

Introduction to Array Processing

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Introduction to Array Processing Overview



Contents:

1.	Introduction	1
2.	Spatial Sampling	2
3.	Steering Vector	3
4.	Data Independent Beamformer Design	4
5.	Statistically Optimum Beamformer Design	8
6.	Beamforming and MIMO	9
7.	Broadband beamforming1	0
8.	Summary	1

1.1 Intuitive Beamforming

A far eld wavefront arrives at a linear uniformly spaced sensor array:

due to the direction of arrival (DOA) and nite propagation speed, the wavefront will arrive at di erent sensors with a delay



with appropriate processing (beamforming), the sensor signals can be aligned to create constructive interference at the output x(t);

the above is a simple delay-and-sum beamformer [12, 32, 50, 33].

1.2 Spatial Sampling

For unambiguous spatial sampling, we need to take at least two samples per wavelength of the highest frequency component in the array signals [32];



analogy from temporal sampling (Nyquist): take at least two samples per period (relating to the highest frequency component);

Wavelength and frequency f are related by the propagation speed c in the medium: = $\frac{c}{f}$;



maximum sensor distance

$$d = \frac{\max}{2} = \frac{c}{2f_{\max}}$$

time delay between sensors

$$=\frac{d\sin()}{c}=\frac{\sin()}{2f_{\max}}$$

Spatial and Temporal Sampling

Consider the array signals $x_0(t)$ and $x_1(t)$ due to a source $e^{j(t+0)}$:

$$e^{j(t+0)}$$

 $d \sin(t) = e^{jt}$
 $d \sin(t) = e^{jt}$

sampling with $t = nT_{\rm s}$ leads to

$$x_0[n] = e^{j nT_s}$$
 and $x_1[n] = e^{j (nT_s)}$

with $f_{\max} = \frac{f_s}{2} = \frac{1}{2T_s}$ and normalised angular frequency $= T_s$, $x_0[n] = e^{j-n}$ and $x_1[n] = e^{j-n} e^{-j-\sin(-)}$



1.3 Steering Vector

A narrowband source with normalised angular frequency illuminates an M-element linear array of equi-spaced sensors:

$$\mathbf{x}[n] = \begin{array}{cccc} x_0[n] & & 1 \\ x_1[n] & & e^{j - \sin(-)} \\ \vdots & & \vdots \\ x_{M-1}[n] & & e^{j(M-1) - \sin(-)} \end{array} = e^{j - n} \mathbf{s}$$

the vector **s** characterises the phase shifts of waveform with frequency and DOA measured at the array sensors;

since a narrowband signal e^{j-n} only causes phase shifts rather than delays, constructive interference can be accomplished by a set of complex multipliers rather than processors (t - m), m = 0 - 1 - (M - 1);

beamforming problem: how to select the set of complex coe cients?



1.4 Data Independent Beamformer





Challenge: nd a set of complex multipliers $w_m, m = 0 \ 1 \qquad (M \ 1)$:

to steer the array characteristic towards this source, the output

$$y[n] = [w_0 \ w_1 \ w_{M-1}]e^{j \ n} \qquad \begin{array}{c} e^{j \ \sin()} \\ \vdots \\ e^{j(M-1) \ \sin()} \end{array} = e^{j \ n} \mathbf{w}^{\mathrm{H}} \mathbf{s}$$
should satisfy $y[n] = e^{j \ n}$, leading to $\mathbf{w}^{\mathrm{H}} \mathbf{s} = 1$.

Coe cient Vector

For later convenience and compatibility, the Hermitian transpose operator $^{\rm H}$ is used to denote the coe $\,$ cient vector

$$\mathbf{w}^{\mathrm{H}} = \begin{bmatrix} w_0 & w_1 & \cdots & w_{M-1} \end{bmatrix};$$

as a result, the vector \mathbf{w} hold the complex conjugates of the coe cients,

$$\mathbf{w} = \begin{array}{c} w_0 \\ w_1 \\ \vdots \\ w_{M-1} \end{array};$$

to access the actual unconjugated coe $% \left({{\mathbf{x}}_{i}} \right)$ cients, the beamforming vector ${\mathbf{w}}$ has to be considered

note that

$$\mathbf{w}^{\mathrm{H}}\mathbf{s} = 1$$
 $\mathbf{s}^{\mathrm{H}} \mathbf{w} = 1$



