nin ters Fourier dönüşümünü alırsak çıkış sinyali y(t) bulunur:

Tablo 5.2 Bazı Fourier Dönüşüm Çiftleri

x(t)	X(w)	X(s)
$\delta(t)$	1	1
u(t)	$\pi\delta(w) + \frac{1}{jw}$	$\frac{1}{s}$
-u(-t)	$\pi\delta(w) - \frac{1}{jw}$	$\frac{1}{s}$
1	$2\pi\delta(w)$	
sgn(t)	2 jw	
tu(t)		$\frac{1}{s^2}$
$e^{-at}u(t)$	$\frac{1}{jw+a}$	$\frac{1}{s+a}$
$-e^{-at}u(-t)$	$\frac{1}{jw+a}$	$\frac{1}{s+a}$
$te^{-at}u(t)$	$\frac{1}{(jw+a)^2}$	$\frac{1}{(s+a)^2}$
$-te^{-at}u(-t)$	$\frac{1}{(jw+a)^2}$	$\frac{1}{(s+a)^2}$

x(t)	X(w)	X(s)
$e^{-at}\cos(w_0t)u(t)$	$\frac{jw+a}{(jw+a)^2+w_0^2}$	$\frac{s+a}{(s+a)^2+w_0^2}$
$e^{-at}\sin(w_0t)u(t)$	$\frac{w_0}{(jw+a)^2 + w_0^2}$	$\frac{w_0}{(s+a)^2 + w_0^2}$
$e^{\mp jw_0t}$	$2\pi\delta(w\pm w_0)$	
$\cos(w_0t)$	$\pi\delta(w-w_0)+\pi\delta(w+w_0)$	
$\sin(w_0t)$	$-j\pi\delta(w-w_0)+j\pi\delta(w+w_0)$	
e-a t	$\frac{2a}{a^2 + w^2}$	
$\frac{1}{a^2+t^2}$	$\frac{\pi}{a}e^{-a w }$	
$P_a(t)$	$2\frac{\sin(aw)}{w}$	
$\frac{\sin(at)}{\pi t}$	$P_a(w)$	
e-at2	$\sqrt{\frac{\pi}{a}}e^{-\frac{w^2}{4a}}$	
$\sum_{k=-\infty}^{\infty} \delta(t - kT_0)$	$w_0 \sum_{k=-\infty}^{\infty} \delta(w - kw_0)$	

Y(jw)'yi kısmi kesirlere ayıralım:

$$Y(jw) = \frac{3(jw)^2 + 6.2jw + 4}{(jw)^2 + 1.6jw + 1} \frac{1}{jw + 1}$$

$$= \frac{2}{jw + 1} + \frac{jw + 2}{(jw)^2 + 1.6jw + 1}$$

$$= \frac{2}{jw + 1} + \frac{s + 2}{(jw + 0.8)^2 + 0.36}$$

$$= \frac{2}{jw + 1} + \frac{jw + 0.8}{(jw + 0.8)^2 + (0.6)^2} + 2\frac{0.6}{(jw + 0.8)^2 + (0.6)^2}$$

Y(s)'nin ters Fourier dönüşümünü alırsak çıkış sinyali y(t) bulunur:

$$y(t) = 2e^{-t}u(t) + e^{-0.8t}\cos 0.6t u(t) + 2e^{-0.8t}\sin 0.6t u(t)$$

## Bu ders notu için faydalanılan kaynaklar

## EEEN343 Sinyaller ve Sistemler Ders Notlan

MIT OpenCourseWare http://ocw.mit.edu

6.003 Signals and Systems Fall 2011 Prof. Dr. Serdar İplikçi

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