

# Part B: Adaptive signal processing



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RLS

## RLS algorithm

Initialization:

$$\underline{\mathbf{r}}_{ex}[0] = \underline{\mathbf{0}} ; \overline{\mathbf{R}}_x^{-1}[0] = \delta^{-1} \mathbf{I} \text{ with } \delta \text{ small}$$

For  $k \geq 0$  :

$$\underline{\mathbf{g}}[k+1] = \frac{\overline{\mathbf{R}}^{-1}[k] \underline{\mathbf{x}}[k+1]}{\lambda^2 + \underline{\mathbf{x}}^t[k+1] \overline{\mathbf{R}}^{-1}[k] \underline{\mathbf{x}}[k+1]}$$

$$\overline{\mathbf{R}}^{-1}[k+1] = \lambda^{-2} \left( \overline{\mathbf{R}}^{-1}[k] - \underline{\mathbf{g}}[k+1] \cdot \underline{\mathbf{x}}^t[k+1] \overline{\mathbf{R}}^{-1}[k] \right)$$

$$\underline{\mathbf{r}}_{ex}[k+1] = \lambda^2 \underline{\mathbf{r}}_{ex}[k] + \underline{\mathbf{x}}[k+1] e[k+1]$$

$$\underline{\mathbf{w}}[k+1] = \overline{\mathbf{R}}_x^{-1}[k+1] \cdot \underline{\mathbf{r}}_{ex}[k+1]$$

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## RLS

Compare RLS with "LMS/Newton"

"LMS/Newton":  $\underline{\mathbf{w}}[k+1] = \underline{\mathbf{w}}[k] + 2\alpha \mathbf{R}_x^{-1} \underline{\mathbf{x}}[k] r[k]$

Update RLS can be rewritten as (see Appendix):

$$\begin{aligned} \underline{\mathbf{w}}[k+1] &= \underline{\mathbf{w}}[k] + \underline{\mathbf{g}}[k+1] (e[k+1] - \underline{\mathbf{x}}^t[k+1] \underline{\mathbf{w}}[k]) \\ &= \underline{\mathbf{w}}[k] + \underline{\mathbf{g}}[k+1] r[k+1] \end{aligned}$$

with gain vector:

$$\underline{\mathbf{g}}[k+1] = \frac{\overline{\mathbf{R}}^{-1}[k] \underline{\mathbf{x}}[k+1]}{\lambda^2 + \underline{\mathbf{x}}^t[k+1] \overline{\mathbf{R}}^{-1}[k] \underline{\mathbf{x}}[k+1]}$$

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Effective memory:  $1/(1-\lambda)$

$\lambda$ : 0.99~1

$\bar{R}^{-1}[k]$ : should be Hermitian, positive definiteness

$$r_{ij}[k] = r_{ji}^*[k]$$

$$\bar{R}^{-1}[k] = [\bar{R}^{-1}[k] + (\bar{R}^{-1}[k])^H]/2$$



# ADSP Advanced digital signal processing

## Main content ADSP course

- Part A: Stochastic Signal Processing
- Part B: Adaptive Signal Processing
- **Part C: Array Signal Processing (ASP) (including DOA)**
- Part D: Adaptive Array Signal Processing (AASP)

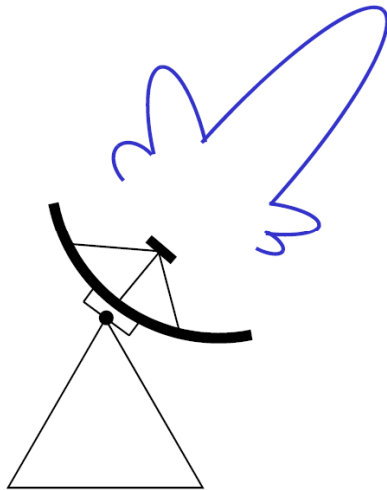
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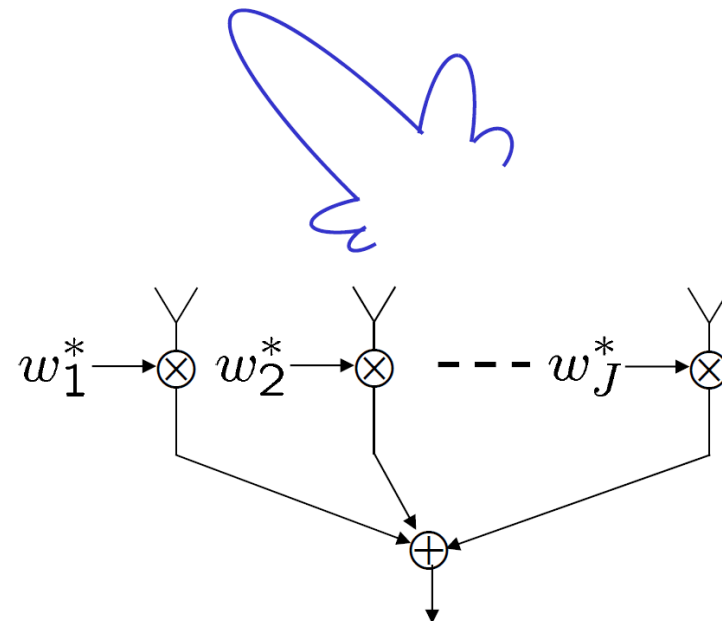
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## Introduction



Parabolic dish antenna  
(*continuous aperture*)



Sensor array antenna  
(*discrete spatial aperture*)

