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Index

- acoustic echo cancellation, *see* acoustic echo canceller
- acoustic echo canceller, 1, 133, 133–161, 163, 164, 205
 - challenges, 135–137
 - computational complexity, 182
 - monophonic, 133, 135–136
 - multi-channel, 1, 133–137
 - nonlinear, 135
 - realization, 1, 179–181
- acoustic echoes, 1, 45, 48, 133–161, 205, 207
- acoustic environment, 1, 2, 4–6, 24
 - car, 1, 12, 23, 120, 124, 126, 132, 191, 192, 196, 198, 199, 208, 226–227
 - home, 1, 2
 - multimedia room, 11, 12, 112, 118, 119, 121, 123, 124, 126, 178, 187, 189, 191, 200–202, 220, 222, 227–228
 - office, 1, 2, 208
 - reverberant, 8, 10, 23, 42, 43, 99, 101, 106, 107, 136, 163, 206
 - time-invariant, 10
 - time-varying, 8, 10, 26, 42, 94, 99, 100, 131, 132, 141, 142, 154, 158, 163
- acoustic human/machine front-end, 1–3, 44, 45, 133, 163, 205, 208
- adaptation algorithm, 26, 39, 113, 115, 131, 164, 167, 168, 170
- adaptation control, 136, 155, 160, 164, 173, 175, 179–181, 207
 - computational complexity, 183
 - experimental results, 185–188, 191–193
 - realization, 176–177, 183
 - strategy, 127–130, 206
- adaptation problems, 3, 125–127, 163, 170
- adaptive filter, 7, 26, 39, 113, 115, 135, 136, 138, 139, 141, 142, 164
 - constrained, 164, 169–170
 - multi-channel DFT-domain, 39, 137, 163–170
 - speed of convergence, 39, 39, 91, 113, 115, 131, 135, 136, 139, 141, 142, 153, 164, 169, 175, 179–181, 202
 - tracking capability, 39, 115, 128, 131–133, 135, 141–143, 154, 160, 165, 167, 170, 185, 188, 207
- AEC, *see* acoustic echo canceller
- AEC first, 137, 142, 143, 160, 177, 207
 - computational complexity, 182
 - experimental results, 147, 156–158
 - properties, 138–141, 158–161
 - realization, 179–181
- array, *g*
 - aperture, 2, 3, 54, 60, 87, 93, 94, 152
 - broadside, 50, 53, 54, 58, 61–64
 - endfire, 50, 54, 61
 - harmonically nested, 62
 - imperfections, 2, 55, 80, 82, 86, 97, 99
 - uniform, 9, 17, 22, 53, 55, 61, 62, 64
- array gain, 56, 56–58, 60, 128
- array pattern, 53, 53, 171

- array sensitivity, 55–56, 58, 82, 97
- ASR, *see* speech recognizer
- automatic speech recognition, *see* speech recognizer
- beamformer, 8
 - challenges, 1–2
 - complementary fixed, 177, 210, 212, 216
 - data-dependent, 1, 41, 50, 66–97, 205–206, 209
 - data-independent, 41, 50, 63–66, 96
 - eigenvector, 91–93
 - fixed, 49, 66, 102, 138, 145, 176, 210, 216
 - response, 51–52
 - sensor weights, 49
 - shading coefficients, *see* beamformer, sensor weights
 - uniformly weighted, 52, 53, 54, 58, 60–66, 193, 212, 214, 219
 - vector-space interpretation, 49
 - weight vector, 49
- beamformer first, 137, 141–143, 160, 179
- beampattern, *see* array pattern
- beamsteering, 50, 52, 54, 58, 60, 61
- blocking matrix, 89, 91, 102–106, 108–110, 206
 - adaptation, 127–129, 176–177, 206
 - adaptive realization, 172–173
 - analysis, 110–113
 - computational complexity, 184
 - experimental results, 122–127, 185–193
- cancellation of the desired signal, 2, 38, 58–59, 79, 93, 95, 97, 105, 113, 117
 - by the blocking matrix, 116, 122–127
- combination of beamforming and acoustic echo cancellation, 3, 133–161, 163, 179–181, 206–207
- constraint matrix, 78, 100
- constraint vector, 78, 100
- constraints
 - derivative, 79
 - directional, 79
 - eigenvector, 79
 - quiescent pattern, 79
 - spatial, 79–80, 100
 - spatio-temporal, 99, 100–101, 106
- coordinate system, 6
- correlation function
 - estimation, 18–19, 76, 168, 169, 209–223
 - examples, 20–23
 - spatio-spectral, 14
 - spatio-temporal, 13
- correlation function coefficient, 13
- correlation matrix
 - examples, 20–23
 - sample spatio-temporal, 29
 - spatial, 17
 - spatio-spectral, 17
 - spatio-temporal, 16–17
- data space, 33
- dereverberation, 42, 94, 105, 208
- desired signal, 43–44
- diagonal loading, 82, 86, 92, 130, 175
- diffuse, *see* wave field, diffuse
- directivity, 55, 57
- directivity index, 55, 63
- discrete-time Fourier transform, 10
- double-talk, 2, 125–127, 154, 176, 206
 - detector, 191, 208, 216–217, 219
- DFTFT, *see* discrete-time Fourier transform
- echo return loss enhancement, 135, 135, 137–138
- eigenvalue decomposition, 31
- EIR, *see* power ratio of acoustic echoes to local interference
- equalization, *see* inverse modelling
- ERLE, *see* echo return loss enhancement
- estimation space, 33
- far field, 8
- fast Fourier transform, 167, 182
- FFT, *see* fast Fourier transform
- Fourier integral, 8
- fractional delay filter, 50
- Frost beamformer, 78
- GEIC, 152–158, 160, 181, 207
 - experimental results, 156–158
- generalized sidelobe canceller, 2, 42, 87–91, 97, 143–158

