SCHEDULE 1

VIRTUAL INTERCONNECTION

SCHEDULE 1 – CONTENTS

SCHEDULE 1

VIRTUAL INTERCONNECTION

1. GENERAL

1.1 This Schedule details the establishment of Virtual Interconnection between SingTel's Network and AAT's Network and describes the forecasting and provisioning procedures for Interconnection provided under this Agreement.

1.2 Each Party agrees to interconnect and keep interconnected their respective Network subject to and in accordance with the terms and conditions of this Schedule.

2. INTERCONNECT CONFIGURATION

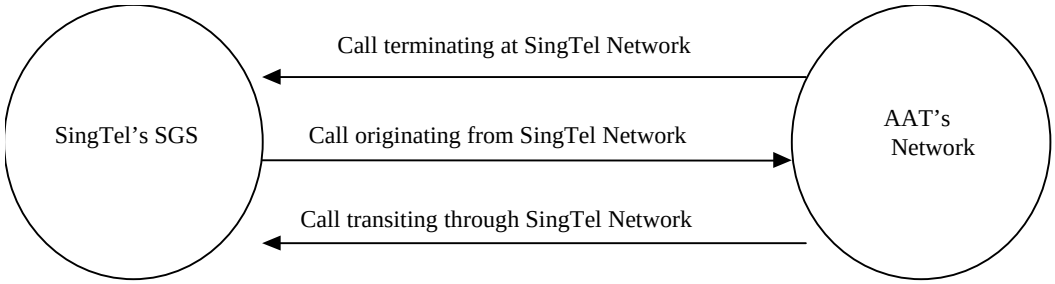
2.1 AAT must interconnect its Network with the SingTel Network at SingTel's SBO Gateway Switches (**SGS**) as specified in Section 2F of Annex A.

2.2 AAT will ensure that the Interconnection Link(s) is provisioned and maintained with sufficient Interconnect Capacity

2.3 The Minimum Interconnection Capacity for Interconnection by AAT to a SingTel SGS is two (2) 2Mbps E1 links. The Minimum Interconnection Capacity for Interconnection by AAT to two (2) of SingTel SGS is four (4) 2Mbps E1 links.

2.4 If AAT interconnects to SingTel's Network with Interconnection Capacity less than the Minimum Interconnection Capacity, AAT acknowledges that the Network performance may not be equivalent to other Licensees which have fulfilled the Minimum Interconnection Capacity for Interconnection to two (2) of SingTel SGSs. SingTel shall not be liable for any degradation in Call handling and/or Network performance experienced by a Requesting Licensee who does not fulfil the Minimum Interconnection Capacity.

2.5 An Interconnection Link may comprise different types of circuit groups. The circuits in each circuit group will convey traffic in a specific direction (one-way) as shown below.



Circuit Groups

2.6 AAT is responsible for the correct dimensioning of the type of circuit groups (one-way) that it will require.

2.7 If SingTel or AAT wish to interconnect at additional POIs, that Party may request an additional POI and the Party will negotiate in good faith in relation to the interconnect configuration applicable to such additional Interconnection.

2.8 Neither Party shall be required to commence works on the installation of Network Facilities to support a new POI under clause 2.7 until the Party has agreed on the interconnect configuration under clause 2.7.

3. POINT OF INTERCONNECTION

3.1 The Parties agree that the POI will be located on AAT's side of the SingTel SGS Digital Distribution Frame (**DDF**) at the SingTel SGS

3.2 Each Party is responsible for the provisioning and maintenance of Network Facilities on its "side" of the POI.

3.3 The Interconnection between SingTel's SGS and AAT's Network located at a different site will be implemented by means of 2Mbps E1 Interconnection Link.

3.4 AAT will be responsible for connection of the Interconnection Link between the POI and AAT's Network.

3.5 AAT shall acquire Local Leased Circuits from SingTel to form part of the Interconnection Link. AAT's leased Interconnection Link consists of multiple 2 Mbit/s (E1) circuits terminated at a DDF in SingTel's designated SGS exchanges.

3.6 At AAT's request, SingTel will lease Local Leased Circuits at 2Mbps (E1) G.703 interface to AAT's Network based on SingTel's standard terms and conditions for supply of Local Leased Circuit service.

4. TECHNICAL REQUIREMENTS AND SPECIFICATIONS

4.1 Transmission

4.1.1 The Parties' shall interface at 2Mbps level and comply with ITU-T Rec G.703.

4.2 Signalling

4.2.1 The Parties shall comply with CCS SS7 – MTP and ISUP as specified in Section 1 of Annex A.

4.2.2 AAT shall obtain its own Signalling Point Code from the Authority.

4.2.3 Both Parties shall provide CCS SS7 Signalling Link(s) up to a maximum of two (2) SS7 Signalling Links for Interconnection to one (1) SGS (or POI) or a maximum of four (4) SS7 Signalling Links for Interconnection to two (2) SGSs.

4.2.4 AAT shall adhere to additional SS7 signalling requirement as follows:

(a) The Calling Party Number and Redirecting Number if available shall be conveyed for all Calls across the Network connection without manipulation.

(b) The Dummy CLI received from Mobile Operators from inbound and outbound roamers shall be an eight (8) digit Dummy CLI.

(c) The number dialled by the Calling subscriber shall not be changed or amended for the routing of international outgoing Calls from AAT's Network to SingTel's SGS. In addition, the NOA of the Called Party Number shall be set to 'International'.

(d) AAT's Network shall deliver international incoming Calls to SingTel's SGS terminating to SingTel's domestic Network with the international Call indicator 'A' bit of FCI parameter of the IAM set to '1'.