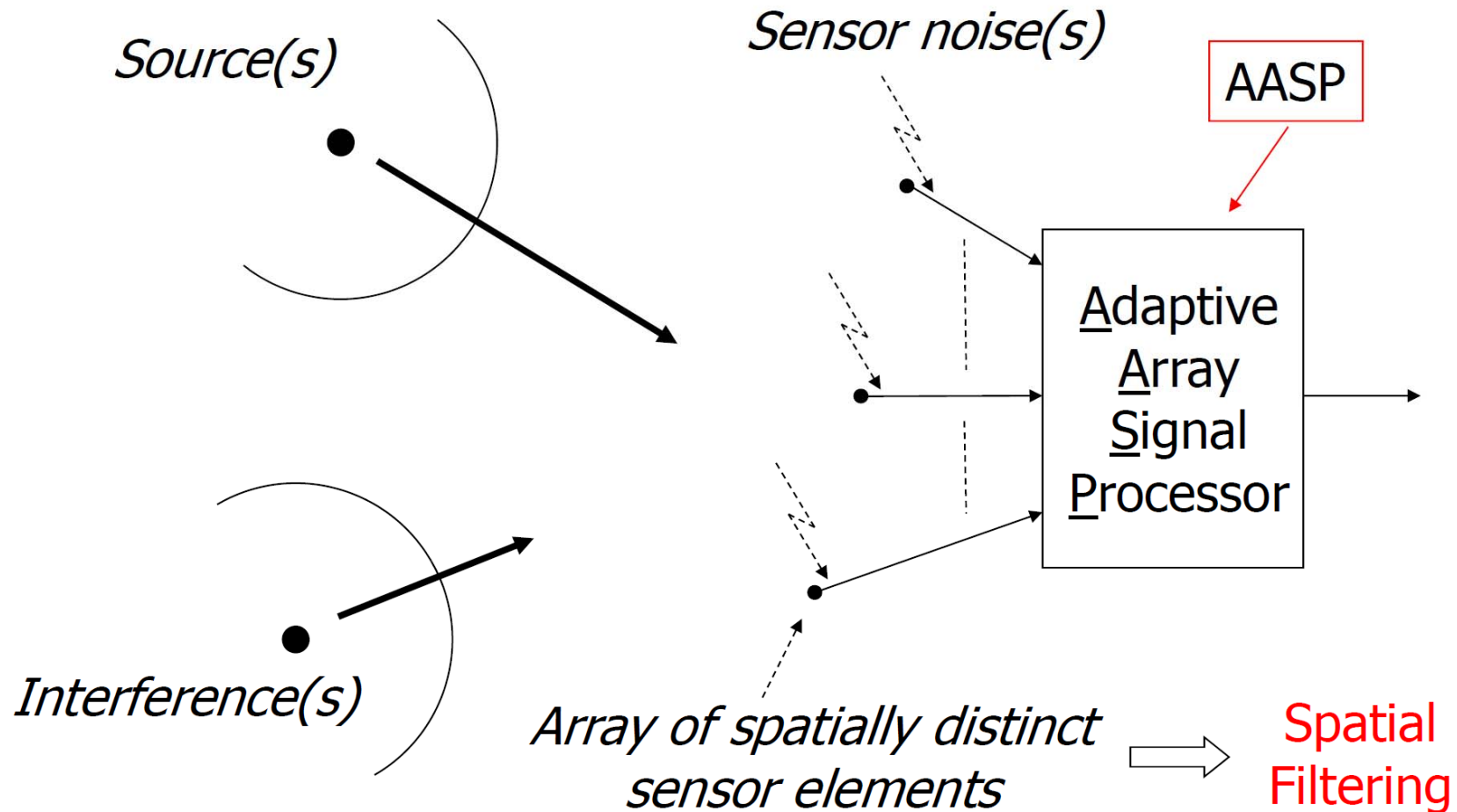
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## Introduction



# Part C: Array signal processing

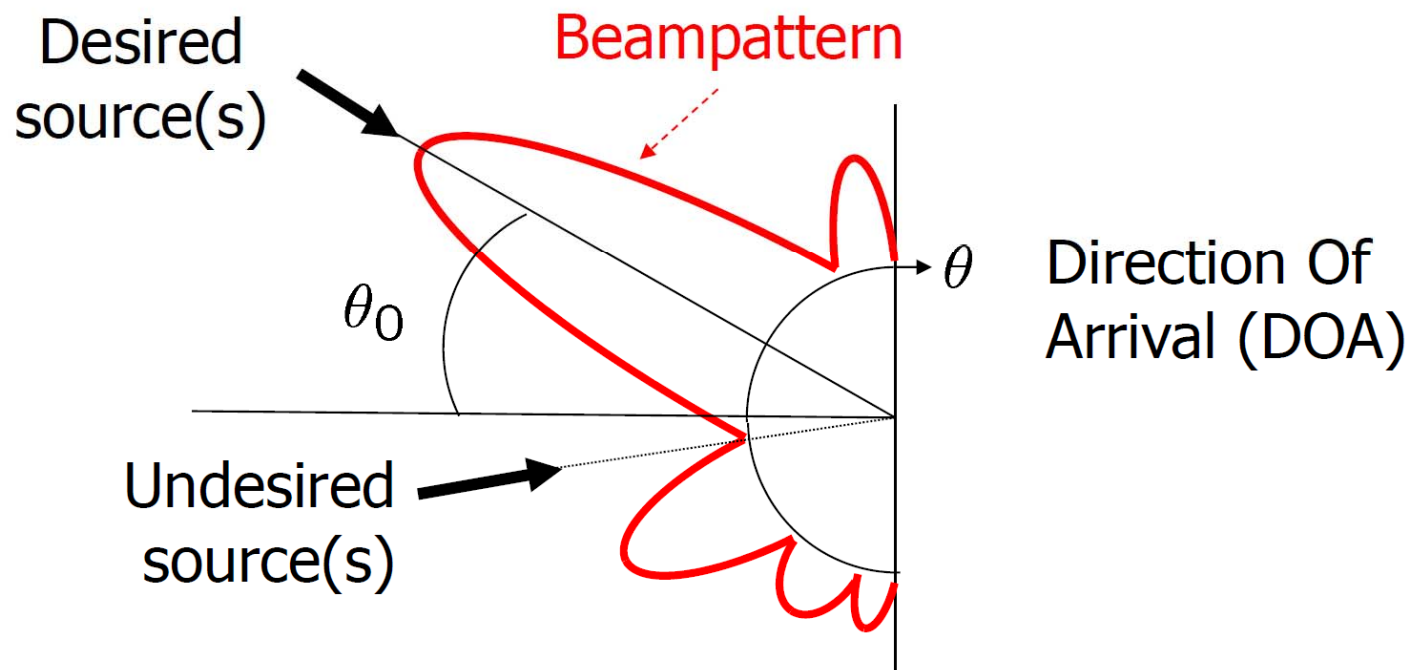


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## Introduction

Result of spatial filtering:



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## Introduction

### **Beamforming:**

Spatio temporal filtering to either direct or block the radiation or reception of signals in specified directions.

### **Result Array Signal Processing:**

Spatial filtering: Separate signals with possible overlapping frequency content but from different spatial locations.





