

CPC COOPERATIVE PATENT CLASSIFICATION

G PHYSICS (NOTES omitted)

INSTRUMENTS

G10 MUSICAL INSTRUMENTS; ACOUSTICS (NOTES omitted)

G10L SPEECH ANALYSIS OR SYNTHESIS; SPEECH RECOGNITION; SPEECH OR VOICE PROCESSING; SPEECH OR AUDIO CODING OR DECODING

NOTE

This subclass does not cover:

- devices for the storage of speech signals, which are covered by subclasses [G11B](#) and [G11C](#);
- encoding of compressed speech signals for transmission or storage, which is covered by group [H03M 7/30](#).

WARNING

In this subclass non-limiting references (in the sense of paragraph 39 of the Guide to the IPC) may still be displayed in the scheme.

13/00	Speech synthesis; Text to speech systems	2015/025	• • {Phonemes, fenemes or fenones being the recognition units}
13/02	• Methods for producing synthetic speech; Speech synthesisers	2015/027	• • {Syllables being the recognition units}
2013/021	• • {Overlap-add techniques}	15/04	• Segmentation; Word boundary detection
13/027	• • Concept to speech synthesisers; Generation of natural phrases from machine-based concepts (generation of parameters for speech synthesis out of text G10L 13/08)	15/05	• • Word boundary detection
13/033	• • Voice editing, e.g. manipulating the voice of the synthesiser	15/06	• Creation of reference templates; Training of speech recognition systems, e.g. adaptation to the characteristics of the speaker's voice (G10L 15/14 takes precedence)
13/0335	• • • {Pitch control}	15/063	• • • {Training}
13/04	• • Details of speech synthesis systems, e.g. synthesiser structure or memory management	2015/0631	• • • {Creating reference templates; Clustering}
13/043	• • • {Synthesisers specially adapted to particular applications}	2015/0633	• • • • {using lexical or orthographic knowledge sources}
13/047	• • • Architecture of speech synthesisers	2015/0635	• • • • {updating or merging of old and new templates; Mean values; Weighting}
13/06	• Elementary speech units used in speech synthesisers; Concatenation rules	2015/0636	• • • • • {Threshold criteria for the updating}
13/07	• • Concatenation rules	2015/0638	• • • • {Interactive procedures}
13/08	• Text analysis or generation of parameters for speech synthesis out of text, e.g. grapheme to phoneme translation, prosody generation or stress or intonation determination	15/065	• • Adaptation
2013/083	• • {Special characters, e.g. punctuation marks}	15/07	• • • to the speaker
13/086	• • {Detection of language}	15/075	• • • • {supervised, i.e. under machine guidance}
13/10	• • Prosody rules derived from text; Stress or intonation	15/08	• Speech classification or search
2013/105	• • • {Duration}	2015/081	• • {Search algorithms, e.g. Baum-Welch or Viterbi}
15/00	Speech recognition (G10L 17/00 takes precedence)	15/083	• • {Recognition networks (G10L 15/142 , G10L 15/16 take precedence)}
15/005	• {Language recognition}	2015/085	• • • {Methods for reducing search complexity, pruning}
15/01	• Assessment or evaluation of speech recognition systems	2015/086	• • • {Recognition of spelled words}
15/02	• Feature extraction for speech recognition; Selection of recognition unit	2015/088	• • • {Word spotting}
2015/022	• • {Demisyllables, biphones or triphones being the recognition units}	15/10	• • using distance or distortion measures between unknown speech and reference templates
		15/12	• • using dynamic programming techniques, e.g. dynamic time warping [DTW]
		15/14	• • using statistical models, e.g. Hidden Markov Models [HMMs] (G10L 15/18 takes precedence)
		15/142	• • • • {Hidden Markov Models [HMMs]}
		15/144	• • • • • {Training of HMMs}

15/146 {with insufficient amount of training data, e.g. state sharing, tying, deleted interpolation}	17/00	Speaker identification or verification
15/148 {Duration modelling in HMMs, e.g. semi HMM, segmental models or transition probabilities}	17/005	. {Speaker recognisers specially adapted for particular applications (G07C 9/00071 takes precedence)}
15/16	. . using artificial neural networks	17/02	. Preprocessing operations, e.g. segment selection; Pattern representation or modelling, e.g. based on linear discriminant analysis [LDA] or principal components; Feature selection or extraction
15/18	. . using natural language modelling	17/04	. Training, enrolment or model building
15/1807	. . . {using prosody or stress}	17/06	. Decision making techniques; Pattern matching strategies
15/1815	. . . {Semantic context, e.g. disambiguation of the recognition hypotheses based on word meaning}	17/08	. . Use of distortion metrics or a particular distance between probe pattern and reference templates
15/1822	. . . {Parsing for meaning understanding}	17/10	. . Multimodal systems, i.e. based on the integration of multiple recognition engines or fusion of expert systems
15/183	. . . using context dependencies, e.g. language models	17/12	. . Score normalisation
15/187 Phonemic context, e.g. pronunciation rules, phonotactical constraints or phoneme n-grams	17/14	. . Use of phonemic categorisation or speech recognition prior to speaker recognition or verification
15/19 Grammatical context, e.g. disambiguation of the recognition hypotheses based on word sequence rules	17/16	. Hidden Markov models [HMMs]
15/193 Formal grammars, e.g. finite state automata, context free grammars or word networks	17/18	. Artificial neural networks; Connectionist approaches
15/197 Probabilistic grammars, e.g. word n-grams	17/20	. Pattern transformations or operations aimed at increasing system robustness, e.g. against channel noise or different working conditions
15/20	. Speech recognition techniques specially adapted for robustness in adverse environments, e.g. in noise, of stress induced speech (G10L 21/02 takes precedence)	17/22	. Interactive procedures; Man-machine interfaces
15/22	. Procedures used during a speech recognition process, e.g. man-machine dialogue	17/24	. . the user being prompted to utter a password or a predefined phrase
2015/221	. . {Announcement of recognition results}	17/26	. Recognition of special voice characteristics, e.g. for use in lie detectors; Recognition of animal voices
15/222	. . {Barge in, i.e. overridable guidance for interrupting prompts}	19/00	Speech or audio signals analysis-synthesis techniques for redundancy reduction, e.g. in vocoders; Coding or decoding of speech or audio signals, using source filter models or psychoacoustic analysis (in musical instruments G10H)
2015/223	. . {Execution procedure of a spoken command}	2019/0001	. {Codebooks}
2015/225	. . {Feedback of the input speech}	2019/0002	. . {Codebook adaptations}
2015/226	. . {Taking into account non-speech characteristics}	2019/0003	. . {Backward prediction of gain}
2015/227	. . . {of the speaker; Human-factor methodology}	2019/0004	. . {Design or structure of the codebook}
2015/228	. . . {of application context}	2019/0005	. . . {Multi-stage vector quantisation}
15/24	. Speech recognition using non-acoustical features	2019/0006	. . . {Tree or treillis structures; Delayed decisions}
15/25	. . using position of the lips, movement of the lips or face analysis	2019/0007	. . {Codebook element generation}
15/26	. Speech to text systems (G10L 15/08 takes precedence)	2019/0008	. . . {Algebraic codebooks}
15/265	. . {Speech recognisers specially adapted for particular applications (devices for signalling identity of wanted subscriber in a telephonic communication equipment controlled by voice recognition H04M 1/271 ; speech interaction details in interactive information services in a telephonic communication system H04M 3/4936)}	2019/0009	. . . {Orthogonal codebooks}
15/28	. Constructional details of speech recognition systems	2019/001	. . . {Interpolation of codebook vectors}
15/285	. . {Memory allocation or algorithm optimisation to reduce hardware requirements}	2019/0011	. . {Long term prediction filters, i.e. pitch estimation}
15/30	. . Distributed recognition, e.g. in client-server systems, for mobile phones or network applications	2019/0012	. . {Smoothing of parameters of the decoder interpolation}
15/32	. . Multiple recognisers used in sequence or in parallel; Score combination systems therefor, e.g. voting systems	2019/0013	. . {Codebook search algorithms}
15/34	. . Adaptation of a single recogniser for parallel processing, e.g. by use of multiple processors or cloud computing	2019/0014	. . . {Selection criteria for distances}
		2019/0015	. . . {Viterbi algorithms}
		2019/0016	. . {Codebook for LPC parameters}
		19/0017	. {Lossless audio signal coding; Perfect reconstruction of coded audio signal by transmission of coding error (G10L 19/24 takes precedence)}
		19/0018	. {Speech coding using phonetic or linguistic decoding of the source; Reconstruction using text-to-speech synthesis}

19/0019	• {Vocoders specially adapted for particular applications}	19/125	• Pitch excitation, e.g. pitch synchronous innovation CELP [PSI-CELP]
19/002	• Dynamic bit allocation (for perceptual audio coders G10L 19/032)	19/13	• Residual excited linear prediction [RELP]
19/005	• Correction of errors induced by the transmission channel, if related to the coding algorithm	19/135	• Vector sum excited linear prediction [VSELPA]
19/008	• Multichannel audio signal coding or decoding, i.e. using interchannel correlation to reduce redundancies, e.g. joint-stereo, intensity-coding, matrixing (arrangements for reproducing spatial sound H04R 5/00 ; stereophonic systems, e.g. spatial sound capture or matrixing of audio signals in the decoded state H04S)	19/16	• . . Vocoder architecture
19/012	• Comfort noise or silence coding	19/167	• . . . {Audio streaming, i.e. formatting and decoding of an encoded audio signal representation into a data stream for transmission or storage purposes}
19/018	• Audio watermarking, i.e. embedding inaudible data in the audio signal	19/173	• . . . {Transcoding, i.e. converting between two coded representations avoiding cascaded coding-decoding}
19/02	• using spectral analysis, e.g. transform vocoders or subband vocoders	19/18	• . . . Vocoders using multiple modes
19/0204	• . . {using subband decomposition}	19/20	• using sound class specific coding, hybrid encoders or object based coding
19/0208	• . . . {Subband vocoders}	19/22	• Mode decision, i.e. based on audio signal content versus external parameters
19/0212	• . . {using orthogonal transformation}	19/24	• Variable rate codecs, e.g. for generating different qualities using a scalable representation such as hierarchical encoding or layered encoding
19/0216	• . . . {using wavelet decomposition}	19/26	• . . Pre-filtering or post-filtering
19/022	• . . Blocking, i.e. grouping of samples in time; Choice of analysis windows; Overlap factoring	19/265	• . . . {Pre-filtering, e.g. high frequency emphasis prior to encoding}
19/025	• . . . Detection of transients or attacks for time/frequency resolution switching	21/00	Processing of the speech or voice signal to produce another audible or non-audible signal, e.g. visual or tactile, in order to modify its quality or its intelligibility (G10L 19/00 takes precedence)
19/028	• . . Noise substitution, i.e. substituting non-tonal spectral components by noisy source (comfort noise for discontinuous speech transmission G10L 19/012)	21/003	• Changing voice quality, e.g. pitch or formants
19/03	• . . Spectral prediction for preventing pre-echo; Temporary noise shaping [TNS], e.g. in MPEG2 or MPEG4	21/007	• . characterised by the process used
19/032	• . . Quantisation or dequantisation of spectral components	21/01	• . . . Correction of time axis
19/035	• . . . Scalar quantisation	21/013	• . . . Adapting to target pitch
19/038	• . . . Vector quantisation, e.g. TwinVQ audio	2021/0135	• {Voice conversion or morphing}
19/04	• using predictive techniques	21/02	• Speech enhancement, e.g. noise reduction or echo cancellation (reducing echo effects in line transmission systems H04B 3/20 ; echo suppression in hands-free telephones H04M 9/08)
19/06	• . . Determination or coding of the spectral characteristics, e.g. of the short-term prediction coefficients	21/0202	• . . {Applications}
19/07	• . . . Line spectrum pair [LSP] vocoders	21/0205	• . . . {Enhancement of intelligibility of clean or coded speech}
19/08	• . . Determination or coding of the excitation function; Determination or coding of the long-term prediction parameters	21/0208	• . . Noise filtering
19/083	• . . . the excitation function being an excitation gain (G10L 25/90 takes precedence)	2021/02082	• . . . {the noise being echo, reverberation of the speech}
19/087	• . . . using mixed excitation models, e.g. MELP, MBE, split band LPC or HVXC	2021/02085	• . . . {Periodic noise}
19/09	• . . . Long term prediction, i.e. removing periodical redundancies, e.g. by using adaptive codebook or pitch predictor	2021/02087	• . . . {the noise being separate speech, e.g. cocktail party}
19/093	• . . . using sinusoidal excitation models	21/0216	• . . . characterised by the method used for estimating noise
19/097	• . . . using prototype waveform decomposition or prototype waveform interpolative [PWI] coders	2021/02161	• {Number of inputs available containing the signal or the noise to be suppressed}
19/10	• . . . the excitation function being a multipulse excitation	2021/02163	• {Only one microphone}
19/107	• Sparse pulse excitation, e.g. by using algebraic codebook	2021/02165	• {Two microphones, one receiving mainly the noise signal and the other one mainly the speech signal}
19/113	• Regular pulse excitation	2021/02166	• {Microphone arrays; Beamforming}
19/12	• . . . the excitation function being a code excitation, e.g. in code excited linear prediction [CELP] vocoders	2021/02168	• {the estimation exclusively taking place during speech pauses}
		21/0224	• Processing in the time domain
		21/0232	• Processing in the frequency domain
		21/0264	• . . . characterised by the type of parameter measurement, e.g. correlation techniques, zero crossing techniques or predictive techniques

21/0272	. . Voice signal separating	25/48	. . specially adapted for particular use
21/028	. . . using properties of sound source	25/51	. . for comparison or discrimination
21/0308	. . . characterised by the type of parameter measurement, e.g. correlation techniques, zero crossing techniques or predictive techniques	25/54	. . . for retrieval
21/0316	. . by changing the amplitude	25/57	. . . for processing of video signals
21/0324	. . . Details of processing therefor	25/60	. . . for measuring the quality of voice signals
21/0332 involving modification of waveforms	25/63	. . . for estimating an emotional state
21/034 Automatic adjustment	25/66	. . . for extracting parameters related to health condition (detecting or measuring for diagnostic purposes A61B 5/00)
21/0356	. . . for synchronising with other signals, e.g. video signals	25/69	. . for evaluating synthetic or decoded voice signals
21/0364	. . . for improving intelligibility	25/72	. . for transmitting results of analysis
2021/03643 {Diver speech}	25/75	. for modelling vocal tract parameters
2021/03646 {Stress or Lombard effect}	25/78	. Detection of presence or absence of voice signals (switching of direction of transmission by voice frequency in two-way loud-speaking telephone systems H04M 9/10)
21/038	. . using band spreading techniques	2025/783	. . {based on threshold decision}
21/0388	. . . Details of processing therefor	2025/786	. . . {Adaptive threshold}
21/04	. Time compression or expansion	25/81	. . for discriminating voice from music
21/043	. . by changing speed	25/84	. . for discriminating voice from noise
21/045	. . . using thinning out or insertion of a waveform	25/87	. . Detection of discrete points within a voice signal
21/047 characterised by the type of waveform to be thinned out or inserted	25/90	. Pitch determination of speech signals
21/049 characterised by the interconnection of waveforms	2025/903	. . {using a laryngograph}
21/055	. . for synchronising with other signals, e.g. video signals	2025/906	. . {Pitch tracking}
21/057	. . for improving intelligibility	25/93	. Discriminating between voiced and unvoiced parts of speech signals (G10L 25/90 takes precedence)
2021/0575	. . . {Aids for the handicapped in speaking}	2025/932	. . {Decision in previous or following frames}
21/06	. Transformation of speech into a non-audible representation, e.g. speech visualisation or speech processing for tactile aids (G10L 15/26 takes precedence)	2025/935	. . {Mixed voiced class; Transitions}
2021/065	. . . {Aids for the handicapped in understanding}	2025/937	. . {Signal energy in various frequency bands}
21/10	. . Transforming into visible information	99/00	Subject matter not provided for in other groups of this subclass
2021/105	. . . {Synthesis of the lips movements from speech, e.g. for talking heads}		
21/12	. . . by displaying time domain information		
21/14	. . . by displaying frequency domain information		
21/16	. . Transforming into a non-visible representation (devices or methods enabling ear patients to replace direct auditory perception by another kind of perception A61F 11/04)		
21/18	. . Details of the transformation process		
25/00	Speech or voice analysis techniques not restricted to a single one of groups G10L 15/00-G10L 21/00		
25/03	. characterised by the type of extracted parameters		
25/06	. . the extracted parameters being correlation coefficients		
25/09	. . the extracted parameters being zero crossing rates		
25/12	. . the extracted parameters being prediction coefficients		
25/15	. . the extracted parameters being formant information		
25/18	. . the extracted parameters being spectral information of each sub-band		
25/21	. . the extracted parameters being power information		
25/24	. . the extracted parameters being the cepstrum		
25/27	. characterised by the analysis technique		
25/30	. . using neural networks		
25/33	. . using fuzzy logic		
25/36	. . using chaos theory		
25/39	. . using genetic algorithms		
25/45	. characterised by the type of analysis window		