4.3 Interconnect Testing – Transmission and Signalling

4.4 Routing

4.4.1 The Parties shall route Interconnected Calls in accordance with the agreed arrangements applicable to that Call Type under Schedule 2.

4.4.2 The Parties shall comply with the testing procedures in Section 1A of Annex A.

5. DECOMMISSIONING

5.1 Subject to clauses 5.2 and 5.3, a Party (Decommissioning Party) may, for whatever reason: upon giving no less than six (6) months prior written notice to the other Party, close, replace or relocate any SGS in respect of which an Interconnection Link is connected, decommission an Interconnection Link or close a POI (**Decommissioning**).

5.2 (a) Subject to paragraph (b), the Decommissioning Party shall be responsible for and bear all direct costs incurred by both Parties in carrying out the Decommissioning. Direct costs are limited to removal of equipment and cabling at the POI.

(b) When the request for Decommissioning is at the direction of a Third Party pursuant to a legal obligation binding on the Decommissioning Party, each Party shall bear its own costs associated with the Decommissioning together with the direct costs incurred by that Party in respect of the establishment of alternative arrangements necessary to support the provision of Origination and Termination Services provided at the time of the Decommissioning.

5.3 (a) If AAT is the Decommissioning Party, AAT must also comply with the terms and conditions under which Local Leased Circuits are supplied by SingTel (including the payment of any early termination payments).

(b) If SingTel is the Decommissioning Party, SingTel will arrange for the connection of Local Leased Circuits to the alternative or relocated SGS, Interconnection Link or POI.

5.4 Upon the happening of an event which causes or is likely to cause significant damage to the Network of a Party or which endangers or is likely to endanger the health or safety of any person (**Emergency Event**), that Party may close or replace its SGS or a POI or Decommission an Interconnection Link without prior written

notice to the other Party, provided that it notifies the other Party as soon as practicable that the Emergency Event exists and that emergency relocation work is being or will be performed. The costs incurred as a result of, or in connection with, a closure or replacement of an SGS or a POI or Decommissioning of an Interconnection Link under this sub-clause shall be borne as follows:

(a) where the occurrence of the Emergency Event was caused as a result of the negligence or wilful act or omission of the first-mentioned Party, its employees, agents or contractors, that Party shall bear the costs associated with the closure, replacement or Decommissioning incurred by both Parties; and

(b) where the occurrence of the Emergency Event was not caused as a result of the negligence or wilful act or omission of the first-mentioned Party, its employees, agents or contractors, each Party shall bear its own costs associated with the closure, replacement or Decommissioning.

6. FORECASTING AND PROVISIONING OF INTERCONNECT CAPACITY

6.1 This section applies to Forecasts to be provided by AAT to SingTel in relation to Interconnect Capacity, if AAT reaches a minimum Interconnect Capacity of sixty-three (63) E1s for Interconnection with the SingTel Network.

6.2 On or near 1 February and 1 August of each year, AAT will advise SingTel of its estimates of the total number of lines or Subscribers in service on 1 April and 1 October in that year and each subsequent year for three (3) years.

6.3 AAT shall provide to SingTel the Forecasts for Interconnect Capacity on or near 1 March and 1 September of each year and in the formats in Annex C.

6.4 The Forecasts shall be for periods commencing six (6) months from 1 April and 1 October (**Forecast Date**) respectively, and be for a period of thirty-six (36) months, in intervals of six (6) months for the first twelve (12) months, and yearly thereafter.

6.5 SingTel will respond to a Forecast within fifteen (15) Business Days, or any other period as may be agreed, of receipt, or such other period as may be agreed. The response shall be either:

(xii) an acknowledgment that SingTel is able to provide the Forecasted Interconnect Capacity in the first six (6) month period by a particular date (**Forecast Delivery Date**); or

(xiii) an acknowledgment that SingTel is able to provide the Forecasted Interconnect Capacity in the first six (6) month period, but not be able to provide those quantities in accordance with the Forecast timeframes; or

(xiv) an advice that SingTel is unable to provide the Forecasted Interconnect Capacity in the first six (6) month period as procurement is required.

6.6 Where procurement is required in order to meet the Forecast, and SingTel advises AAT pursuant to clause 6.5, SingTel shall seek confirmation of the Forecast. As a general guide, the provisioning time is twelve (12) months from the Forecast Date.

6.7 If AAT seeks Interconnect Capacity at a level other than the Forecast level or on a Forecast Delivery Date other than the agreed Forecast Delivery Date, it may make a request to SingTel to study the feasibility of such a request. SingTel shall undertake a feasibility study, and AAT shall pay a fee to SingTel for the conduct of the study in response to the request.

6.8 SingTel is under no obligation to provide Interconnect Capacity other than in accordance with the accepted Forecast Capacity requirements and the accepted Forecast delivery times.

6.9 Clauses 6.7 and 6.8 are not intended to create a process in substitution for the normal Forecasting process.

6.10 AAT agrees that:

(a) the first six (6) months of each Forecast given by it is a commitment for the full quantity of Interconnect Capacity which SingTel will necessarily supply or install in order to meet that Forecast; and

(b) it will pay all Charges calculated in accordance with Schedule 3 relating to its failure to utilise the full quantity of Interconnect Capacity in relation to a Forecast; and

(c) there shall be no variation in the Forecasts as provided to SingTel.

6.11 If, following acceptance of a Forecast, SingTel is unable to provide the Interconnect Capacity or provide the Interconnect Capacity by the Forecast Delivery Date, in whole or part, it must advise AAT promptly and both Parties must, where practicable, negotiate a new delivery timetable. To assist in this negotiation, SingTel may offer alternatives where available.