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## Introduction

Furthermore:

- Able to 'look' in several directions simultaneously
- Signal enhancement: averaging different sensor measurements improves SNR
- Flexible spatial discrimination: size of spatial aperture can be adapted.
- Adaptivity --> able to adapt response

**AASP is versatile and flexible**

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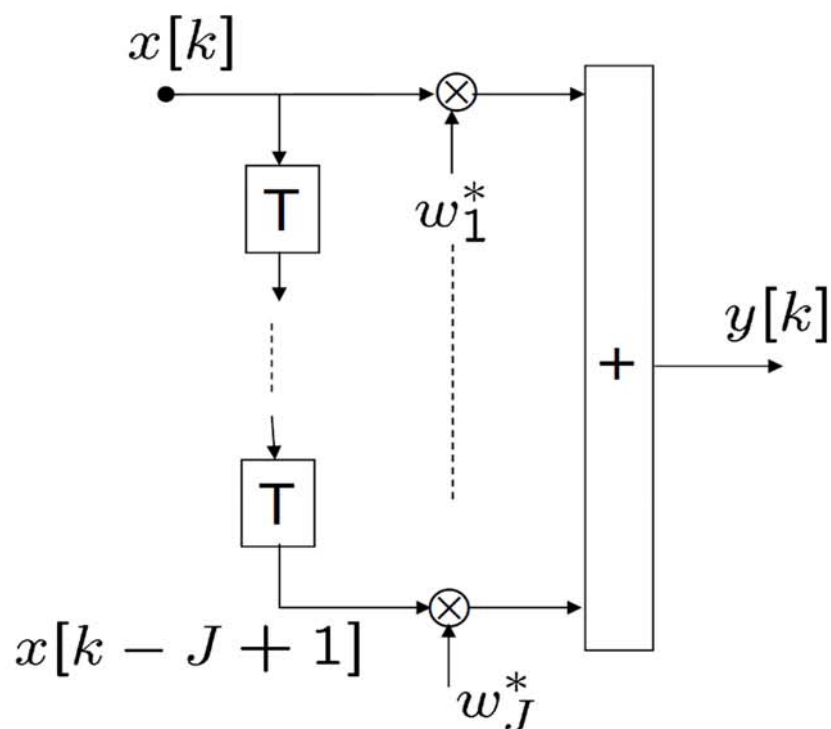
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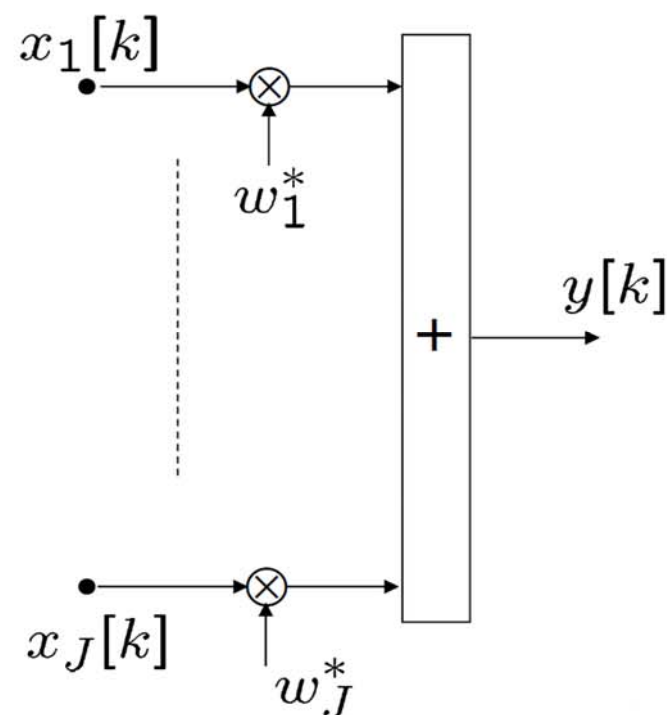
## Introduction

To provide insight into various aspects of AASP we use familiar methods and techniques from FIR filtering.

FIR:



Array:





However main differences ASP and FIR filtering:

- Source can have several parameters of interest (e.g. range, azimuth and elevation angle, polarization, temporal frequency content)
- Different signal often mutually correlated (multipath)
- Spatial sampling often nonuniform and multidimensional
- Uncertainty must often be included in characterization of individual sensor response and location (robust ASP techniques required)



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## Introduction

### General objective of AASP:

Detect or enhance desired signal(s) (increase SNR), while simultaneously reducing unwanted interference.

### Physical form array varies according to medium:

Microphone for pressure variations in air

Hydrophone for pressure variations in water (sonar)

Geophone for land base seismology

Radar of electromagnetic waves

Etc.

### Information of interest:

- Signal itself (teleconferencing, communications)
- Location of source (=DOA) (radar, sonar)
- Number of sources

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## ADSP

## Introduction

### AASP applications:

Radar: Phased-array, air traffic control

Sonar: Source localization and classification

Communication: Directional transmission and reception

Imaging: Ultrasonic, optical, tomographic

Geophysical exploration: Earth crust mapping, oil exploration

Astrophysical exploration: High resolution imaging of universe

Biomedical: Fetal heart monitoring

Acoustic: Hearing aids, transparent communication

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### Different scenario's

Bandwidth source  
Source position  
Array geometry  
Discrete-time signal representation  
Array signal model  
ASP unit

#### Assumptions:

Superposition principle applies to  
propagating wave signals

Homogeneous, lossless medium, neglect  
dispersion,

diffraction, changes in propagation speed

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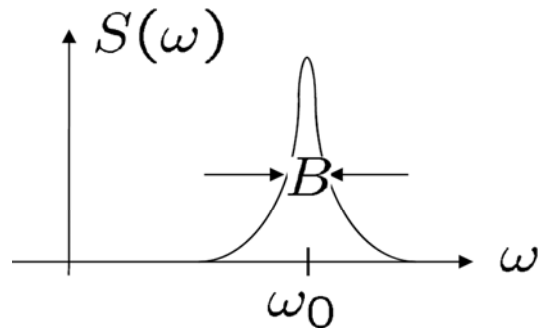
## Scenarios

### **Bandwidth source**

Analytical representation:  $s(t) = A(t)e^{j(\omega_0 t + \phi(t))}$

**Narrowband:**  $A(t)$  and  $\phi(t)$  vary slower than  $e^{j\omega_0 t}$

Narrowband:  $|\tau| \ll 1/B \Rightarrow$



$A(t - \tau) \approx A(t) = 1$  (usually)