Narrowband Beamforming Single Source

The expression $\mathbf{s}^{\mathrm{H}} \mathbf{w} = 1$ forms a system with one equation and M unknowns Strat

general solution to an underdetermined system Ax = b is the right pseudo-inverse A [10],

$$\mathbf{x} = \mathbf{A} \ \mathbf{b} = \mathbf{A}^{\mathrm{H}} (\mathbf{A} \mathbf{A}^{\mathrm{H}})^{-1} \mathbf{b} ;$$

here:

$$\mathbf{w} = (\mathbf{s}^{\mathrm{H}})$$
 $1 = \mathbf{s}$ $(\mathbf{s}^{\mathrm{H}} \mathbf{s})$ 1 $1 = \frac{\mathbf{s}}{\mathbf{s}}$ $\frac{2}{2} = \frac{1}{M}\mathbf{s}$

the complex conjugation for ${\bf w}~$ inverts and therefore compensates the phase of the steering vector, which could have been foreseen

the formulation via the pseudo-inverse will be powerful for more complicated cases.



Narrowband Beamformer Example

Source parameters: $= \frac{1}{2}$ and = 30; array parameter: M = 5; steering vector (with $\sin() = \frac{1}{4}$):

$$\mathbf{s}^{\mathrm{T}} = [1 \ \mathrm{e}^{-\mathrm{j}\frac{1}{4}} \qquad \mathrm{e}^{-\mathrm{j}\frac{4}{4}}]$$

coe cient vector is given by $\mathbf{w} = (\mathbf{s}^{H})$; numerical solution in Matlab; Omega=1/4; theta = pi/6; M=5; s = exp(-sqrt(-1)*Omega*sin(theta)*(0:(M-1)')); w = pinv(s');angle([s conj(w)])/pi yields: -0.000000.00000 -0.25000 0.25000 -0.50000 0.50000 -0.75000 0.75000 -1.000001.00000



Beam Pattern I

- The beamformer has a unit gain towards a source with frequency Ω and DoA θ; what is its gain response towards other angles of arrival?
- \blacktriangleright the beam pattern measures the response of a beamformer by sweeping the angle ψ of a source with frequency Ω

$$g(\Omega,\psi) = \mathbf{w}^{\mathrm{H}} \mathbf{s}_{\Omega,\psi}$$

beam pattern for the previous example:





Beam Pattern II

• Beam patterns for $\Omega = \frac{\pi}{2}$ and $\vartheta = 30^{\circ}$ with variable M:





analogous characteristic in the time domain: increased temporal support leads to higher frequency resolution.



Interference

Many scenarios contain a source of interest and a number of interferers: signal of interest:

0 0 two interferers:

1 1, 2 2



we would like to control the beampattern to place spatial nulls in the directions of the interfering sources;

Problem formulation and solution :

Narrowband BF Example Multiple Sources

The signal of interest illuminates an M = 5 element array at a frequency $_0 = \frac{1}{2}$ with a DoA $_0 = 30$ two interferers at $_1 = _2 = _0$ are present with DoA $_1 = ~45$ and $_2 = 60$ results via right pseudo-inverse of steering vectors

S ₀₀	${f s}_{1\ 1}$	\mathbf{s}_{2}	w	w
0.00	0.00	0.00	-42.81	0.3172
45.00	63.64	-77.94	-105.01	0.3004
90.00	127.28	-155.89	-90.00	0.2343
135.00	-169.08	126.17	-74.99	0.3004
180.00	-105.44	48.23	-137.19	0.3172

the angle of ${\bf w}$ is no longer intuitive; also note that the coe $% {\bf v}$ cients in ${\bf w}$ no longer have the same modulus

amongst a manifold of solutions, the right pseudo-inverse provides the minimum norm solution.



Multiple Source Example — Beampattern



the pseudo-inverse is the minimum norm solution, keeping the general gain response as low as possible;

the minimum norm property protects against spatially white noise.