6.12 Delivery of Forecast Capacity

6.12.1 The Parties shall use their reasonable endeavours to ensure that its Network Facilities on its side of the relevant POI are provisioned on the Forecast Delivery Date (or as otherwise agreed) and maintained in accordance with the Forecasts.

6.12.2 Delivery of Interconnect Capacity is taken to occur on the Forecast Delivery Date or on such other date as may be agreed (as applicable) provided that SingTel has advised that it is ready to commence testing.

6.13 Recovery for Over Forecasting

6.13.1 If the Actual Usage by AAT of the Interconnect Capacity is greater than or equal to ninety (90) percent of its Forecast, no over forecasting Charges shall apply.

6.13.2 If the Actual Usage by AAT of the Interconnect Capacity is less than ninety (90) percent of its Forecast, AAT shall pay the amount calculated in accordance with Schedule 3 until such time as Actual Usage reaches ninety (90) percent.

6.13.3 Where a dispute arises in respect to the Actual Usage under this clause 6.13, the matter will be referred for resolution in accordance with Schedule 5 – Dispute Resolution.

6.14 Virtual Interconnection

6.14.1 Notwithstanding anything else in this clause 6, AAT remains responsible for Forecasting and ordering sufficient Capacity on Local Leased Circuits under SingTel's standard terms and conditions to enable Virtual Interconnection to occur.

SCHEDULE 1

ANNEXURES

ANNEX A

INTERFACE SPECIFICATIONS

ANNEX A - INTERFACE SPECIFICATIONS

SECTION 1 - CCS NO. 7 SIGNALLING SYSTEM (MTP & ISUP) INTERFACE SPECIFICATION

1 General

1.1 The CCS No.7 signalling system interface shall conform to the followings:

(A) MTP following ITU-T Rec. Q.701 - Q.707

(B) ISUP following ITU-T Rec. Q.761 to Q.764 and Q.766 to support basic bearer service and supplementary services for voice and non-voice applications.

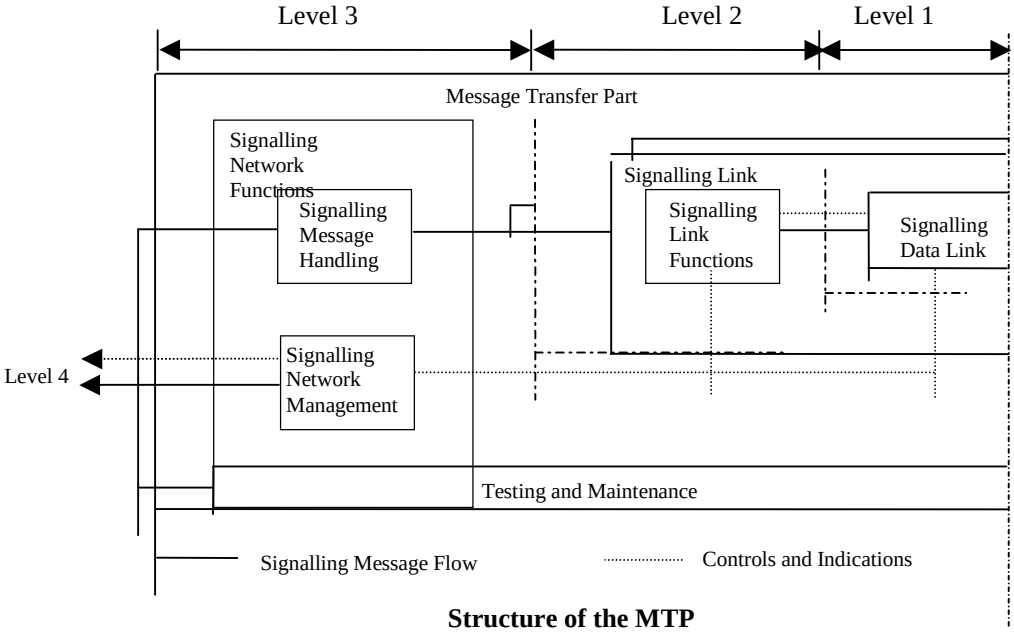
1.2 The interface specifications contained in this section are based on the interworking with SingTel's IGS/SGS.

A MTP

A1 STRUCTURE

A1.1 The MTP shall conform to ITU-T Rec. Q.701-Q.707 and comprise three (3) functional levels. The structure of the MTP is given below.

A1.2 The MTP shall allow messages from all different users to be transferred on the same signalling data link.



Structure of the MTP

A2 SIGNALLING DATA LINK (MTP LEVEL 1)