$\phi(t - \tau) \approx \phi(t) = 0$  (usually)

$$\Rightarrow s(t - \tau) = A(t - \tau)e^{j\phi(t - \tau)}e^{j\omega_0(t - \tau)} \approx e^{-j\omega_0\tau} \cdot s(t)$$

Thus for narrow band: Time delay  $\Rightarrow$  phase shift

*In this course:* **mainly narrowband**

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## Scenarios

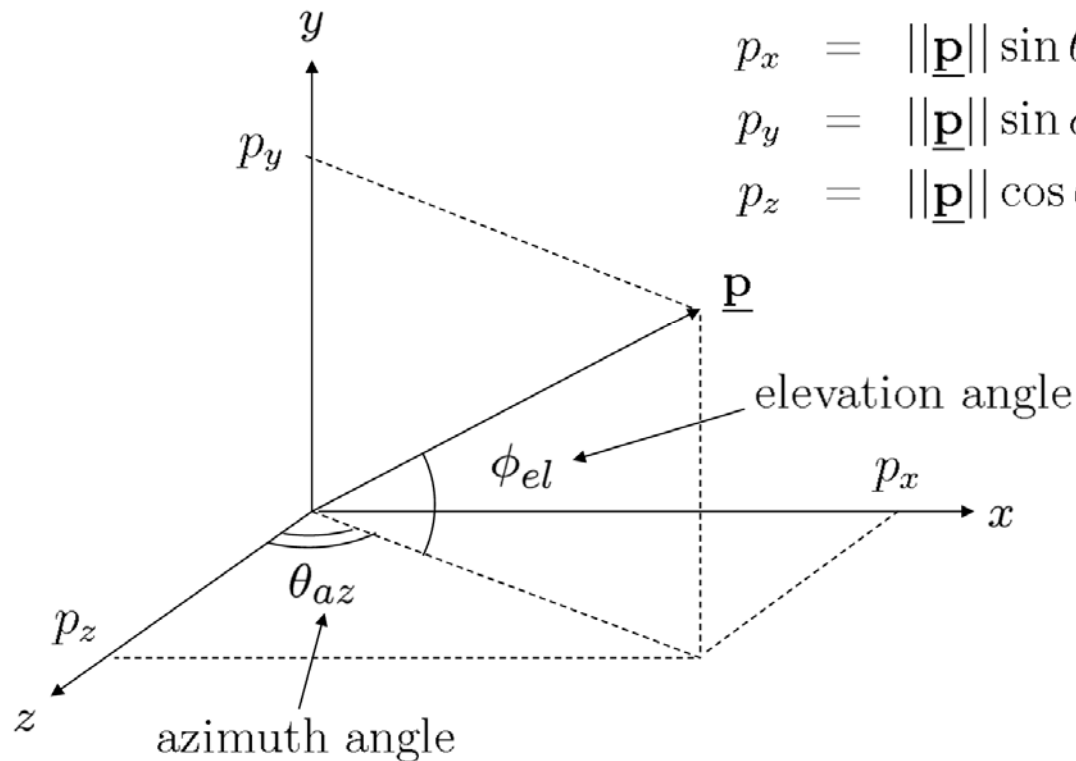
### Source position

$$\underline{\mathbf{p}} = (p_x, p_y, p_z)^t$$

$$p_x = ||\underline{\mathbf{p}}|| \sin \theta_{az} \cos \phi_{el}$$

$$p_y = ||\underline{\mathbf{p}}|| \sin \phi_{el}$$

$$p_z = ||\underline{\mathbf{p}}|| \cos \theta_{az} \cos \phi_{el}$$





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## ADSP

## Scenarios

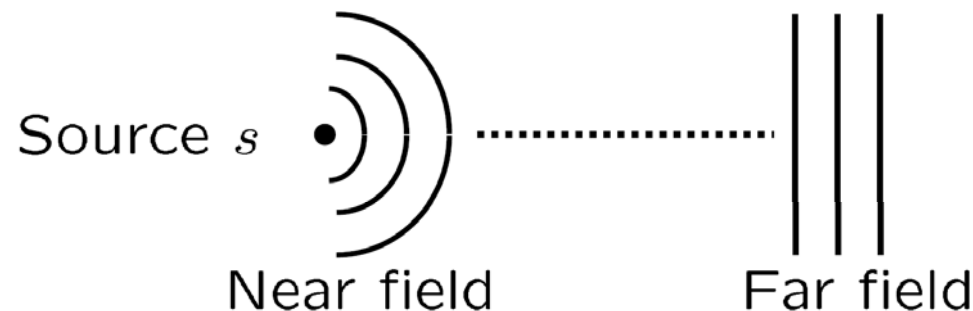
**Array aperture:** Volume (1D length) that collects incoming energy

**Far field:**

- Distance source - array  $\gg$  array aperture
- Plane wavefront

**Near field:**

- Distance source - array  $\ll$  array aperture
- Spherical wavefront



*In this course mainly* **Far field**

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## Scenarios

Propagation for **near field** (single frequency) source

$$s(t, \underline{\mathbf{p}}) = \frac{A}{||\underline{\mathbf{p}}||^2} e^{j\omega(t - \frac{||\underline{\mathbf{p}}||}{c})}$$

with  $\omega = 2\pi f$  and  $f = \frac{c}{\lambda}$

$\lambda =$  wavelength;  $c =$  speed in medium

*Note:* For acoustic sound in air  $c \approx 334[\text{m/sec}]$

$\Rightarrow$  Amplitude decays proportional to distance from source

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## Scenarios

In **far field** at position  $\underline{\mathbf{p}}_i$  monochromatic plane wave:

$$s(t, \underline{\mathbf{p}}_i) = Ae^{j\omega(t-\tau_i)} = Ae^{j\omega(t-\frac{\underline{\mathbf{v}}^t}{c} \cdot \underline{\mathbf{p}}_i)} = Ae^{j(\omega t - \underline{\mathbf{k}}^t \cdot \underline{\mathbf{p}}_i)}$$

Direction vector  $\underline{\mathbf{v}}$ ; Wave number vector  $\underline{\mathbf{k}} = \frac{\omega}{c} \cdot \underline{\mathbf{v}}$

