

Text-to-Speech Synthesis

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Contents

	<i>Foreword</i>	<i>page xxiii</i>
	<i>Preface</i>	<i>xxvii</i>
1	Introduction	1
	1.1 What are text-to-speech systems for?	2
	1.2 What should the goals of text-to-speech system development be?	3
	1.3 The engineering approach	4
	1.4 Overview of the book	5
	1.4.1 Viewpoints within the book	5
	1.4.2 Readers' backgrounds	6
	1.4.3 Background and specialist sections	7
2	Communication and language	8
	2.1 Types of communication	8
	2.1.1 Affective communication	8
	2.1.2 Iconic communication	9
	2.1.3 Symbolic communication	10
	2.1.4 Combinations of symbols	11
	2.1.5 Meaning, form and signal	12
	2.2 Human communication	13
	2.2.1 Verbal communication	14
	2.2.2 Linguistic levels	16
	2.2.3 Affective prosody	17
	2.2.4 Augmentative prosody	18
	2.3 Communication processes	18
	2.3.1 Communication factors	19
	2.3.2 Generation	20
	2.3.3 Encoding	21
	2.3.4 Decoding	22
	2.3.5 Understanding	22
	2.4 Discussion	23
	2.5 Summary	24

3	The text-to-speech problem	26
3.1	Speech and writing	26
3.1.1	Physical nature	27
3.1.2	Spoken form and written form	28
3.1.3	Use	29
3.1.4	Prosodic and verbal content	30
3.1.5	Component balance	31
3.1.6	Non-linguistic content	32
3.1.7	Semiotic systems	33
3.1.8	Writing systems	34
3.2	Reading aloud	35
3.2.1	Reading silently and reading aloud	35
3.2.2	Prosody in reading aloud	36
3.2.3	Verbal content and style in reading aloud	37
3.3	Text-to-speech system organisation	37
3.3.1	The common-form model	38
3.3.2	Other models	39
3.3.3	Comparison	40
3.4	Systems	41
3.4.1	A simple text-to-speech system	41
3.4.2	Concept to speech	42
3.4.3	Canned speech and limited-domain synthesis	43
3.5	Key problems in text-to-speech	44
3.5.1	Text classification with respect to semiotic systems	44
3.5.2	Decoding natural-language text	46
3.5.3	Naturalness	47
3.5.4	Intelligibility: encoding the message in signal	48
3.5.5	Auxiliary generation for prosody	49
3.5.6	Adapting the system to the situation	50
3.6	Summary	50
4	Text segmentation and organisation	52
4.1	Overview of the problem	52
4.2	Words and sentences	53
4.2.1	What is a word?	53
4.2.2	Defining words in text-to-speech	55
4.2.3	Scope and morphology	59
4.2.4	Contractions and clitics	60
4.2.5	Slang forms	61
4.2.6	Hyphenated forms	61
4.2.7	What is a sentence?	62
4.2.8	The lexicon	63
4.3	Text segmentation	63

4.3.1	Tokenisation	64
4.3.2	Tokenisation and punctuation	65
4.3.3	Tokenisation algorithms	66
4.3.4	Sentence splitting	67
4.4	Processing documents	68
4.4.1	Markup languages	68
4.4.2	Interpreting characters	70
4.5	Text-to-speech architectures	71
4.6	Discussion	75
4.6.1	Further reading	75
4.6.2	Summary	76
5	Text decoding: finding the words from the text	78
5.1	Overview of text decoding	78
5.2	Text-classification algorithms	79
5.2.1	Features and algorithms	79
5.2.2	Tagging and word-sense disambiguation	82
5.2.3	Ad-hoc approaches	83
5.2.4	Deterministic rule approaches	83
5.2.5	Decision lists	85
5.2.6	Naive Bayes classifier	86
5.2.7	Decision trees	87
5.2.8	Part-of-speech tagging	88
5.3	Non-natural-language text	92
5.3.1	Semiotic classification	92
5.3.2	Semiotic decoding	95
5.3.3	Verbalisation	95
5.4	Natural-language text	97
5.4.1	Acronyms and letter sequences	99
5.4.2	Homograph disambiguation	99
5.4.3	Non-homographs	101
5.5	Natural-language parsing	102
5.5.1	Context-free grammars	102
5.5.2	Statistical parsing	105
5.6	Discussion	105
5.6.1	Further reading	108
5.6.2	Summary	109
6	Prosody prediction from text	111
6.1	Prosodic form	111
6.2	Phrasing	112
6.2.1	Phrasing phenomena	112
6.2.2	Models of phrasing	113