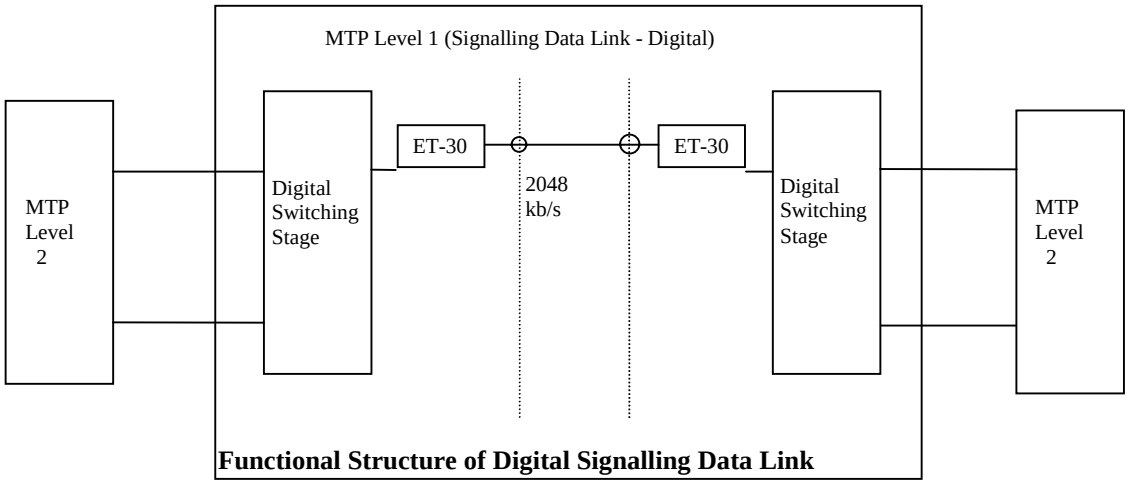
A2.1 Requirements

The Signalling data link (MTP Level 1) shall conform to all requirements as specified in ITU-T Rec. Q.702.

A2.2 Digital Signalling Data Link

(a) A signalling data link shall comprise two (2) data channels operating together in opposite direction at 64 kbit/s which are to be derived from TS 16 of the 2048 kbit/s PCM interface.

(b) The data link shall be assignable by command to any speech channel (time-slot) available for 64 kbit/s user transmission if the standard time slots are not available.



Functional Structure of Digital Signalling Data Link

A3 SIGNALLING LINK (MTP LEVEL 2)

A3.1 Requirements

(a) The Signalling Link (MTP Level 2) shall fulfil the requirements in all details as specified by ITU-T Rec. Q.703.

(b) A single flag shall be sent between consecutive signal units. However, the exchange shall accept signal units which are delimited by either single flag or multiple flags.

(c) The Signalling Link shall cater for messages with signalling information field of up to 272 octets. This shall allow a single message signal unit to accommodate information blocks of up to 268 octets in length accompanied by a routing label.

(d) The spare bits in the link status signal unit shall be coded '0' and shall be ignored at the receiving side.

(e) Basic error correction method shall be used in the exchange. For satellite connection, the preventive cyclic method shall be available.

(f) Level 2 flow control shall be provided in the exchange. The exchange shall prevent excessive oscillation between congested and non-congested states when performing level 2 flow control.

(g) The values of MTP level 2 timers (T1-T7) shall conform to the range/values as specified in ITU-T Rec. Q.703.

A4 SIGNALLING NETWORK FUNCTIONS (MTP LEVEL 3)

A4.1 Requirements

The MTP Level 3 shall include all signalling Network functions as specified in ITU-T Rec. Q.704.

A4.2 Signalling Message Handling

(a) Message Routing Function

(i) The routing label shall be used for message routing purpose. It shall be possible to provide different routing plans for different user parts using the service indicator.

(ii) The routing (eg outgoing Signalling Link sets) for a specific routing label shall be defined by command.

(iii) The exchange acting as SP shall provide load-sharing between links belonging to the same link set for signalling traffic to be sent to a particular signalling point.

(b) Message Distribution

The exchange shall analyse the service indicator to determine which user part the message is to be delivered.

(c) Message Discrimination

The exchange shall analyse the destination code in the routing label to determine whether or not the exchange is the destination point of that message as specified in ITU-T Rec. Q.704.

A4.3 Signalling Network Management

(a) SP Congestion

The exchange shall conform to ITU-T Rec. Q.704 procedures to detect and handle SP congestion.

(b) Signalling Network Congestion

Based on the international standard of flow control, the congestion status shall be provided for the indication of Signalling Link or signalling route set congestion. Under normal operation, when the Signalling Link or route set is not congested, the congestion status is assigned the zero value. The setting of congestion onset, abatement and discard threshold(s) shall conform to ITU-T Rec. Q.704.

A4.4 Signalling Traffic Management

(a) Under normal situations, the signalling traffic to be sent to a particular signalling point shall be evenly distributed over all available Signalling Links within the link set.

(b) Changeover/Changeback

(i) It must be ensured that no messages are lost, duplicated or out-of-sequence during the changeover/changeback procedures. In the changeover procedure, the exchange shall divert traffic pertaining to that unavailable Signalling Link to the alternative Signalling Link with next priority in the same link set.

(ii) Time-controlled changeover shall be used when the exchange of a changeover message is not possible or not desirable.

(c) Management Inhibiting

(i) The exchange shall provide Signalling Link management inhibiting function for maintenance or testing purposes (eg if the link experiences too many changeovers/changebacks in a short time, or there is a significant link error rate). A Signalling Link marked 'inhibited' shall be unavailable to user part-generated signalling traffic. The management inhibit action shall not cause any link status change at level 2. Inhibit request shall only be granted provided that the inhibit action does not cause any previously accessible destinations to become inaccessible at either end of the Signalling Link.

(ii) Periodic tests shall be made on the inhibit status of inhibited links. Such periodic tests shall not overload the signalling Network and shall not perform at signalling point restart.

(iii) Uninhibiting shall be initiated either by management function or by routing functions at either end of the Signalling Link.

(d) Signalling Traffic Flow Control

(i) The exchange shall provide signalling traffic flow control function to limit signalling traffic at its source in the case when the signalling Network is not capable of transferring all traffic offered by the user because of Network failures or congestion situation. The flow control shall also initiate resumption of the normal traffic flow when the normal transfer capability is restored.

(ii) When a MTP user has become unavailable, the MTP shall not send a User Part Unavailable (UPU) message to the partner signalling points. Signalling messages received from the affected user parts of the partner signalling points shall be discarded as long as the local user parts are not available.

Upon receipt of an UPU message, the MTP shall discard these messages.