- Propagation expressed as function of time and space
- Information is preserved while propagating

⇒ Band-limited signal can be reconstructed over all space and time by either:

- *temporally sampling* at given location in space
- *spatially sampling* at given instant of time



Basis for all aperture and sensor array processing techniques

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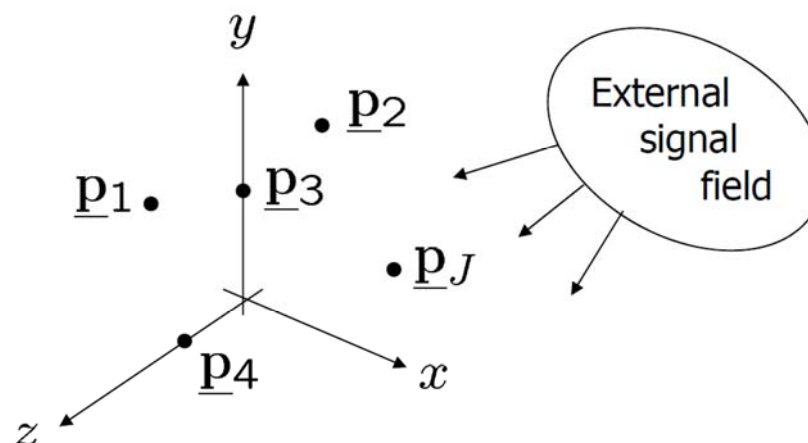
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## Scenarios

### Array geometry

Array can be uniform, nonuniform, linear, circular, ...

Sensors at  $J$  locations  $\underline{p}_i$



*In this course mainly:* **Uniform Linear Array (ULA)**

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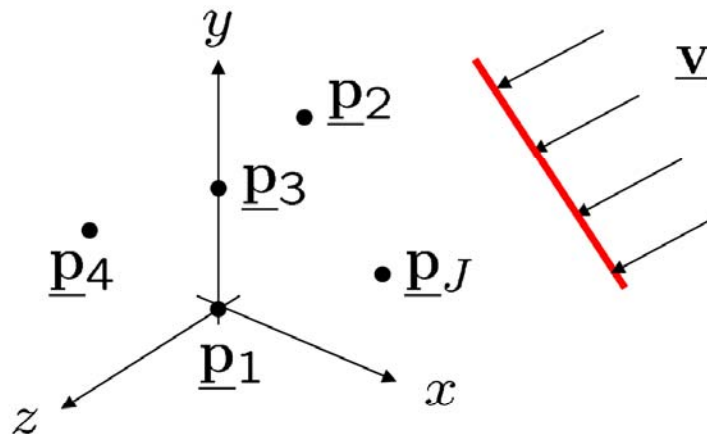


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## Scenarios

*Propagation between two points for **plane wave**:*



*Assume reference  $\underline{p}_1$  :*

$s(t)$  arrives at  $\underline{p}_1$

$\underline{p}_1$  in origin

Analog signal at location  $\underline{p}_i$ :  $s(t - \tau_i) = s(t)e^{-j\omega\tau_i}$

with delay:  $\tau_i = \underline{v}^t \cdot \underline{p}_i / c$  Direction vector  $\underline{v}$

with  $\omega = 2\pi f$  and  $f = \frac{c}{\lambda}$

$\lambda =$  wavelength;  $c =$  speed in medium

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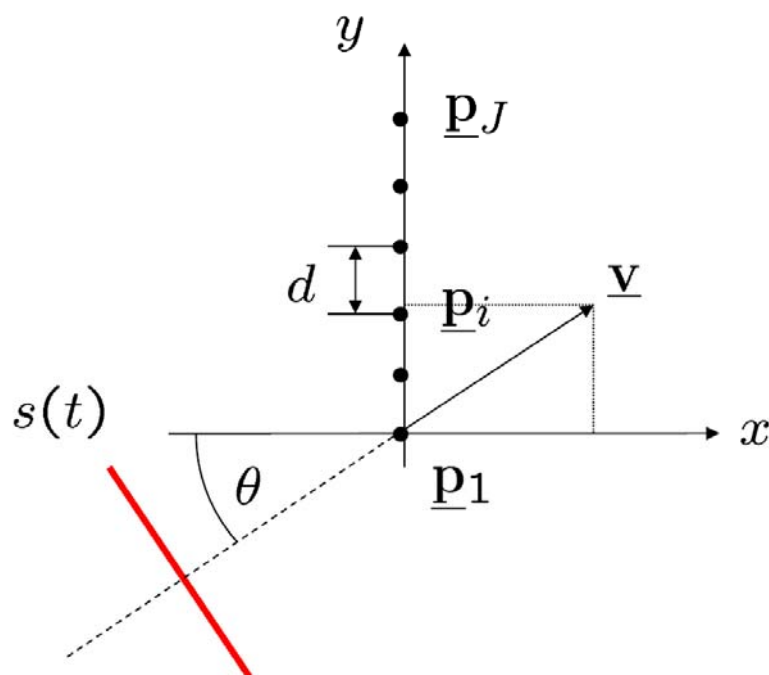
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## Scenarios

Note: Location is 3D quantity

In practice: Direction Of Arrival (DOA) 2D



**ULA:**

If reference  $\underline{p}_1$  at  $(0, 0)$

$$\Rightarrow \underline{p}_i = (0, (i - 1) \cdot d)^t$$

Directional vector:

$$\underline{v} = (\cos(\theta), \sin(\theta))^t$$

$$|\underline{v}| = 1$$

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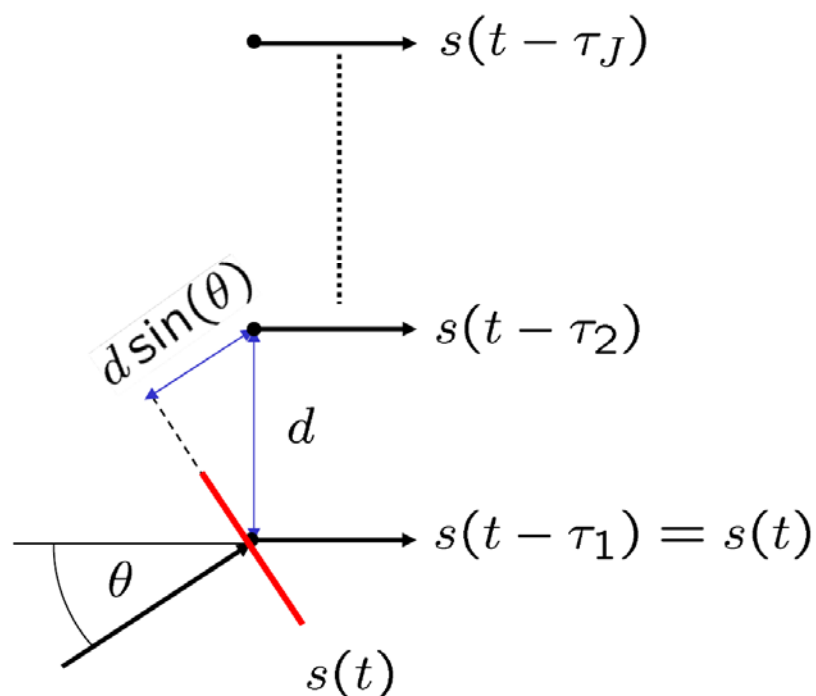
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## Scenarios

### Example: Plane wave, ULA

For far field only one parameter (DOA) characterizing position

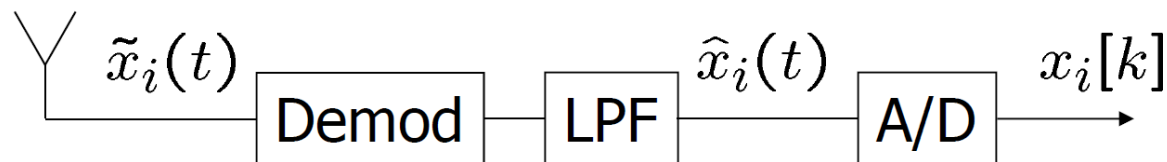


$$\begin{aligned} \Rightarrow \tau_i &= \tau_i(\theta) \\ &= \underline{\mathbf{v}}^t \underline{\mathbf{p}}_i / c \\ &= (i - 1) \frac{d \sin(\theta)}{c} \end{aligned}$$

### ***Discrete-time signal representation***

- Analog sensor signal at  $J$  locations:  $\tilde{x}_i(t)$
- In each sensor  $i = 1 \dots J$  ideal demodulation and LPF takes place to baseband signal  $\hat{x}_i(t)$
- After A/D:

Complex valued discrete-time signal  $x_i[k]$





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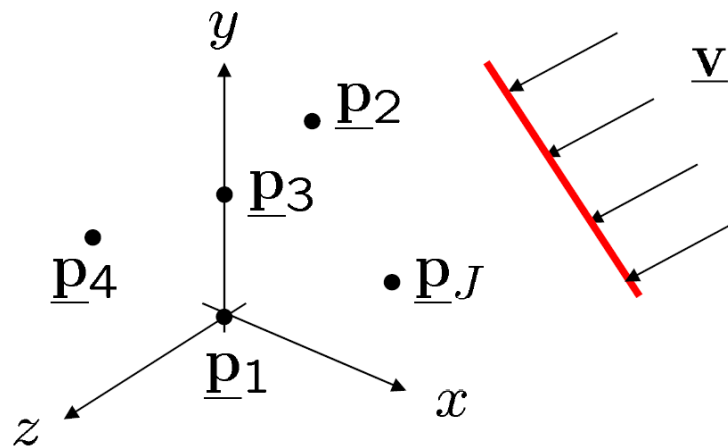


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## Scenarios

*Propagation between two points for **plane wave**:*



*Assume reference  $\underline{p}_1$  :*

$s(t)$  arrives at  $\underline{p}_1$

$\underline{p}_1$  in origin

Analog signal at location  $\underline{p}_i$ :  $s(t - \tau_i) = s(t)e^{-j\omega\tau_i}$

with delay:  $\tau_i = \underline{v}^t \cdot \underline{p}_i / c$

$\Rightarrow$  Discrete-time signal at  $\underline{p}_i$  :  $s[k] \cdot e^{-j\omega\tau_i}$

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## Scenarios

⇒ Discrete-time signal in sensor  $i$  (for ULA):

$$s[k] \cdot e^{-j\omega\tau_i} \equiv s[k] \cdot a_i(\omega, \theta)$$

$$\begin{aligned} \text{with } a_i(\omega, \theta) &= e^{-j\omega\tau_i} \\ &= e^{-j\omega(i-1)\frac{d\sin(\theta)}{c}} \\ &= e^{-j2\pi\frac{c}{\lambda}(i-1)\frac{d\sin(\theta)}{c}} \\ &= e^{-j2\pi(i-1)\frac{d\sin(\theta)}{\lambda}} \end{aligned}$$

*Note:* Usually simplified notation

$$a_i(\omega, \theta) \rightarrow a_i(\theta)$$

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## Scenarios

### *Array signal model*

**Array sensor vector:**  $\underline{\mathbf{x}}[k] = (x_1[k], x_2[k], \dots, x_J[k])^t$

**Noise vector:**  $\underline{\mathbf{n}}[k] = (n_1[k], n_2[k], \dots, n_J[k])^t$

**Steering vector:**  $\underline{\mathbf{a}}(\theta) = (a_1(\theta), a_2(\theta), \dots, a_J(\theta))^t$

with  $a_i(\theta) = e^{-j\omega\tau_i(\theta)}$

*Note:* for ULA:  $a_i(\theta) = e^{-j2\pi(i-1)\frac{d\sin(\theta)}{\lambda}}$

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## Scenarios

**Case: Noisy observation, one source,  $J$  sensors**

for  $i = 1, 2, \dots, J$  :  $x_i[k] = a_i(\theta) \cdot s[k] + n_i[k] \Rightarrow$

$$\underline{\mathbf{x}}[k] = \underline{\mathbf{a}}(\theta) \cdot s[k] + \underline{\mathbf{n}}[k]$$

**Covariance structure:**

$$\mathbf{R}_x = E\{\underline{\mathbf{x}} \cdot \underline{\mathbf{x}}^h\} = \sigma_s^2 \cdot (\underline{\mathbf{a}} \cdot \underline{\mathbf{a}}^h) + \mathbf{R}_n$$

with  $\sigma_s^2 = E\{|s|^2\}$  and  $\mathbf{R}_n = E\{\underline{\mathbf{n}} \cdot \underline{\mathbf{n}}^h\}$