Beamforming Example Variable Interferer I

M=5 sensors, source of interest towards $_{0}=30$, interferer variable:



Beamforming Example Variable Interferer II

M=5 sensors, SOI $_{-0}=30$, one $\,$ xed and one variable interferer:





Beamforming Example Variable Interferer III M = 5 sensors, SOI $_0 = 30$, one variable and three xed interferers:



Data Independent Beamforming



Previous beamformer designs were based on the knowledge of DoA of the signal of interest and of interfering sources;

remaining degrees of freedom are invested to suppress spatially white noise;

using the analogy between spatial and temporal processing, classical lter design techniques can be invoked to design arrays with a bandpass-type angular response;

beamformers based on source parameters (frequency, DoA) rather the actual received waveforms are termed data independent beamformers;

this is in contrast to statistically optimum beamformers, which take the received signal statistics into account.

1.8 Statistically Optimum Beamforming



Statistically optimum beamformer minimise e.g. the signal power of the beamformer output, y[n]; to avoid the trivial solution $\mathbf{w} = \mathbf{0}$, the signal of interest needs to be protected by constraints;

this results in e.g. the following constrained optimisation problem

$$\min_{\mathbf{w}} \mathcal{E} \quad y[n]^2 \qquad \text{subject to} \qquad \mathbf{s}^{\mathsf{H}} \quad \mathbf{w} = 1 ;$$

the solution to this speci c statistically optimum beamformer is known as the minimum variance distortionless response (MVDR) [33].



MVDR Beamformer

Solving the MVDR problem: minimise the power of $y[n] = \mathbf{w}^{H}\mathbf{x}$ subject to the contraint $\mathbf{w}^{H}\mathbf{s}_{0} = 1$; Formulation using a Lagrange multiplier :



the solution ${f w}={f R}_{xx}^{-1}{f s}_{-0}$ is inserted into the constraint equation to determine :

$$\mathbf{s}_{0}^{\mathrm{H}}{}_{0}\mathbf{R}_{xx}^{\mathrm{H}}\mathbf{s}_{0}{}_{0}=1$$

therefore

$$\mathbf{w}_{\mathrm{MVDR}} = \mathbf{s}_{0}^{\mathrm{H}} \mathbf{R}_{xx}^{-1} \mathbf{s}_{0}^{-1} \mathbf{R}_{xx}^{-1} \mathbf{s}_{0}^{-1}$$

this statistically optimum beamformer has various other names, e.g. Capon beamformer [7, 32].



MVDR Beamformer Simple Case



In the case of spatially white noise input, such that

$$\mathbf{R}_{xx} = \begin{array}{c} 2\\ xx \mathbf{I} \end{array} \qquad \qquad \mathbf{R}_{xx}^{-1} = \begin{array}{c} 2\\ xx \mathbf{I} \end{array}$$

the MVDR solution reduces to

$$\mathbf{w}_{\rm MVDR} = \frac{\mathbf{s}_{0}}{\mathbf{s}_{0}} \frac{1}{2} = \frac{\mathbf{s}_{0}}{M} ;$$

this is identical to the data independent beamformer in the absence of interference (i.e. no spatially structured noise);

Generalised Sidelobe Canceller (GSC)

The generalised sidelobe canceller is a speci c implementation of the MVDR beamformer; it transforms the constrained MVDR problem into an unconstrained optimisation problem;

a rst guess at the solution is performed by the quiescent beamformer $\mathbf{w}_{\rm q}$, which is identical to the previously de ned data independent beamformer, obtained by inverting the constraint equation

the quiescent beamformer eliminates interferers captured by C and f, but passes the signal of interest, any interferers unaccounted for in the constraints, and spatially distributed noise.





GSC Idea

GSE idea: produce array signals that are free of any contribution from the signal of interest, and use the resulting signal vector $\mathbf{u}[n]$ to eliminate remaining interference from the quiescent output:



the blocking matrix ${\bf B}$ eliminates the signal of interest and any interferers captured by the constraints;

the vector $\mathbf{w}_{\rm a}$ will be based on the statistics of $\mathbf{u}[n]$ and d[n] to minimise the beamformer output variance $\mathcal{E}-e[n]^{-2}$.