A4.5 Signalling Link Management

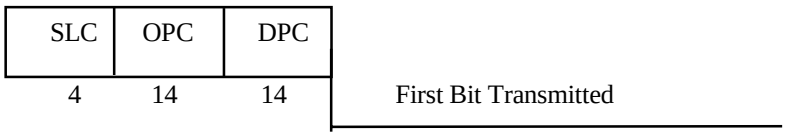
The exchange shall provide Signalling Link management functions as specified in ITU-T Rec. Q.704 to initiate and control actions aimed at restoring the normal availability of links and link sets.

A4.6 Signalling Route Management

The exchange shall provide signalling route management function as specified in ITU-T Rec. Q.704 to transfer information about changes in the availability of signalling routes in the Signalling Network so as to enable remote SPs to take appropriate signalling traffic management actions.

A4.7 Format and Codes of Signalling Network Management Messages

(a) The label structure of Signalling Network Management messages coincides with the standard routing label as follows:



The Signalling Link Code (SLC) indicates the Signalling Link, connecting the destination and originating points, to which the message is related. It is coded as 0000 for messages not related to a Signalling Link.

(b) A list of heading code allocation of Signalling Network Management messages are given in Table A4-1 hereof.

A4.8 Time-out Values and Tolerances

The exchange shall provide MTP level 3 timers with values and tolerances as specified in ITU-T Rec. Q.704.

Message	H1 H0	0000	0001	0010	0011	0100	0101	0110	0111	1000	1001	1010	1011	1100	1101	1110	1111
	0000																
CHM	0001		COO	COA			CBD	CBA									
ECM	0010		ECO	ECA													
FCM	0011			TFC													
TFM	0100		TFP				TFA										
RSM	0101		RST														
MIM	0110		LIN	LUN	LIA	LUA	LID	LFU	LLT	LRT							
TRM	0111		TRA														
	1000																
	1001																
	1010																
	1011																
	1100																
	1101																
	1110																
	1111																

CHM = Changeover and Changeback Message
ECM = Emergency Changeover Message
RSM = Signalling-route-set-test Messages
TRM = Traffic-restart-allowed Messages
FCM = Signalling-Traffic-flow-control Messages
TFM = Transfer-prohibited-transfer-allow-transfer-restricted Messages
MIM = Management-inhibit Messages

Table A4-1: Heading Code Allocation of Signalling Network Management Messages

A5 MESSAGE TRANSFER PART SIGNALLING PERFORMANCE

A5.1 The exchange MTP shall achieve a signalling performance as specified in ITU-T Rec. Q.706.

A5.2 The exchange CCS No.7 signalling equipment shall be highly reliable such that not more than one in 10^7 messages will be lost due to failure in the MTP. Other performance factors (eg changeover performance times, etc.) shall also be conformed.

A6 TESTING AND MAINTENANCE

A6.1 The exchange shall provide Signalling Network testing and maintenance requirements as specified in ITU-T Rec. Q.707.

A6.2 An on-line Signalling Link test shall be applicable to Signalling Link to be activated or restored. The Signalling Link becomes available only if the test is successful. It shall be able to send Signalling Link test messages at regular intervals on an in-service Signalling Link using command.

A6.3 The Signalling Network testing and maintenance messages shall be carried on the signalling channel in message signal units with service indication '0001'.

A6.4 The exchange shall provide timers in ITU-T Rec. Q.707.

B ISUP SPECIFICATION

B1 ISUP Addressing

The ISUP address structure shall be capable of handling E.164 addresses in the Calling and Called number, and re-directing address information elements.

B2 Message Formats and Codes

B2.1 Basic ISUP messages and signalling information as defined in ITU-T Rec. Q.762 with their formats and contents as specified in ITU-T Rec. Q.763 shall be provided. The encoding of the message types and parameter names are as shown in Table B2-1 and Table B2-2 respectively. Additional ISUP messages and parameters may be provided to support ISDN supplementary services defined in ITU-T Rec.73X-series.

Message Type			Coding
ACM	-	Address complete	0 0 0 0 0 1 1 0
ANM	-	Answer	0 0 0 0 1 0 0 1
BLO	-	Blocking	0 0 0 1 0 0 1 1
BLA	-	Blocking acknowledgment	0 0 0 1 0 1 0 1
CPG	-	Call progress	0 0 1 0 1 1 0 0
CGB	-	Circuit group blocking	0 0 0 1 1 0 0 0
CGBA	-	Circuit group blocking acknowledgment	0 0 0 1 1 0 1 0
CQM	-	Circuit group query	0 0 1 0 1 0 1 0
CQR	-	Circuit group query response	0 0 1 0 1 0 1 1
GRS	-	Circuit group reset	0 0 0 1 0 1 1 1
GRA	-	Circuit group reset acknowledgment	0 0 1 0 1 0 0 1
CGU	-	Circuit group unblocking	0 0 0 1 1 0 0 1
CGUA	-	Circuit group unblocking acknowledgment	0 0 0 1 1 0 1 1
CFN	-	Confusion	0 0 1 0 1 1 1 1
CON	-	Connect	0 0 0 0 0 1 1 1
COT	-	Continuity (receive only)	0 0 0 0 0 1 0 1
FAA	-	Facility accepted	0 0 1 0 0 0 0 0
FRJ	-	Facility reject	0 0 1 0 0 0 0 1
FAR	-	Facility request	0 0 0 1 1 1 1 1
FOT	-	Forward transfer	0 0 0 0 1 0 0 0
INF	-	Information	0 0 0 0 0 1 0 0
INR	-	Information request	0 0 0 0 0 0 1 1
IAM	-	Initial address	0 0 0 0 0 0 0 1
REL	-	Release	0 0 0 0 1 1 0 0
RLC	-	Release complete	0 0 0 1 0 0 0 0
RSC	-	Reset circuit	0 0 0 1 0 0 1 0
RES	-	Resume	0 0 0 0 1 1 1 0
SAM	-	Subsequent address	0 0 0 0 0 0 1 0
SUS	-	Suspend	0 0 0 0 1 1 0 1
UBL	-	Unblocking	0 0 0 1 0 1 0 0
UBA	-	Unblocking acknowledgment	0 0 0 1 0 1 1 0
USR	-	User-to-user information	0 0 1 0 1 1 0 1
SingTel internal use			1 1 1 1 1 1 1 1