For spatially white noise:  $\mathbf{R}_n = \sigma_n^2 \mathbf{I}$

*Note:* Time indices are skipped for simplicity



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## Scenarios

**Case: Noisy observation,  $P$  sources,  $J$  sensors**

$$\text{for } i = 1, 2, \dots, J : x_i[k] = \sum_{p=1}^P a_i(\theta_p) \cdot s_p[k] + n_i[k] \Rightarrow$$

$$\underline{\mathbf{x}}[k] = \mathbf{A} \cdot \underline{\mathbf{s}}[k] + \underline{\mathbf{n}}[k]$$

$J \times P$  steering matrix  $\mathbf{A} = (\underline{\mathbf{a}}(\theta_1), \underline{\mathbf{a}}(\theta_2), \dots, \underline{\mathbf{a}}(\theta_P))$

$P \times 1$  signal vector  $\underline{\mathbf{s}}[k] = (s_1[k], s_2[k], \dots, s_P[k])^t$

**Covariance structure:**

$$\mathbf{R}_x = E\{\underline{\mathbf{x}} \cdot \underline{\mathbf{x}}^h\} = \mathbf{A} \mathbf{R}_s \mathbf{A}^h + \mathbf{R}_n$$

with  $\mathbf{R}_s = E\{\underline{\mathbf{s}} \cdot \underline{\mathbf{s}}^h\}$  and  $\mathbf{R}_n = E\{\underline{\mathbf{n}} \cdot \underline{\mathbf{n}}^h\} = \sigma_n^2 \mathbf{I}$

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## Scenarios

**General case:**

Noisy observation, desired + undesired signals:

$$\underline{\mathbf{x}}[k] = \underline{\mathbf{x}}_d[k] + \underline{\mathbf{x}}_u[k] + \underline{\mathbf{n}}[k]$$

$\underline{\mathbf{x}}_d[k]$  :  $P$  independent desired sources

$\underline{\mathbf{x}}_u[k]$  :  $Q$  independent undesired sources

$\underline{\mathbf{n}}[k]$  : Spatially white noise

**Covariance structure:**

$$\mathbf{R}_x = E\{\underline{\mathbf{x}} \cdot \underline{\mathbf{x}}^h\} = \mathbf{R}_{x_d} + \mathbf{R}_{x_u} + \mathbf{R}_n$$

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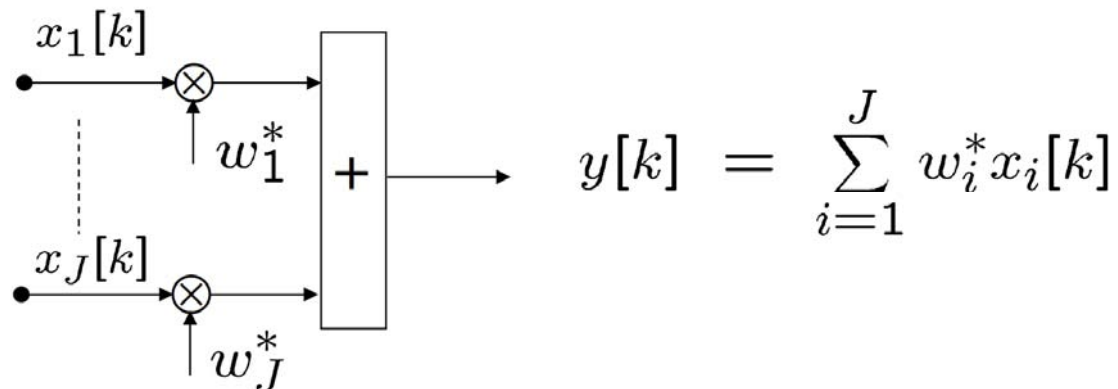


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## Scenarios

Case: Single complex weight for each sensor



Short notation:  $y[k] = \underline{\mathbf{w}}^h \cdot \underline{\mathbf{x}}[k]$

$$\underline{\mathbf{x}}[k] = (x_1[k], \dots, x_J[k])^t$$

$$\underline{\mathbf{w}} = (w_1, \dots, w_J)^t$$

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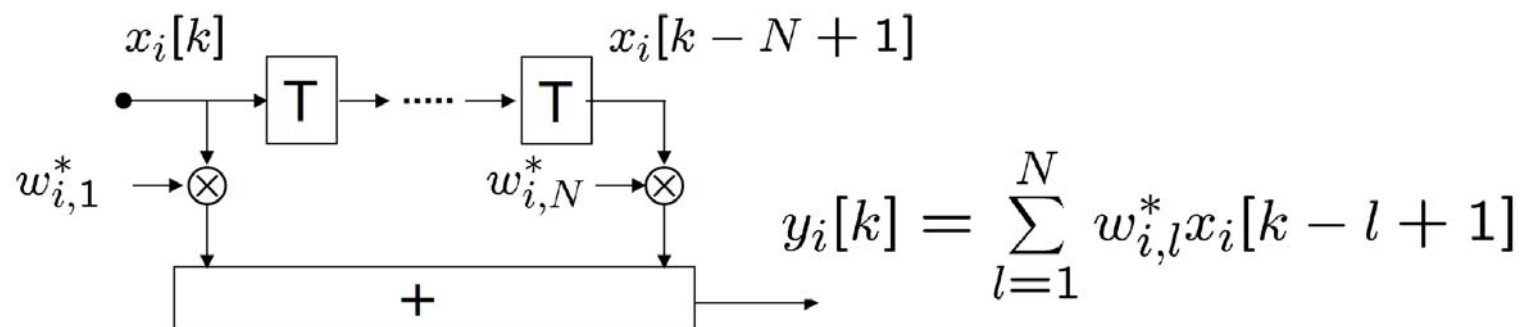