6.3	Prominence	115
6.3.1	Syntactic prominence patterns	116
6.3.2	Discourse prominence patterns	118
6.3.3	Prominence systems, data and labelling	119
6.4	Intonation and tune	121
6.5	Prosodic meaning and function	122
6.5.1	Affective prosody	123
6.5.2	Suprasegmentality	124
6.5.3	Augmentative prosody	125
6.5.4	Symbolic communication and prosodic style	126
6.6	Determining prosody from the text	127
6.6.1	Prosody and human reading	127
6.6.2	Controlling the degree of augmentative prosody	128
6.6.3	Prosody and synthesis techniques	128
6.7	Phrasing prediction	129
6.7.1	Experimental formulation	129
6.7.2	Deterministic approaches	130
6.7.3	Classifier approaches	132
6.7.4	HMM approaches	133
6.7.5	Hybrid approaches	135
6.8	Prominence prediction	136
6.8.1	Compound-noun phrases	136
6.8.2	Function-word prominence	138
6.8.3	Data-driven approaches	138
6.9	Intonational-tune prediction	139
6.10	Discussion	139
6.10.1	Labelling schemes and labelling accuracy	139
6.10.2	Linguistic theories and prosody	141
6.10.3	Synthesising suprasegmental and true prosody	142
6.10.4	Prosody in real dialogues	143
6.10.5	Conclusion	144
6.10.6	Summary	144
7	Phonetics and phonology	146
7.1	Articulatory phonetics and speech production	146
7.1.1	The vocal organs	147
7.1.2	Sound sources	147
7.1.3	Sound output	150
7.1.4	The vocal-tract filter	150
7.1.5	Vowels	151
7.1.6	Consonants	153
7.1.7	Examining speech production	155
7.2	Acoustics, phonetics and speech perception	156

7.2.1	Acoustic representations	156
7.2.2	Acoustic characteristics	159
7.3	The communicative use of speech	160
7.3.1	Communicating discrete information with a continuous channel	161
7.3.2	Phonemes, phones and allophones	162
7.3.3	Allophonic variation and phonetic context	166
7.3.4	Coarticulation, targets and transients	168
7.3.5	The continuous nature of speech	169
7.3.6	Transcription	170
7.3.7	The distinctiveness of speech in communication	171
7.4	Phonology: the linguistic organisation of speech	172
7.4.1	Phonotactics	172
7.4.2	Word formation	179
7.4.3	Distinctive features and phonological theories	181
7.4.4	Syllables	184
7.4.5	Lexical stress	186
7.5	Discussion	189
7.5.1	Further reading	189
7.5.2	Summary	190
8	Pronunciation	192
8.1	Pronunciation representations	192
8.1.1	Why bother?	192
8.1.2	Phonemic and phonetic input	193
8.1.3	Difficulties in deriving phonetic input	194
8.1.4	A structured approach to pronunciation	195
8.1.5	Abstract phonological representations	196
8.2	Formulating a phonological representation system	197
8.2.1	Simple consonants and vowels	197
8.2.2	Difficult consonants	199
8.2.3	Diphthongs and affricates	201
8.2.4	Approximant–vowel combinations	201
8.2.5	Defining the full inventory	203
8.2.6	Phoneme names	204
8.2.7	Syllabic issues	206
8.3	The lexicon	207
8.3.1	Lexicon and rules	208
8.3.2	Lexicon formats	210
8.3.3	The offline lexicon	213
8.3.4	The system lexicon	214
8.3.5	Lexicon quality	215
8.3.6	Determining the pronunciations of unknown words	216
8.4	Grapheme-to-phoneme conversion	218