- DTFT-domain description, 90–91
- vector-space interpretation, 88
- geometric gain, *see* directivity
- grating lobe, 61
- GSAEC, 145–152, 160, 179–181, 207
 - analysis, 147
 - experimental results, 145–147, 152
- GSC, *see* generalized sidelobe canceller
- HTK, 193–194
- interference, 43, 45–46
 - correlated, 93–94
 - local, 45–46
- interference cancellation, 26, 38
- interference canceller, 90, 106–107, 206
 - analysis, 113–115
 - computational complexity, 184
 - DTFT-domain description, 91
 - experimental results, 118–122, 185–193
 - optimization criterion, 89
 - optimum weight vector, 89, 107
 - realization, 115, 173–175
- interference rejection, 60
 - by the blocking matrix, 116–117, 125–127
- inverse modeling, 26, 36–38, 110–115
- LCLSE beamformer, *see* linearly-constrained least-squares error beamformer
- LCMV beamformer, *see* linearly-constrained minimum variance beamformer
- least mean-squares algorithm, 39, 163, 184
- least-squares error beamformer, 42, 67–72, 75–76, 95, 96
 - cost function, 69
 - eigenspace interpretation, 69–72
 - optimum weight vector, 69
- least-squares error optimization, 25–33, 38–39
 - normal equation, 29
 - optimization criterion, 28
 - solution, 32
 - vector-space interpretation, 33
- least-squares linear filter, 26
- linearly-constrained least-squares error
 - beamformer, 67, 77–91, 95, 96
 - eigenspace interpretation, 81
 - optimization criterion, 77
 - optimum weight vector, 78, 101
 - relation with LSE beamforming, 84–85
 - robustness improvement, 82–84, 113
 - vector-space interpretation, 78
- linearly-constrained minimum variance
 - beamformer, 66, 84–85, 95, 96, 209
- LMS algorithm, *see* least mean-squares algorithm
- LSE beamformer, *see* least-squares error beamformer
- LSE optimization, *see* least-squares error optimization
- magnitude squared coherence, 15
- mainlobe, 54, 61, 65, 66, 102, 176
- matrix inversion lemma, 69
- mean, 13
- method of least-squares, 25
- microphone, *see* sensor
- microphone array, *see* array
- MIMO system, *see* multiple-input multiple output system
- minimum mean-squared error beamformer, 66, 72–76, 95, 96
 - cost function, 72
 - interpretation, 72–75
 - optimum weight vector, 72
- minimum mean-squared error optimization, 26, 33–35, 39
 - normal equation, 34
 - optimization criterion, 34
 - optimum filter weights, 34
- minimum statistics, 177, 209, 217
- minimum-variance distortionless response beamformer, 73, 85, 86
- MINT, *see* multiple-input multiple output inverse theorem
- mismatched conditions, *see* array, imperfections
- MMSE beamformer, *see* minimum mean-squared error beamformer
- MMSE optimization, *see* minimum mean-squared error optimization