Table B2-1 : Encoding of the Message Types

Parameter Name	Coding
Access delivery information	0 0 1 0 1 1 1 0
Access transport	0 0 0 0 0 0 1 1
Automatic congestion level	0 0 1 0 0 1 1 1
Backward Call indicators	0 0 0 1 0 0 0 1
Call diversion information	0 0 1 1 0 1 1 0
Called Party number	0 0 0 0 0 1 0 0
Calling Party number	0 0 0 0 1 0 1 0
Calling Party's category	0 0 0 0 1 0 0 1
Cause indicators	0 0 0 1 0 0 1 0
Circuit group supervision message type indicator	0 0 0 1 0 1 0 1
Circuit state indicator	0 0 1 0 0 1 1 0
Connected number	0 0 1 0 0 0 0 1
Continuity indicators (received only)	0 0 0 1 0 0 0 0
End of optional parameters	0 0 0 0 0 0 0 0
Event information	0 0 1 0 0 1 0 0
Forward Call indicators	0 0 0 0 0 1 1 1
Information indicators	0 0 0 0 1 1 1 1
Information request indicators	0 0 0 0 1 1 1 0
Nature of connection indicators	0 0 0 0 0 1 1 0
Optional backward Call indicators	0 0 1 0 1 0 0 1
Optional forward Call indicators	0 0 0 0 1 0 0 0
Original called number	0 0 1 0 1 0 0 0
Range and Status	0 0 0 1 0 1 1 0
Redirecting number	0 0 0 0 1 0 1 1
Redirection number	0 0 0 0 1 1 0 0
Redirection number restriction	0 1 0 0 0 0 0 0
Redirection information	0 0 0 1 0 0 1 1
Signalling Point Code	0 0 0 1 1 1 1 0
Subsequent number	0 0 0 0 0 1 0 1
Suspend/Resume indicators	0 0 1 0 0 0 1 0
Transmission medium requirement	0 0 0 0 0 0 1 0
User service information	0 0 0 1 1 1 0 1
User-to-user indicators	0 0 1 0 1 0 1 0
User-to-user information	0 0 1 0 0 0 0 0
SingTel internal use	1 1 1 0 1 1 1 1 1 1 1 1 0 0 1 1 1 1 1 1 0 1 0 0 1 1 1 1 1 0 0 1 1 1 1 1 1 0 1 0 1 1 1 1 1 1 0 1 1 1 1 1 1 1 0 1 1 1 1 1 1 1 0 0

Table B2-2 : ISUP Parameter Name Codes

B2.2 The exchange shall conform to the default interpretations of recognised parameters which contain codes currently indicated as being spare in ITU-T Rec. Q.763. This is to ensure that the exchange shall be able to interwork with a future version of ISUP. The procedures for handling of the unrecognised parameter values shall follow ITU-T Rec. Q.764 and the guidelines for handling of unrecognised information follow ITU-T Rec. Q.767.

B3 Network Features

B3.1 The exchange shall make an automatic repeated attempt:

(a) on detection of dual seizure if it is the non-control exchange;

(b) on receipt of BLO after sending an IAM and before any backward message has been received;

(c) on receipt of a RSC after sending an IAM and before a backward message has been received; and

(d) on receipt of an unreasonable message during Call setup.

B3.2 The exchange shall provide BLO (UBL) message and CGB (CGU) to permit the switching system or maintenance system to remove (return) traffic from (to) the distant terminal(s) of a circuit or group of circuits because of a fault or to permit testing.

B3.3 Circuit group query test procedure shall be provided in the exchange to audit the circuit state on a demand or routine basis. The range field of the CQM shall range from N=0 (single circuit) to maximum 31. If this value is exceeded the CQM message shall be discarded.

B4 Basic Signalling Procedures

B4.1 The exchange shall conform to all requirements as specified in ITU-T Rec. Q.764 whether acting as an originating exchange, a transit exchange or a destination exchange.

B4.2 Normal Call Release

The exchange shall provide release procedures based on a two (2) message (release, release complete) approach.

(a) The Release Message (**REL**) initiates release of the circuit switched connection and it is required that the circuit is re-selectable from the subsequent exchange within the cross-office transfer time, TCU as specified in ITU-T Rec. Q.766.

(b) Release Complete Message (**RLC**) shall be returned to the preceding exchange when the switched path is released and the circuit is re-selectable.

B4.3 Abnormal Condition

The exchange shall conform to ITU-T Rec. Q.764 on the handling of abnormal conditions which are listed below:

(a) dual seizure;

(b) transmission alarm handling for digital inter-exchange circuits;

(c) reset of circuits and circuit groups;

(d) failure in the blocking/unblocking sequence;

(e) receipt of unreasonable signalling information message which includes:

(i) unexpected messages,

(ii) unrecognised signalling information messages,

(iii) unrecognised parameters,

(iv) unrecognised parameter values (both mandatory and optional),

the exchange shall recognise all messages specified in Table B2-1;

(f) failure to receive a RLC message;

(g) failure to receive a response to an INR message; and

(h) unable to release in response to a REL message.