8.4.1	Rule-based techniques	218
8.4.2	Grapheme-to-phoneme alignment	219
8.4.3	Neural networks	219
8.4.4	Pronunciation by analogy	220
8.4.5	Other data-driven techniques	221
8.4.6	Statistical techniques	221
8.5	Further issues	222
8.5.1	Morphology	222
8.5.2	Language origin and names	223
8.5.3	Post-lexical processing	223
8.6	Summary	224
9	Synthesis of prosody	225
9.1	Intonation overview	225
9.1.1	F ₀ and pitch	226
9.1.2	Intonational form	226
9.1.3	Models of F ₀ contours	227
9.1.4	Micro-prosody	229
9.2	Intonational behaviour	229
9.2.1	Intonational tune	229
9.2.2	Downdrift	230
9.2.3	Pitch range	233
9.2.4	Pitch accents and boundary tones	234
9.3	Intonation theories and models	236
9.3.1	Traditional models and the British school	236
9.3.2	The Dutch school	237
9.3.3	Autosegmental–metrical and ToBI models	237
9.3.4	The INTSINT model	239
9.3.5	The Fujisaki model and superimpositional models	239
9.3.6	The Tilt model	242
9.3.7	Comparison	244
9.4	Intonation synthesis with AM models	245
9.4.1	Prediction of AM labels from text	246
9.4.2	Deterministic synthesis methods	246
9.4.3	Data-driven synthesis methods	247
9.4.4	Analysis with autosegmental models	248
9.5	Intonation synthesis with deterministic acoustic models	248
9.5.1	Synthesis with superimpositional models	249
9.5.2	Synthesis with the Tilt model	249
9.5.3	Analysis with Fujisaki and Tilt models	250
9.6	Data-driven intonation models	250
9.6.1	Unit-selection-style approaches	251
9.6.2	Dynamic-system models	252

9.6.3	Hidden Markov models	253
9.6.4	Functional models	254
9.7	Timing	254
9.7.1	Formulation of the timing problem	255
9.7.2	The nature of timing	255
9.7.3	Klatt rules	256
9.7.4	The sums-of-products model	257
9.7.5	The Campbell model	258
9.7.6	Other regression techniques	259
9.8	Discussion	259
9.8.1	Further reading	260
9.8.2	Summary	260
10	Signals and filters	262
10.1	Analogue signals	262
10.1.1	Simple periodic signals: sinusoids	263
10.1.2	General periodic signals	265
10.1.3	Sinusoids as complex exponentials	266
10.1.4	Fourier analysis	269
10.1.5	The frequency domain	270
10.1.6	The Fourier transform	275
10.2	Digital signals	278
10.2.1	Digital waveforms	279
10.2.2	Digital representations	280
10.2.3	The discrete-time Fourier transform	280
10.2.4	The discrete Fourier transform	281
10.2.5	The z -transform	282
10.2.6	The frequency domain for digital signals	283
10.3	Properties of transforms	284
10.3.1	Linearity	284
10.3.2	Time and frequency duality	284
10.3.3	Scaling	285
10.3.4	Impulse properties	285
10.3.5	Time delay	286
10.3.6	Frequency shift	286
10.3.7	Convolution	287
10.3.8	Analytical and numerical analysis	287
10.3.9	Stochastic signals	288
10.4	Digital filters	288
10.4.1	Difference equations	289
10.4.2	The impulse response	289
10.4.3	The filter convolution sum	292
10.4.4	The filter transfer function	293

10.4.5	The transfer function and the impulse response	293
10.5	Digital filter analysis and design	294
10.5.1	Polynomial analysis: poles and zeros	294
10.5.2	Frequency interpretation of the z -domain transfer function	297
10.5.3	Filter characteristics	298
10.5.4	Putting it all together	304
10.6	Summary	305
11	Acoustic models of speech production	309
11.1	The acoustic theory of speech production	309
11.1.1	Components in the model	309
11.2	The physics of sound	311
11.2.1	Resonant systems	311
11.2.2	Travelling waves	313
11.2.3	Acoustic waves	315
11.2.4	Acoustic reflection	317
11.3	The vowel-tube model	318
11.3.1	Discrete time and distance	319
11.3.2	Junction of two tubes	320
11.3.3	Special cases of junctions	322
11.3.4	The two-tube vocal-tract model	323
11.3.5	The single-tube model	325
11.3.6	The multi-tube vocal-tract model	327
11.3.7	The all-pole resonator model	329
11.4	Source and radiation models	330
11.4.1	Radiation	330
11.4.2	The glottal source	330
11.5	Model refinements	333
11.5.1	Modelling the nasal cavity	333
11.5.2	Source positions in the oral cavity	334
11.5.3	Models with vocal-tract losses	335
11.5.4	Source and radiation effects	336
11.6	Discussion	336
11.6.1	Further reading	339
11.6.2	Summary	339
12	Analysis of speech signals	341
12.1	Short-term speech analysis	341
12.1.1	Windowing	342
12.1.2	Short-term spectral representations	343
12.1.3	Frame lengths and shifts	345
12.1.4	The spectrogram	348
12.1.5	Auditory scales	351