- multiple-input multiple output inverse theorem, 36–37, 111, 112, 114, 115
- multiple-input multiple-output system, 25, 164
- MVDR beamformer, *see* minimum-variance distortionless response beamformer
- narrowband assumption, 19–20, 42, 57, 66
- near field, 8
- near-field/far-field reciprocity, 51
- noise
 - cocktail-party, 12
 - sensor, 22, 46
- nullspace, 31
- optimum linear filtering, 25–39
 - applications, 35–38
- overlap-save method, 164, 165
- peak-to-zero distance, 54, 65
- performance measure, 7
 - data-dependent, 56–60, 116–117
 - data-independent, 53–56
- periodogram, 18, 215
- post-filter, 73, 76, 96
- power pattern, 53
 - MMSE beamformer, 74
 - MVDR beamformer, 86
 - uniform sensor weights, 54, 55, 61, 62
 - windowing functions, 64, 65
- power ratio of acoustic echoes to local interference, 140
- principle of orthogonality, 33, 68, 103, 104, 108
- principle of superposition, 8
- principle-component beamformer, 92
- probability density function, 13
- propagation delays, inter-sensor, 20
- propagation delays, intersensor, 44, 50, 53
- propagation speed, *see* sound velocity
- pseudoinverse, 32
- psychoacoustic masking, 76
- quiescent weight vector, 88, 102, 112, 171
- random field, 5, 13
 - spatio-temporal, *see* spatio-temporal random field
- real-time, 26, 39, 138, 181, 193
- recursive least-squares algorithm, 39, 136, 167, 175
- reference path, 87, 147, 158, 172, 173
- regularization, 169
- relative mismatch, *see* system error
- relative sidelobe level, 54, 64
- reverberation time, 2, 12, 196, 205
- RGSC, *see* robust generalized sidelobe canceller
- RLS algorithm, *see* recursive least-squares algorithm
- robust generalized sidelobe canceller, 3, 97, 99–132, 206
 - analysis, 110–115
 - computational complexity, 181–184
 - DTFT-domain description, 107–110
 - experimental results, 118–127, 185–203
 - realization, 170–177
 - relation with the SC, 110
- robustness, 2, 38, 66, 79, 82–84, 86, 94, 97, 99, 100, 102, 112, 113, 125, 127, 132, 163, 164, 174, 193
- room impulse response, 5, 12, 99, 111, 122, 181
- room transfer function, 11
- sampling
 - spatial, 9
 - temporal, 10
- second-order statistics, 13, 44, 96, 209
- sensor
 - calibration, 66, 99
 - directional, 9
 - mismatch, 2, 66, 82, 99
 - omnidirectional, 9
 - sensor array, *see* array
 - sensor noise, *see* noise, sensor
 - sensor placement, 60–63, 95
 - sensor signals, 47–48
- sidelobe cancelling path, 89, 102, 117, 125, 145, 152, 170–172, 176
- signal model, 43–48
- signal-to-interference-plus-noise ratio, 2, 56
 - biased estimation, 177, 210–213, 223

- unbiased estimation, 76, 209
- singular value, 31
- singular value decomposition, 31, 30–32
- singular vector, 31
- SINR, *see* signal-to-interference-plus-noise ratio
- sound pressure, 6, 9, 43
- sound velocity, 6, 12
- source localization, 51
- space-time filter, 5, 8
- space-time signal, 5–24
 - deterministic model, 5
 - monochromatic, 6–8
 - short-time stationary, 18
 - wideband, 8
- spatial aliasing, 60–63
- spatial coherence function, 14
- spatial coherence matrix, 17
 - diffuse wave field, 22
 - estimation, 213–215
 - free field, 21
- spatio-temporal covariance function, 13
- spatio-temporal filtering, *see* beamformer
- spatio-temporal impulse responses, 8
- spatio-temporal random field, *see* space-time random field
- spatio-temporal random fields, 13, 13–23
- spatio-temporal signal, *see* space-time signal
- speech enhancement, 1, 24, 93, 163, 193
- speech recognizer, 1, 193–195
 - experimental results, 195–202
 - statistical model, 5
- steering vector, 20, 52
- stochastic process, 5
- SVD, *see* singular value decomposition
- system error, 36, 135, 144
- system identification, 26, 35–36, 38, 110–115, 133, 136
- system misalignment, *see* system error
- TIDigits database, 194
- tracking region, 44
- variance, 13
- wave equation, 6
- wave field
 - diffuse, 2, 22, 43, 45, 163
 - monochromatic plane wave, 7, 20–21, 51
 - monochromatic spherical wave, 7
 - multi-path propagation, 10, 42, 93
 - point source, 7, 10, 21, 45
 - propagation, 5–12
 - spatially and temporally white, 21, 57
 - spherically isotropic, *see* wave field, diffuse
- wavenumber vector, 7
- wavenumber-frequency response, *see* beamformer response
- white noise gain, 57–58
- Wiener filter, 26, 39
- word accuracy, 193, 195–203

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