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## Scenarios

Case: FIR filter for each sensor



Short notation for  $i = 1, 2, \dots, J$ :

$$\underline{\mathbf{x}}_i[k] \longrightarrow \boxed{\underline{\mathbf{w}}_i^*} \longrightarrow y_i[k] = \underline{\mathbf{w}}_i^h \cdot \underline{\mathbf{x}}_i[k]$$

$$\underline{\mathbf{x}}_i[k] = (x_i[k], \dots, x_i[k-N+1])^t$$

$$\underline{\mathbf{w}}_i = (w_{i,1}, \dots, w_{i,N})^t$$



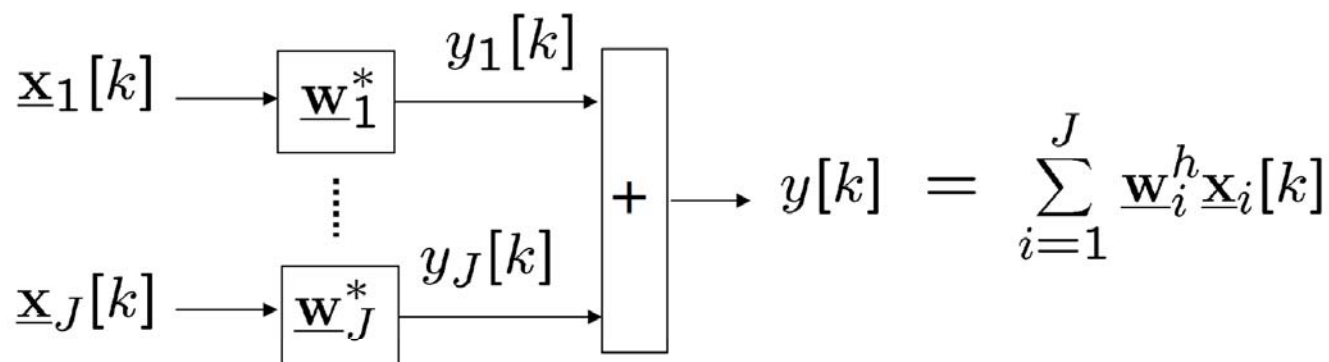
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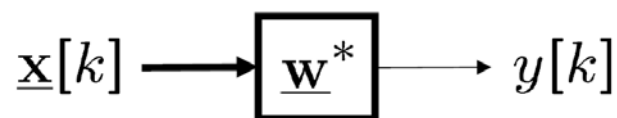
## Scenarios



**Short notation:**  $y[k] = \underline{w}^h \cdot \underline{x}[k]$

$$\underline{x}[k] = (\underline{x}_1[k], \dots, \underline{x}_J[k])^t$$

$$\underline{w} = (\underline{w}_1, \dots, \underline{w}_J)^t$$



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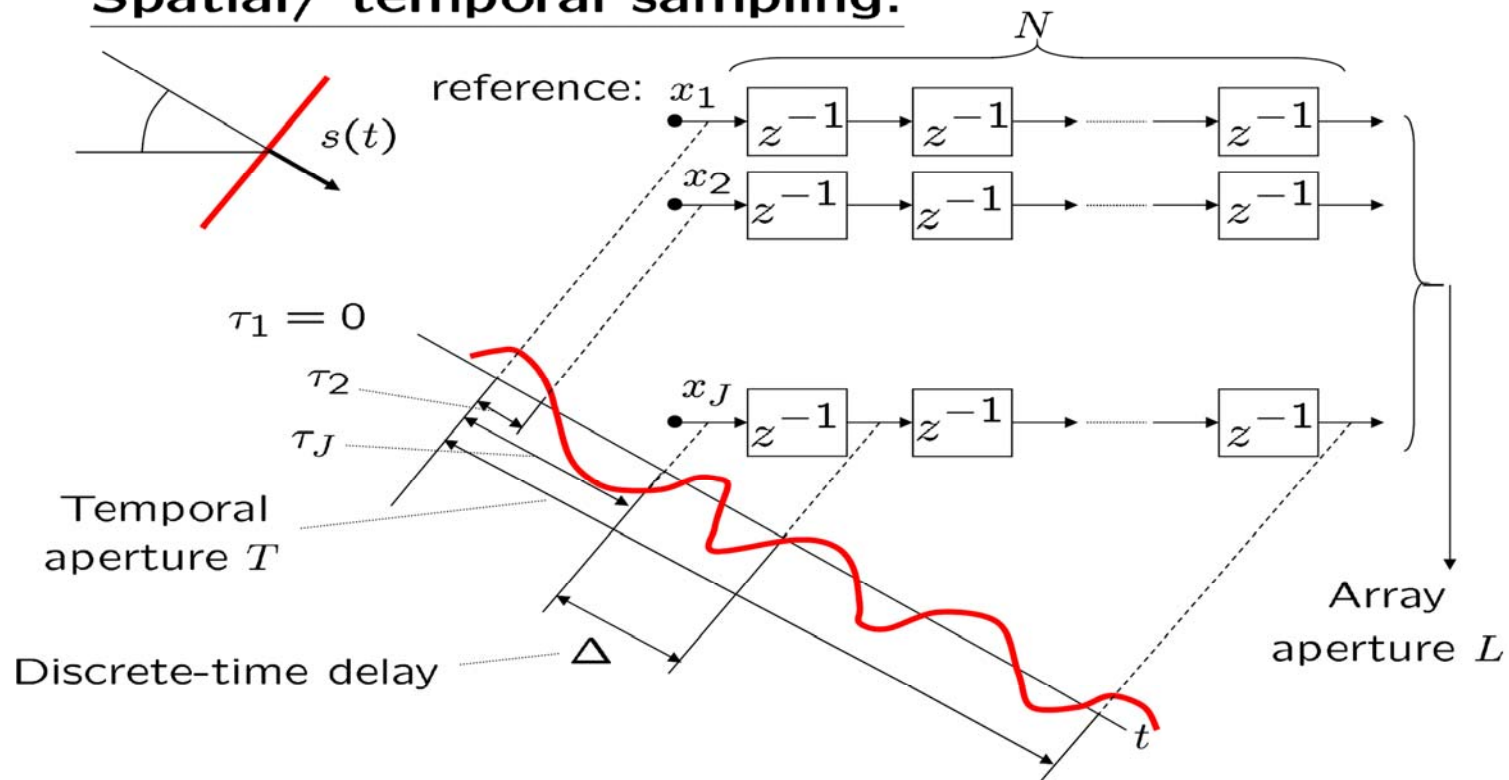


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## Scenarios

Spatial/ temporal sampling:





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## Scenarios

*Notes:*

- Propagating source signal is sampled at  $J \cdot N$  nonuniformly spaced points
- Temporal aperture:  $T(\theta)$
- Array aperture  $L$ : array length in wave length  
For ULA:  $L = J \cdot d$
- FIR provide not simple frequency depending weighting of each channel. Weights effect **both** temporal and spatial response

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## Scenarios

How to cope with broadband signals:

