

Type	num	Settable?	Encoding	Units	Global	Valid range/values	Meaning & Use
AD_BITS_PER_SAMPLE	82	uint	bits	Y			Width of A/D input word
AD_OVER	104	uint		Y			Count of A/D overrange events
AGC_ENABLE	62	Y	boolean				Automatic gain control (Linear demod only)
AGC_HANGTIME	64	Y	float32	sec		>= 0	Time delay before automatic gain increase on lowered signal (linear demod only)
AGC_RECOVERY_RATE	65	Y	float32	dB/sec		>= 0	Gain increase rate on lowered signal (linear demod only)
AGC_THRESHOLD	67	Y	float32	dBFS		<= 0	Target demodulator output level on noise only
BASEBAND_POWER	46	float32	dB				Signal power at channel downconverter filter output, relative to unity
BIN_BYTE_DATA	9	bytes				Limited by IP packet size	Vector representing spectrum data, 1 bin/byte, unsigned, with meaning SPECTRUM_BASE + SPECTRUM_STEP * x dB
BIN_COUNT	94	Y	uint	bins		> 0	Number of FFT bins wanted in spectrum data
BIN_DATA	96	float32 vector					Vector of 4-byte float32s with linear spectrum data, spectrum mode only. Order: DC...max positive freq, max neg freq...-1
BLOCKS_SINCE_POLL	103	uint					Count of status frames sent since last command received. Deprecated
CALIBRATE	24	Y	double64		Y		Frequency calibration factor for tuner reference and A/D sample clock. Actual freq = nominal * (1 + CALIBRATE), ie, 0 means "on frequency"
CLEAROPTS	7	Y	uint			32 bits	1-bits turn off specified option (32 max)
CMD_CNT	2	uint		Y			Server count of received commands
COHERENT_BIN_SPACING	92	float32		y			deprecated; can be calculated from INPUT_SAMPLERATE, FILTER_BLOCKSIZE, FILTER_FIR_LENGTH
COMMAND_TAG	1	Y	uint		Y		generated by controller, echoed by server to confirm command
CONVERTER_OFFSET	88	float32	Hz	Y			Frequency offset of external frequency converter. Not yet implemented
CROSSOVER	95	Y	float32			positive	Value of RESOLUTION_BW above which the wideband spectrum analyzer is used.
DC_I_OFFSET	28	float32		Y			DC offset of I-channel A/D converter (only direct conversion front ends)
DC_Q_OFFSET	29	float32		Y			DC offset of Q-channel A/D converter (only direct conversion front ends)
DEEMPH_GAIN	87	float32	dB				Static gain correction when de-emphasis used to maintain subjectively equal loudness
DEEMPH_TC	86	float32	s			non-negative	Deemphasis time constant (0 = off), FM only
DEMOD_TYPE	48	Y	uint	enum		0 - 4	Demodulator type, enum: 0 = linear; 1 = FM/PM; 2 = Wideband FM with stereo demodulator; 3 = spectrum; 4 = spectrum v2
DESCRIPTION	4	string		Y		free-form UTF-8	description of front end (antenna, etc). Generated by front end, passed through 'radio'. Need not be null terminated.
DIRECT_CONVERSION	32	boolean		Y			Front end uses direct conversion with DC spike and 1/f noise that should be avoided
DOPPLER_FREQUENCY	37	Y	double64	Hz			Doppler tuning offset (untested)
DOPPLER_FREQUENCY_RATE	38	Y	double64	Hz/sec			Rate of change of Doppler tuning effort (untested). Limited to 1 bin/frame time
ENVELOPE	56	Y	boolean				Use envelope detector in linear demodulator
EOL	0						End of option list. No length or value field
FE_HIGH_EDGE	101	float32	Hz	Y	> FE_LOW_EDGE		Upper edge of A/D converter input frequency band (negative for tuners with high side injection)
FE_ISREAL	102	boolean		Y			Y: front end uses a single A/D converter; N: front end uses dual (I/Q) A/D converters
