[File No.402-30/2001-FN] Telecom Regulatory Authority of India NOTIFICATION

New Delhi 15th November 2002

In exercise of the powers conferred upon it under sub–clause (b) (v) of sub section (1) of section 11 of TRAI (Amendment) Act 2000 in regard to laying down the standards of quality of service to be provided by the service providers, the Telecom Regulatory Authority of India hereby makes the following Regulation:

REGULATION ON QUALITY OF SERVICE FOR VOIP BASED INTERNATIONAL LONG DISTANCE SERVICE, 2002 (3 of 2002)

Section-I Title, Extent and Commencement

Short title, extent and commencement

- 1. (i) The title of this regulation shall be 'Regulation on Quality of Service for VOIP based International Long Distance Service, 2002'
 - (ii) This regulation shall be applicable to all the International long Distance Service Providers using VOIP based networks.
 - (iii) This regulation shall come into effect from the date of its notification in the Official Gazette.

Section-II

Definitions

2 In this Regulation, unless the context otherwise requires:

(i) 'Act' means the Telecom Regulatory Authority of India Act, 1997, as amended by, TRAI (Amendment) Act, 2000 and as amended from time to time.

(ii) 'Authority' means the Telecom Regulatory Authority of India.

(iii) '**MOS**' means Mean Opinion Score, a subjective measure of speech quality as defined in ITU-T Recommendation P.800, 'Methods for Subjective Determination of Transmission Quality, August 1996'

(iv) '**Regulation**' means the Regulation on Quality of Service for VOIP based International Long Distance Service notified by the Authority under the 'Act'.

(v) '**R-Value'** is the objective measure of speech quality denoted as the resultant value of the 'Transmission Rating Factor' as defined in ITU-T Recommendations G.107, 'E-Model, Computation Model for Use in Transmission Planning, August 2001' and G.108, 'Application of the E-Model, A Planning Guide, September 1999'.

(vi) 'Service Provider' in the context of this regulation means a legal entity holding an ILD licence

(vii) 'Quality of Service (QOS)' means the indicator of performance of a network and of the degree to which the network conforms to the stipulated norms.

(viii) **'Voice over Internet Protocol (VOIP)'** means a technology that enables the carriage of real time voice traffic over a packet network by using Internet Protocol (IP).

AP	Access Provider		
BSO	Basic Service Operator		
BHCA	Busy Hour Call Attempt		
CCR	Call Completion Rate		
CCS7	Common Channel Signaling system no. 7		
DTMF	Dual Tone Multi Frequency		
ILDO	International Long Distance Operator		
IP	Internet Protocol		
IPTD	IP Packet Transfer Delay		
ISDN	Integrated Services Digital Network		
ITU-T	International Telecommunication Union –		
	Telecommunication standardization sector		
IUT	Implementation Under Test		
IVR	Interactive Voice Response		
MOS	Mean Opinion Score		
ms	milli second		
NLDO	National Long Distance Operator		
PLMN	Public Land Mobile Network		
PoP	Point of Presence		
PSTN	Public Switched Telephone Network		
QoS	Quality of Service		
TEC	Telecom Engineering Centre		
TRAI	Telecom Regulatory Authority of India		
VoIP	Voice over Internet Protocol		

(ix) Acronyms & Abbreviations

 ${}^{1}[(x)$ The Advantage Factor: The E-Model provides for advantage factor allowing downward adjustment in the R-value required in situations where the user is being provided a

¹ Ins. by the First Amendment Regulations, 2004, reg. 2 (w.e.f. 20.01.2004)

certain "advantage"; or some specific transmission media is used for instance international connection with satellite link.

Example:- When VOIP technology is used in an international connection with satellite link, service may be considered Toll Quality even if the end-to-end delay exceeds 150 ms, but is below 400 ms due to the Advantage Factor of 10 points in R-value.]

Section-III

Objective of laying down Quality of Service Benchmarks:

- **3.** The objective of laying down Quality of Service benchmarks is to:
 - (i) Ensure customer satisfaction by laying down standards of network performance, which the service provider is required to achieve by proper engineering of his network.
 - (ii) Measure the Quality of Service from time to time and to compare that with the specified norms so as to monitor the level of performance, provided by various service providers' networks.
 - (iii) Protect the interests of subscribers of the ILD services in regard to Quality of Service, particularly a minimum level of voice quality, which he expects, when he makes an ISD call and pays for it.

Section-IV

4. <u>Quality Of Service Benchmarks for ILD networks engineered by employing VoIP technology</u>

4.1 Determination of QoS: -

4.1.1 End-to-End QoS Parameters

¹[The ILD segment of the network will be engineered to meet the following end-to-end service quality parameters;

4.1.1.1 Toll Quality Networks:

• MOS > 4 or R-value of 80 or higher (with advantage factor in case of international satellite link involved in which case MOS > 3.5 or R- value > 70 shall be acceptable).

1 Subs. by the First Amendment Regulations, 2004, reg. 3.1 (w.e.f. 20.01.2004), for the following:

"The ILD segment of the network will be engineered to meet the following end-to-end service quality parameters:

4.1.1.1 Toll Quality Networks:

- $MOS \ge 4$ or R-value of 80 or higher
- One-way end-to-end delay $\leq 150 \text{ ms}$
- Packet loss not to exceed 0.1%
- Jitter should not exceed 5 ms
- Transparency to DTMF tones
- Services covered in addition of voice to include: G3 Fax; voice-band modem @ 14.4 kbps or higher"

- One-way end-to-end delay < 150 ms without satellite link and < 400 ms with satellite link.
- Packet loss not to exceed 0.1%
- Jitter should not exceed 5 ms
- Transparency to DTMF tones
- Services covered in addition of voice to include: G3 Fax; voice band modem @ 14.4 kbps or higher.]

¹[* * * *]

4.1.2 VoIP ILD Equipment Performance:

From the end-to-end performance parameters specified above, the contribution of the ILD segment of the end to end connection in regard to parameters specified in pre-para shall be as indicated below:

4.1.2.1 Toll quality:

- $MOS \ge 4$ or R-value of 80 or higher
- One-way IPTD (IP Packet Transfer Delay) ≤ 50 ms
- Transparency to DTMF tones
- Services covered in addition of voice to include: G3 Fax ; voice-band modem @ 14.4 kbps or higher

²[* * * *]

In both cases, echo-cancellers complying with ITU-T G.168 shall be employed in the ILD segment at each end. The above parameters shall be met for voice traffic even in the presence of other types of traffic such as data/ video.

4.2 Test & Measurement Setup

The measurement set up and methodology is described in the following paragraph.

Apropos the reference network configuration described in Figure-4.1, the measurements required at the maximum traffic handling capacity over various interface spans and the expected results are indicated in Table 4.2. The network configuration should be similar to the actual network deployed by the ILDO in terms of the number of nodes and transmission links. Figure 4.2 depicts the test setup along with the measurements to be made. The national PSTN sections can be realized by using real telephone switches deployed in the network of BSO and NLDO. For

- $MOS \ge 3$ or R-value of 70 or higher
- One-way end-to-end delay $\leq 400 \text{ ms}$
- Packet loss not to exceed 2%
- Jitter not to exceed 10 ms"

2 Del. the following words by the First Amendment Regulations, 2004, reg. 3.2 (w.e.f. 20.01.2004):

- "4.1.2.2 Below Toll quality:
- $MOS \ge 3$ or R-value equal or greater than 70
- One-way IPTD (IP Packet Transfer Delay) $\leq 150 \text{ ms}^{"}$

¹ Del. the following words by the First Amendment Regulations, 2004, reg. 3.2 (w.e.f. 20.01.2004):

[&]quot;4.1.1.2 Below Toll quality Networks:

International section, the two mated setups offered by the ILDO can be connected via looped back transmission links over the actual physical facility -whether fiber or satellite – or via simulateddelay links. In the case of end-to-end measurements, the test equipment is to be connected to reference points X1 and X4 while in the case of VoIP section; measurements could be with reference points X2and X3 (see figure 4.2). In both the cases, the International transmission link has to be looped back.

Two-stage testing is prescribed – lab tests and pre-commissioning (*in situ*) tests. All the measurements as given in the following Table shall be conducted in each testing stage. In the first stage, i.e. lab testing stage, the tests and measurements shall be conducted only on the VoIP set up offered by the ILDO in the lab or at ILDO's actual VoIP installation but the equipment shall not be connected to the PSTN/ISDN/PLMN. In contrast, the pre-commissioning test shall be conducted *in situ* over the VoIP section as well as the entire International connection in the actual inter-connection scenario. It may be noted that the second stage of testing may not be taken up if the equipment does not pass the lab tests and measurements.