GSC Blocking Matrix

In order to project away from the constraints,

B
$$\mathbf{C} = \mathbf{B} \quad \mathbf{s}_{0 \quad 0} \quad \mathbf{s}_{1 \quad 1} \qquad \mathbf{s}_{r \quad 1 \quad r \quad 1} = \mathbf{0}$$

assuming that the r constraints are linearly independent, the singular value decomposition of the constraint matrix yields

$$\mathbf{B} \quad \mathbf{U}_0 \quad \mathbf{U}_0 \qquad \begin{array}{c|c} \mathbf{0} & & \\ & \ddots & & \mathbf{0} \\ \hline & & r \quad 1 \\ \hline & \mathbf{0} & & \mathbf{0} \end{array} \quad \mathbf{V}^{\mathrm{H}} = \mathbf{0}$$

the matrix $\mathbf{U}_0 = \mathbb{C}^{M-(M-r)}$ spans the nullspace of \mathbf{C}^{H} , and $\mathbf{B} = (\mathbf{U}_0)^{\mathrm{H}} = \mathbb{C}^{(M-r)-M}$

has the required property, as $(\mathbf{U}_0~)^H~~\mathbf{U}_0~~\mathbf{U}_0~~=[\mathbf{0}~\mathbf{I}]~~=\mathbf{0}.$



GSC Unconstrained Optimisation

The beamforming vector \mathbf{w}_a is adjusted to minimise the output power; the MMSE or Wiener solution is given by

$$\mathbf{w}_{\mathrm{a}} = \mathbf{R}_{uu}^{-1} \ \mathbf{p} = rac{\mathbf{B}\mathbf{R}_{xx}(\mathbf{C}^{\mathrm{H}}) \ \mathbf{f}}{\mathbf{B}\mathbf{R}_{xx}\mathbf{B}^{\mathrm{H}}}$$

with the covariance matrix

$$\mathbf{R}_{uu} = \mathcal{E} \ \mathbf{u}[n] \ \mathbf{u}^{\mathrm{H}}[n] = \mathbf{B} \ \mathcal{E} \ \mathbf{x}[n] \ \mathbf{x}^{\mathrm{H}}[n] \quad \mathbf{B}^{\mathrm{H}} = \mathbf{B} \mathbf{R}_{xx} \mathbf{B}^{\mathrm{H}}$$

and the cross-correlation vector

$$\mathbf{p} = \mathcal{E} \mathbf{u}[n] d [n] = \mathbf{B} \mathbf{R}_{xx} \mathbf{w}_{q}$$

iterative optimisation schemes, such as the least mean squares (LMS) algorithm [12, 50] may be used to approximate the MMSE solution.



1.9 Beamforming and MIMO Processing

Assume a transmission scenario with an M-element transmit (Tx) antenna array and an N-element receive (Rx) array;



in the absence of scatterers and any attenuation, the far eld transmission from the transmit antenna is characterised by a steering vector $s_{T_x}^H$;

the incoming waveform at the Rx device is described by another steering vector \mathbf{s}_{Rx} ; the overall MIMO system between a Tx vector $\mathbf{x} \quad \mathbb{C}^M$ and an Rx vector $\mathbf{y} \quad \mathbb{C}^N$ is described as

$$\mathbf{y} = \mathbf{s}_{\mathrm{Rx}} \ \mathbf{s}_{\mathrm{Tx}}^{\mathrm{H}} \ \mathbf{x} = \ \mathbf{Hx}$$

the MIMO system matrix $\mathbf{H}=\mathbf{s}_{\mathrm{Rx}}~~\mathbf{s}_{\mathrm{Tx}}^{\mathrm{H}}$ is rank one only.



The far eld assumption is convenient for beamforming, but leads to a rank one MIMO system matrix which is incompatible with the desire to extract multiple independent subchannels or with to achieve diversity;

rich scattering in connection with MIMO usually implies multiple re ections of signals;

together with a su ciently large antenna spacing means that the far eld assumption is invalid and the MIMO system matrix is not rank de cient;

some suggestions of $\mbox{ su ciently large spacing imply an antenna element distance of <math display="inline">d>10$;

recall spatial sampling requires $d < \frac{1}{2}$!

Beamforming with Spatial Aliasing



For a exible spatial sampling with d = 0, $0 < \mathbb{R}$, the steering vector for a waveform with normalised angular frequency and DoA is

$$\mathbf{y} = e^{j n} \qquad \begin{array}{c} 1 \\ e^{j2} & \sin() \\ \vdots \\ e^{j2} & (M-1) & \sin() \end{array} = \mathbf{s}_2 \qquad e^j$$

inspecting \mathbf{s}_2 the steering vector is aliased to a di erent frequency 2; although the correct frequency can be retrieved unambigiously from temporal sampling of any array element, at various di erent angles could provide the same steering vector \mathbf{s}_2 ;

the array performs spatial undersampling, resulting in spatial aliasing.