12.2	Filter-bank analysis	352
12.3	The cepstrum	353
12.3.1	Cepstrum definition	353
12.3.2	Treating the magnitude spectrum as a signal	353
12.3.3	Cepstral analysis as deconvolution	355
12.3.4	Cepstral analysis discussion	356
12.4	Linear-prediction analysis	357
12.4.1	Finding the coefficients: the covariance method	358
12.4.2	The autocorrelation method	360
12.4.3	Levinson–Durbin recursion	361
12.5	Spectral-envelope and vocal-tract representations	362
12.5.1	Linear-prediction spectra	362
12.5.2	Transfer-function poles	364
12.5.3	Reflection coefficients	366
12.5.4	Log area ratios	367
12.5.5	Line-spectrum frequencies	367
12.5.6	Linear-prediction cepstra	369
12.5.7	Mel-scaled cepstra	370
12.5.8	Perceptual linear prediction	370
12.5.9	Formant tracking	370
12.6	Source representations	372
12.6.1	Residual signals	372
12.6.2	Closed-phase analysis	374
12.6.3	Open-phase analysis	377
12.6.4	Impulse/noise models	378
12.6.5	Parameterisation of glottal-flow signals	379
12.7	Pitch and epoch detection	379
12.7.1	Pitch detection	379
12.7.2	Epoch detection: finding the instant of glottal closure	381
12.8	Discussion	384
12.8.1	Further reading	385
12.8.2	Summary	386
13	Synthesis techniques based on vocal-tract models	387
13.1	Synthesis specification: the input to the synthesiser	387
13.2	Formant synthesis	388
13.2.1	Sound sources	389
13.2.2	Synthesising a single formant	390
13.2.3	Resonators in series and parallel	391
13.2.4	Synthesising consonants	392
13.2.5	A complete synthesiser	394
13.2.6	The phonetic input to the synthesiser	394
13.2.7	Formant-synthesis quality	397

13.3	Classical linear-prediction synthesis	399
13.3.1	Comparison with formant synthesis	399
13.3.2	The impulse/noise source model	400
13.3.3	Linear-prediction diphone-concatenative synthesis	401
13.3.4	A complete synthesiser	403
13.3.5	Problems with the source	404
13.4	Articulatory synthesis	405
13.5	Discussion	407
13.5.1	Further reading	409
13.5.2	Summary	410
14	Synthesis by concatenation and signal-processing modification	412
14.1	Speech units in second-generation systems	413
14.1.1	Creating a diphone inventory	414
14.1.2	Obtaining diphones from speech	414
14.2	Pitch-synchronous overlap and add (PSOLA)	415
14.2.1	Time-domain PSOLA	416
14.2.2	Epoch manipulation	417
14.2.3	How does PSOLA work?	421
14.3	Residual-excited linear prediction (RELP)	421
14.3.1	Residual manipulation	423
14.3.2	Linear-prediction PSOLA	423
14.4	Sinusoidal models	424
14.4.1	Pure sinusoidal models	425
14.4.2	Harmonic/noise models	426
14.5	MBROLA	429
14.6	Synthesis from cepstral coefficients	429
14.7	Concatenation issues	431
14.8	Discussion	433
14.8.1	Further reading	433
14.8.2	Summary	433
15	Hidden-Markov-model synthesis	435
15.1	The HMM formalism	435
15.1.1	Observation probabilities	436
15.1.2	Delta coefficients	438
15.1.3	Acoustic representations and covariance	439
15.1.4	States and transitions	440
15.1.5	Recognising with HMMs	440
15.1.6	Language models	443
15.1.7	The Viterbi algorithm	444
15.1.8	Training HMMS	447
15.1.9	Context-sensitive modelling	451