FE_LOW_EDGE	100	float32	Hz	Y	<FE_HIGH_EDGE		Low edge of A/D converter input frequency band (negative for tuners with high side injection)
FILTER_BLOCKSIZE	42	uint	samples	Y	>0		Input samples per downconverter FFT processing block
FILTER_DROPS	77	uint	blocks				Number of frame drops by digital downconverter
FILTER_FIR_LENGTH	43	uint	samples	Y	>0		Overlap samples per downconverter FFT processing block. Sets maximum impulse duration of downconverter channel filter
FILTER2	44	Y	uint	frames		0-10	Size of secondary filter input in units of downconverter frames (eg, 20 ms). 0=off
FILTER2_BLOCKSIZE	73	uint	samples		>0		Input samples per Filter2 block
FILTER2_FIR_LENGTH	74	uint	samples		>0		Impulse response length in secondary filter (filter2)
FILTER2_KAISER_BETA	75	Y	float32		>=0		Kaiser β factor for secondary filter (filter2) window design
FIRST_LO_FREQUENCY	34	Y	double64	Hz	y		Front end tuner frequency. N/A for direct sampling front ends
FM_SNR	66	float32	dB				Estimated SNR in FM; equal to channel SNR when moments squelch is disabled
FREQ_OFFSET	59	float32	Hz				Estimated signal frequency error. Only when PLL is enabled in linear modes
GAIN	68	Y	float32	dB			Demodulator gain (constant for FM, variable for linear)
GAINSTEP	81	uint		Y			Front end analog gain, arbitrary units, hardware specific
GPS_TIME	3	uint	ns	Y			Nanoseconds since GPS epoch of 6 January 1980 00:00:00 UTC. Generated by front end, passed through 'radio'
HEADROOM	63	Y	float32	dBFS		<= 0	Target channel output audio level, block average
HIGH_EDGE	40	Y	float32	Hz	LOW_EDGE < HIGH_EDGE < +Fs/2		Upper edge of post-mixer filter
IF_GAIN	27	Y	uint	dB	Y	0-255	Relative gain of baseband analog amplifier in tuner just ahead of A/D converter. Hardware dependent, not used by all front ends
IF_POWER	45	float32	dBFS	Y			A/D output level
INDEPENDENT_SIDEband	50	Y	boolean				LSB in left channel, USB in right channel - currently unimplemented
INPUT_SAMPLES	13	uint					Count of input data samples
INPUT_SAMPLERATE	10	uint	Hz				Sample rate of RTP input data stream
IQ_IMBALANCE	30	float32		Y			Relative gain of I and Q channels (only direct conversion front ends). 1 = no error
IQ_PHASE	31	float32	radians	Y	-pi/2 to +pi/2		Relative phase error of I & Q channels. 0 = no error
KAISER_BETA	41	Y	float32		>= 0		Kaiser β factor for downconverter filter window design
LNA_GAIN	25	Y	uint	dB	Y	0-255	Relative gain of analog input to receiver. Hardware dependent, not used by all front ends
LOCK	78	boolean					Will ignore frequency tuning commands. Not yet implemented
LOW_EDGE	39	Y	float32	Hz	-Fs/2 < LOW_EDGE < HIGH_EDGE		Lower edge of post-mixer filter
MINPACKET	72	Y	uint	frames		0-10	Minimum number of receiver frames in an output IP packet, unless packet is already at MTU
MIXER_GAIN	26	Y	uint	dB	Y	0-255	Relative gain of mixer in analog receiver/downconverter. Hardware dependent, not used by all front ends
NOISE_BW	15	float32	bins		>0		Relative noise bandwidth of each FFT bin in spectral display, depends on WINDOW_TYPE and SPECTRUM_SHAPE. 1 for rectangular
NOISE_DENSITY	47	float32	dBMJ				Estimated noise spectral power density, N0, in and near downconverter channel
OPUS_APPLICATION	112	Y	uint			2048 (VOIP), 2049 (AUDIO), 2051 (Low Delay)	The application type parameter to the Opus encoder.
OPUS_BANDWIDTH	113	Y	uint	Hz		1101 (4 kHz), 1102 (6 kHz), 1103 (8 kHz), 1104 (12 kHz) 1105 (20 kHz)	Input bandwidth to be considered important by Opus encoder
OPUS_BITRATE	71	Y	uint	bits/sec			Target bitrate of Opus-compressed audio. 0 = auto
OPUS_DTX	111	Y	boolean				enable discontinuous transmission by the Opus encoder
OPUS_FEC	114	Y	uint	percent		0-100	Frame loss percentage the Opus encoder should use when in FEC mode
OUTPUT_CHANNELS	49	Y	uint			1 - 2	mono (~1) stereo (~2); for front ends, 1 channel = real, 2 channels = complex (IQ)
OUTPUT_DATA_DEST_SOCKET	17	Y	socket				Destination (multicast) IP address and port of output data stream
OUTPUT_DATA_PACKETS	22	uint	count				Count of RTP output data packets
OUTPUT_DATA_SOURCE_SOCKET	16	socket		Y			Source IP and port of output RTP data stream
OUTPUT_ENCODING	107	Y	uint			1-7	1=S16LE, 2=S16BE, 3=Opus, 4=F32BE, 5=AX25, 6=F16BE, 7=Opus with application=voip
OUTPUT_ERRORS	23	uint	count				Count of send errors on channel output packet stream
OUTPUT_LEVEL	69	float32	dBFS				Output level, frame average
OUTPUT_METADATA_PACKETS	21	uint	count	y			Count of metadata packets sent
OUTPUT_SAMPLES	70	uint	samples				Output sample count
OUTPUT_SAMPLERATE	20	Y	uint	Hz		>0	Sample rate of RTP output data stream
OUTPUT_SSRC	18	Y	uint			32 bits; 0 and ffffff reserved	RTP stream ID of output stream
OUTPUT_TTL	19	uint	hops	Y	0 <= ttl <= 255		IP Time-to-live (hop count limit) of output data stream (not metadata, which can be different)
PEAK_DEVIATION	60	float32	Hz		>= 0		Peak deviation (FM demodulators only)
PL_DEVIATION	89	float32	Hz		>= 0		Measured deviation of PL tone (FM demodulator only, tone squelch enabled)
PL_TONE	61	float32	Hz		>= 0		PL tone squelch frequency (FM demodulator only); 0 = no tone
PLL_BW	55	Y	float32	Hz		>0	Noise bandwidth of PLL loop filter
PLL_ENABLE	51	Y	boolean				Enable 0 Hz carrier tracking & squelch (Linear mode); PLL demodulation (FM)
PLL_LOCK	52	boolean					Indicate whether PLL is in lock (controlled by squelch threshold settings)
PLL_PHASE	54	float32	radians?		0 - 2 pi		Relative phase of PLL numerically controlled oscillator
PLL_SNR	58	float32	dB				Phase lock loop signal-to-noise ratio; = 10log10($\lambda^2/Q^2 - 1$), in-phase to quadrature power ratio
PLL_SQUARE	53	Y	boolean				Square feedback to PLL; use for DSB AM and BPSK. Implies PLL_ENABLE
PLL_WRAPS	109	signed int	rotations				Count of complete 360 degree (2 pi radian) rotations of PLL
PRESET	85	Y	string			defined in presets.conf	Set demodulator mode - configured by modes.conf on 'radio'
RADIO_FREQUENCY	33	Y	double64	Hz	Y	>0 for real inputs. 0 reserved to mean "idle channel"	RF tuning frequency that comes out of the downconverter at 0 Hz. I.e., "carrier frequency"
RESOLUTION_BW	93	Y	float32	Hz		positive	Width of each bin in spectrum data
RF_AGC	99	Y	boolean		Y		Enable front end RF automatic gain control
RF_ATTEN	97	Y	float32	dB	Y	<= 0	Front end attenuation, hardware dependent (not present on all front ends). Setting turns off RF_AGC
RF_GAIN	98	Y	float32	dB	Y		Front end gain; hardware dependent. Setting turns off RF_AGC
RF_LEVEL_CAL	110	float32	dBrn	Y			Input power that gives 0 dBFS from A/D converter with all RF gain and atten = 0
RTP_PT	105	uint				0-127	Real Time Protocol Payload Type for current output stream
RTP_TIMESNAP	8	uint				32 bits	Snapshot of RTP 32-bit timestamp field
SAMPLES_SINCE_OVER	108	uint		Y			Input A/D samples since last overrange event
SECOND_LO_FREQUENCY	35	double64	Hz				Digital down converter frequency = -RADIO_FREQUENCY+DOPPLER_FREQUENCY-FIRST_LO_FREQUENCY. Negative of IF frequency, <0 for direct sampling front ends. May be >0 or <0 for direct conversion (I/Q) front ends. >0 for tuners with high-side injection, <0 for tuners with low-side injection.
SETOPTS	6	Y	uint			32 bits	1-bits turn on specified option (32 max)
SHIFT_FREQUENCY	36	Y	double64	Hz			Post-downconversion shift frequency, used primarily for CW. With IF filter centered, <0 shifts the LSB up in output frequency, >0 shifts the USB up in output frequency
SNR_SQUELCH	57	Y	boolean				Y=enable SNR squelch in all modes. N=enable moments squelch (in FM mode only)
SPECTRUM_AVG	12	Y	uint	frames		>= 1	Number of consecutive periodograms (power spectra) to be averaged in each response.
SPECTRUM_BASE	11	Y	float32	dB			Used to interpret BIN_BYTE_DATA
SPECTRUM_FFT_N	76	uint	bins			> 0	Number of bins in analysis FFTs. Calculated from BIN_COUNT and RESOLUTION_BW
SPECTRUM_OVERLAP	116	Y	float32			0-1	If SPECTRUM_AVG > 1, sets overlap of averaged FFT windows