Spatial Undersampling Example

- Beamforming parameters: signal of interest with $\Omega = \frac{\pi}{2}$, direction of arrival $\vartheta = 30^{\circ}$, M = 32 array elements;
- data independent beamformer design with correct spatial sampling $(d = \lambda/2)$ and incorrect spatial sampling $(d = 10\lambda)$:



MIMO systems perform beamforming, but may dissipate energy into aliased directions.



1.10 Broadband MVDR Beamformer

Each sensor is followed by a tap delay line of dimension L, giving a total of ML coe cients in a vector $\mathbf{v} \in \mathbb{C}^{ML}$ [5]



Broadband MVDR Beamformer Constraints



A larger input vector $\mathbf{x}_n \quad \mathbb{C}^{ML}$ is generated, also including lags;

the general approach is similar to the narrowband system, minimising the power of $e[n]={\bf v}^{\rm H}{\bf x}_n;$

however, we require several constraint equations to protect the signal of interest, e.g.

$$\mathbf{C} = \begin{bmatrix} \mathbf{s}(s \ 0) & \mathbf{s}(s \ 1) & \mathbf{s}(s \ L \ 1) \end{bmatrix}$$
(1)

these L constraints pin down the response to unit gain at L separate points in frequency:

$$\mathbf{C}^{\mathrm{H}}\mathbf{v} = \mathbf{1} ; \qquad (2)$$

generally $\mathbf{C} = \mathbb{C}^{ML-L}$, but simpli cations can be applied if the look direction is towards broadside.

Broadband Generalised Sidelobe Canceller

A quiescent beamformer $\mathbf{v}_q=~\mathbf{C}^H~~\mathbf{1}~~\mathbb{C}^{ML}$ picks the signal of interest;



the quiescent beamformer is optimal for AWGN but generally passes structured interference;

the output of the blocking matrix **B** contains interference only, which requires $[\mathbf{BC}]$ to be unitary; hence **B** \mathbb{C}^{ML} $(M^{-1})^{L}$;

an adaptive noise canceller \mathbf{v}_{a} = $\mathbb{C}^{(M-1)L}$ aims to remove the residual interference:



note: all dimensions are determined by $\ M \ L$.

Broadband Beamformer Example

We assume a signal of interest from = 30; three interferers with angles $_{i}$ 40 10 80 active over the frequency range = 2 [0 1; 0 45] at signal to interference ratio of -40 dB;



M=8 element linear uniform array is also corrupted by spatially and temporally white additive Gaussian noise at 20 dB SNR;

tap-delay-line length: L = 150;

cost per iteration: approx. 2 MMACs (standard), can be reduced to 10 kMACs when

e ciently implemented.

Broadband Quiescent Beamformer

Directivity pattern of quiescent standard broadband beamformer:





Optimised Broadband Beamformer

Directivity pattern of the broadband beamformer:





1.11 Summary

Spatial sampling by an array of sensors (e.g. antenna elements) has been explored;

the spatial data window of a narrowband source is characterised by the steering vector; appropriate data independent beamformers can be designed based on the steering vectors of desired sources and interferers;

statistically optimum beamformers are based on the signal statistics;

a speci c statistically optimum beamformer, the generalised sidelobe canceller, has been reviewed it uses signal statistics to improve the performance of a data independent beamformer derived from the constraint equations;

some similarities and di erences between beamforming and MIMO systems have been highlighted;

broadband beamforming requires the inclusion of tap delay lines.





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General wideband beamforming: [19];
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time domain adaptive broadband beamforming: [5, 6, 11, 14, 23, 29, 33];

discrete Fourier transform domain processing: [17, 30, 44]

subband domain beamforming [21, 34, 46, 47, 48, 49, 44];

frequency-invariant broadband beamforming [18, 22, 23, 38];

polynomial matrix-based

beamforming [1, 2, 3, 4, 8, 9, 15, 16, 24, 25, 28, 27, 26, 31, 35, 36, 40, 41, 42, 45, 37, 43].

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