B4.4 Automatic Congestion Control (**ACC**)

(a) When the exchange reaches the overload condition, an automatic congestion level parameter shall be added to all REL message generated by the exchange. This parameter indicates the level of congestion to the adjacent exchanges which in turn, shall reduce their traffic to the overload affected exchange.

(b) If the exchange returns to its normal traffic load, it shall stop including automatic congestion level parameters in REL messages. The adjacent exchanges shall then, after a predetermined time, automatically return to their normal status.

B5 Time Supervision

The exchange shall provide all timers specified in ITU-T Rec. Q.764.

B6 Performance of ISUP

The exchange shall achieve the ISUP signalling performance requirements as specified in ITU-T Rec. Q.766.

ISUP ATTACHMENT

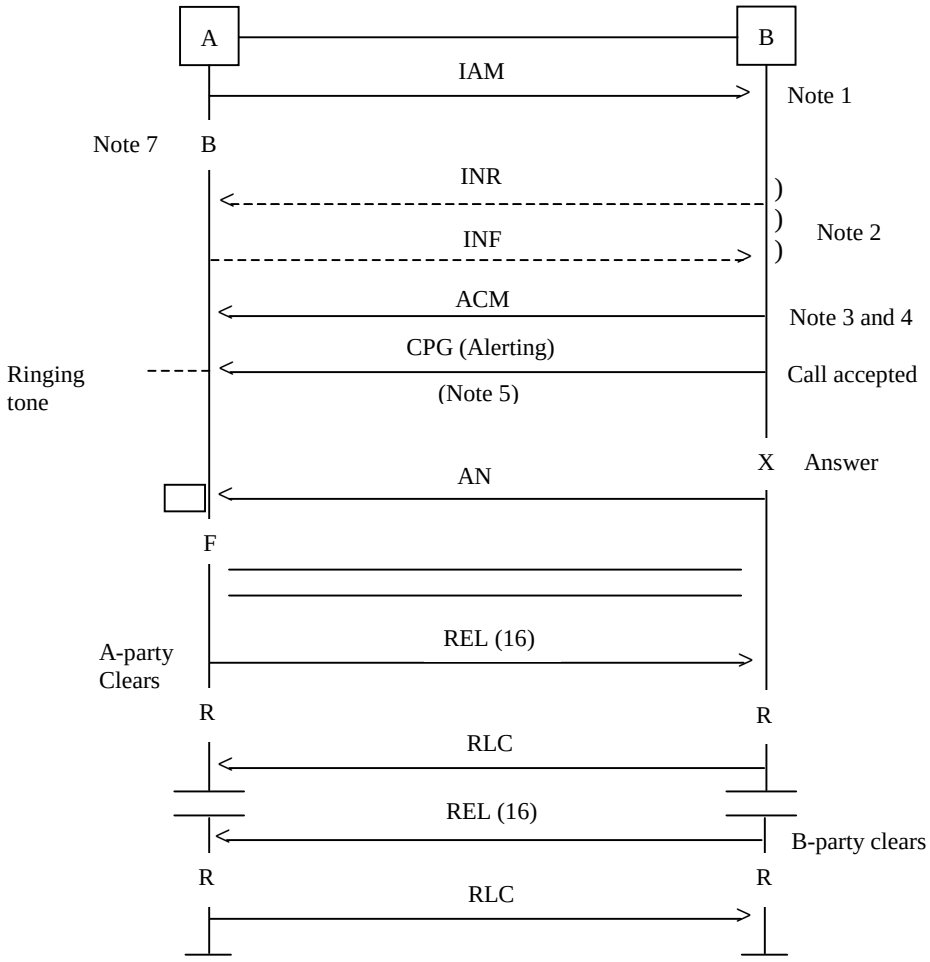
1 ISUP CALL SET-UP SEQUENCE

The ISUP Call set-up sequence for the following traffic cases (all paths using CCS ISUP) are captured in this Attachment. The traffic cases shown here are not exhaustive.

The setting of parameters and indicators shown in this Attachment are tentative and subject to changes later.

Item No.	Description
1.1	Local Call - Successful Call Set-up Enbloc Operation; Terminating Access is ISDN
1.2	Enbloc Operation; Terminating Access is non-ISDN
1.3	Unsuccessful Call Set-up
1.4	Suspend and Resume Procedure

No.	1.1/1	Local Call – Successful Call Setup
		Enbloc Operation; Terminating access is ISDN



Legend :

X	Switch path completed in both direction
R	Remove switch path
B	Switch path completed in backward direction
F	Switch path completed in forward direction
▭	Charging begins (if applicable)

No.	1.1/2	Local Call – Successful Call Setup
		Enbloc Operation; Terminating access is ISDN

Note 1 : (a) CLI is always included in the IAM with an appropriate Presentation/Restriction/Not Available indication for the following traffic cases:-

- All originating Calls
- Rerouted Calls
- Transit Calls (Previous path is CCS-ISDNUP)

When the CLI is provided by the Network, the originating exchange sets the screening indicator to "Network provided". When the CLI is provided by the user or ISPBX, it is verified or screened for validity by the Network. If the user provided CLI is valid, the screening indicator is set to "user provided verified and passed". If the user provided CLI is not valid, the originating exchange defaults to the Network provided CLI.

(b) The "ISDN User Part Preference Indicator" contained within the "Forward Call Indicator" of IAM is derived at the exchange from the bearer capability and supplementary service request contained in the Q.931 SETUP message.