Span	Description	Parameter	Value for QOS parameters when international satellite link is not involved	Value for QOS parameters when international satellite link is involved	Method
X ₂ - X ₃	ILD segment employing	Packet Delay	\leq 50 ms	\leq 50 ms	Individual parameters
	VOIP	Jitter	<u>≤</u> 5 ms	<u>≤</u> 5 ms	measured by
	techniques*	Packet Loss	<u>≤</u> 0.1%	<u>≤</u> 0.1%	using IP
	(excluding	R-Value	<u>> 80</u>	<u>> 80</u>	analyzer <u>plus</u>
	transmission	MOS	<u>></u> 4	<u>></u> 4	the aggregate
	delay)				measure (MOS
					or R-Value)

ſ	Table	4.2:	Measurements	Req	uired
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1 Subs. by the Third Amendment Regulations, 2004, reg. 2 (w.e.f. 16.04.2004), for the following: "Table 4 2: Measurements Required

Tuble 4.2. Measurements Required						
Span	Description	Parameter	Value for Toll Quality Service	Value for Below Toll Quality Service	Method	
$X_2 - X_3$	ILD segment	Packet Delay	≤ 50 ms	≤ 150 ms	Individual parameters	
	employing VOIP	Jitter	\leq 5 ms	≤ 10 ms	measured by using IP	
	techniques*	Packet Loss	<u>≤</u> 0.1%	$\leq 2\%$	analyzer <u>plus</u> the	
	(excluding transmission delay)	R-Value	≥ 80	\geq 70	Aggregate measure (MOS or R-value)	
		MOS	\geq 4	≥ 3		
$X_1 - X_4$	End to End International Connection of which ILD segment is a subset*	Delay	≤ 150 ms	≤ 400 ms	Aggregate measure (MOS or R-value)	
		R-Value	≥ 80	\geq 70		
		MOS	≥ 4	≥3		
		Jitter	<u>≤</u> 5 ms	≤ 10 ms		
		Packet Loss	<u><</u> 0.1%	< 2%]	
		CCS7 Signalling Delay	As per Q.709.	Same as for Toll quality.	CCS7 Signalling Analyzer	

* The terminating PSTN and VoIP Sections in another country are outside the purview of our national administration and as such have to be simulated."

X1 - X4	End to End	Delay	\leq 150 ms	\leq 400 ms	Aggregate
	International	R-Value	≥ 80	≥ 70	measure (MOS
which II D	which ILD	MOS	<u>≥</u> 4	≥ 3.5	or R-value)
	segment is a	Jitter	<u><</u> 5 ms	<u><</u> 5 ms	
	subset*	Packet Loss	<u>≤</u> 0.1%	<u>≤</u> 0.1%	
		CCS7 signalling Delay	As per Q.709	As per Q.709	Signalling Analyzer

* The terminating PSTN and VoIP Sections in another country are outside the purview of our national administration and as such have to be simulated.]

For the complete end to end international connection and the ILD segment, commercially available voice quality monitoring tools are recommended for dynamic measurement of the R-value or MOS at the end of each call. PSTN/ISDN/PLMN call samples are automatically collected by these tools and the R-value or MOS are computed in each case. It would be necessary to make sure that during these measurements, the VoIP equipment is loaded up to its rated BHCA with the help of a telephony traffic simulator. In the case of end-to-end tests, the actual interconnection path offered by the service provider should be used and measurements made during the peak and off-peak hours.

In addition to the above, for Toll Quality service end to end through tests are to be conducted by making International telephone calls, sending fax messages, sending DTMF tones to an IVR system and by sending data via voice band modems over the complete end to end International connection described above.

The end-to-end CCS7 signalling delays can be measured using a CCS7 protocol monitor and noting the message time stamp at the first and the last PSTN/ISDN/PLMN node. The values should be within the values mandated in $Q.767^1$ and $Q.709^2$ and calls should be successful.

The individual parameter measurements on the VoIP section should be made with a suitable Internet Protocol analyzer. The traffic up to the maximum traffic handling capacity of the VoIP equipment can be generated with a telephony traffic simulator. The Call Completion Rate (CCR) should be as mandated under Clause 4 (i) (7) of Section IV of the TRAI Regulation on Quality of Service of Basic and Cellular Mobile Telephone Services, (2 of 2000) dated 5th July 2000, as modified from time to time. The packet size and the payload-to header ratio should be varied by employing various speech codecs offered by the Implementation Under Test (IUT) in the gateway.

4.3 Testing Agency

The testing agency shall be the Telecommunication Engineering Centre of Ministry of Communication.

¹ ITU-T Recommendation Q.767, Application of the ISDN user part of CCS-7 for International ISDN connections, Feb-1991.

² ITU-T Recommendation Q.709, Hypothetical Signaling Reference Connection, March 1993.

Section-V

5. Review:

- (i) The QOS parameters given in Section IV may be reviewed by the Authority from time to time.
- (ii) The Authority, on reference from any affected party, and for good and sufficient reasons, may review and modify this Regulation.

Section-VI Residuary Clauses

6. Over-riding Effect:

Wherever higher quality parameter has been stipulated as a condition of license, the QOS as required by the license will over-ride the parameters given herein.

7. Interpretation:

In case of disputes regarding interpretation of any of the provisions of this Regulation, the decision of the Authority shall be final and binding.



Figure 4.1: Reference Network Configuration



Figure 4.2 : Test & Measurement Setup