15.1.10	Are HMMs a good model of speech?	454
15.2	Synthesis from hidden Markov models	456
15.2.1	Finding the likeliest observations given the state sequence	457
15.2.2	Finding the likeliest observations and state sequence	459
15.2.3	Acoustic representations	459
15.2.4	Context-sensitive synthesis models	463
15.2.5	Duration modelling	464
15.2.6	Signal Processing in HMM synthesis	464
15.2.7	HMM synthesis systems	465
15.3	Labelling databases with HMMs	467
15.3.1	Determining the word sequence	467
15.3.2	Determining the phone sequence	468
15.3.3	Determining the phone boundaries	468
15.3.4	Measuring the quality of the alignments	470
15.4	Other data-driven synthesis techniques	471
15.5	Discussion	471
15.5.1	Further reading	471
15.5.2	Summary	472
16	Unit-selection synthesis	474
16.1	From concatenative synthesis to unit selection	474
16.1.1	Extending concatenative synthesis	475
16.1.2	The Hunt and Black algorithm	477
16.2	Features	479
16.2.1	Base types	479
16.2.2	Linguistic and acoustic features	480
16.2.3	Choice of features	481
16.2.4	Types of features	482
16.3	The independent-feature target-function formulation	484
16.3.1	The purpose of the target function	484
16.3.2	Defining a perceptual space	485
16.3.3	Perceptual spaces defined by independent features	486
16.3.4	Setting the target weights using acoustic distances	488
16.3.5	Limitations of the independent-feature formulation	491
16.4	The acoustic-space target-function formulation	493
16.4.1	Decision-tree clustering	494
16.4.2	General partial-synthesis functions	496
16.5	Join functions	497
16.5.1	Basic issues in joining units	497
16.5.2	Phone-class join costs	498
16.5.3	Acoustic-distance join costs	499
16.5.4	Combining categorical and acoustic join costs	500
16.5.5	Probabilistic and sequence join costs	501

16.5.6	Join classifiers	502
16.6	Searching	504
16.6.1	Base types and searching	505
16.6.2	Pruning	508
16.6.3	Pre-selection	508
16.6.4	Beam pruning	509
16.6.5	Multi-pass searching	509
16.7	Discussion	510
16.7.1	Unit selection and signal processing	511
16.7.2	Features, costs and perception	511
16.7.3	Example unit-selection systems	512
16.7.4	Further reading	514
16.7.5	Summary	515
17	Further issues	517
17.1	Databases	517
17.1.1	Unit-selection databases	517
17.1.2	Text materials	518
17.1.3	Prosody databases	518
17.1.4	Labelling	519
17.1.5	What exactly is hand labelling?	519
17.1.6	Automatic labelling	521
17.1.7	Avoiding explicit labels	521
17.2	Evaluation	522
17.2.1	System testing: intelligibility and naturalness	523
17.2.2	Word-recognition tests	523
17.2.3	Naturalness tests	524
17.2.4	Test data	525
17.2.5	Unit or component testing	525
17.2.6	Competitive evaluations	526
17.3	Audio-visual speech synthesis	527
17.3.1	Speech control	528
17.4	Synthesis of emotional and expressive speech	529
17.4.1	Describing emotion	529
17.4.2	Synthesising emotion with prosody control	529
17.4.3	Synthesising emotion with voice transformation	530
17.4.4	Unit selection and HMM techniques	531
17.5	Summary	531
18	Conclusion	533
18.1	Speech technology and linguistics	533
18.2	Future directions	536
18.3	Conclusion	539

<i>Appendix A Probability</i>	540
A.1 Discrete probabilities	540
A.1.1 Discrete random variables	540
A.1.2 Probability mass functions	541
A.1.3 Expected values	541
A.1.4 Moments of a PMF	542
A.2 Pairs of discrete random variables	542
A.2.1 Marginal distributions	543
A.2.2 Independence	543
A.2.3 Expected values	543
A.2.4 Moments of a joint distribution	544
A.2.5 Higher-order moments and covariance	544
A.2.6 Correlation	544
A.2.7 Conditional probability	545
A.2.8 Bayes' rule	545
A.2.9 Sum of random variables	545
A.2.10 The chain rule	546
A.2.11 Entropy	546
A.3 Continuous random variables	547
A.3.1 Continuous random variables	548
A.3.2 Expected values	548
A.3.3 The Gaussian distribution	549
A.3.4 The uniform distribution	549
A.3.5 Cumulative density functions	549
A.4 Pairs of continuous random variables	550
A.4.1 Independent versus uncorrelated	551
A.4.2 The sum of two random variables	551
A.4.3 Entropy	552
A.4.4 Kullback–Leibler distance	552
 <i>Appendix B Phone definitions</i>	 553
<i>References</i>	556
<i>Index</i>	583