(c) The allocation of "Calling Party's Category" parameter field in the IAM is:-

0 0 0 0 0 0 0 0	Calling Party's category unknown at this time
0 0 0 0 0 0 0 1	Operator, language French
0 0 0 0 0 0 1 0	Operator, language English
0 0 0 0 0 0 1 1	Operator, language German
0 0 0 0 0 1 0 0	Operator, language Russian
0 0 0 0 0 1 0 1	Operator, language Spanish
0 0 0 0 0 1 1 0	)
0 0 0 0 0 1 1 1	) Reserved
0 0 0 0 1 0 0 0	)
0 0 0 0 1 0 0 1	SingTel internal use
0 0 0 0 1 0 1 0	Ordinary Calling Party
0 0 0 0 1 0 1 1	Calling Party with priority
0 0 0 0 1 1 0 0	Data Call (Voice band data)
0 0 0 0 1 1 0 1	Test Call
0 0 0 0 1 1 1 0	Spare
0 0 0 0 1 1 1 1	Payphone
0 0 0 1 0 0 0 0	)
to	) Spare
1 1 0 1 1 1 1 1	)
1 1 1 0 0 0 0 0	Operator without trunk offering facility
1 1 1 0 0 0 0 1	Coinafon

No.	1.1/3	Local Call - Successful Call Setup
		Enbloc Operator; Terminating Access is ISDN

1 1 1 0 0 0 1 0	SingTel internal use
1 1 1 0 0 0 1 1	SingTel internal use
1 1 1 0 0 1 0 0	SingTel internal use
1 1 1 0 0 1 0 1	PBX
1 1 1 0 0 1 1 0	PBX with priority
1 1 1 0 0 1 1 1)	
1 1 1 0 1 0 0 0)	
1 1 1 0 1 0 0 1)	SingTel internal use
1 1 1 0 1 0 1 0)	
1 1 1 0 1 0 1 1)	
1 1 1 0 1 1 0 0)	
to)	Spare for national use
1 1 1 1 1 1 1 0)	
1 1 1 1 1 1 1 1	Spare

Note 2: INR/INF invoked (by end-to-end signalling) for Calls to transfer Information not included in the IAM

Note 3: When no status indication has been received from the ISDN access prior to the destination exchange determining that the complete Called Party number has been received, the indicators in the ACM will be set as follows:-

- Called line status = 'NO INDICATION'
- ISDN access indicator = 'ISDN'

Subsequently, the indication that the destination user is being alerted is transferred in a CPG message which contains an Event indicator set to 'Alerting".

Note 4: When connections are set-up to terminals having an automatic answer feature, the alerting indication may not be received from the Called Party. If a destination exchange receives an answer indication a ANM message is sent provided that an ACM has been sent; otherwise the CON message is sent (This CON then signifies both address complete and answer conditions). Indicators in CON will indicate 'SUBSCRIBER FREE' and 'ISDN' access. The destination exchange will through connect before CON is sent.

No.	1.1/4	Local Call - Successful Call Setup
		Enbloc Operation; Terminating access is ISDN

Note 5 : For telephone Calls within the ISDN, ringing tone will be applied by the terminating exchange as soon as it is known that the subscriber is free.

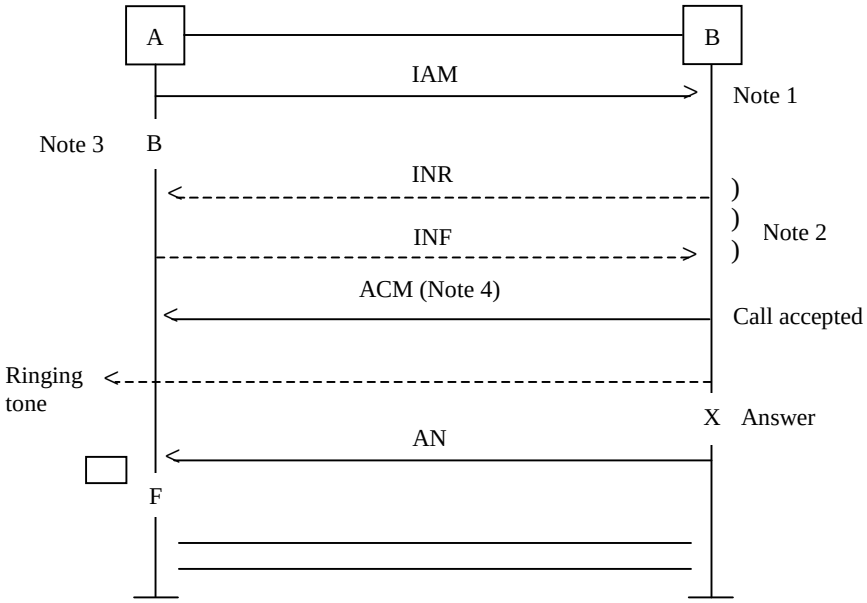
Note 6 : The Answer message received from the destination exchange shall carry a charging indication.

Note 7 : Completion of Transmission Path

(a) At the originating exchange, on speech or 3.1 kHz audio Calls, through-connection of the transmission path will be completed in both directions immediately after the IAM has been sent. For other connection types, through-connect of the transmission path will be completed in the back-ward direction (The transmission path is completed in the forward direction on receipt of a CON or ANM message) immediately after the sending of IAM.

(b) At the intermediate exchange (No interworking encountered), through-connection of the transmission path in both directions will be completed after IAM has been sent.

No.	1.2	Local Call - Successful Call Setup
		Enblock Operation; Terminating access is non-ISDN



Note 1) See Case 1 of item 1.1
Note 2)

Note 3 : See Note 7 of Case 1, item 1.1

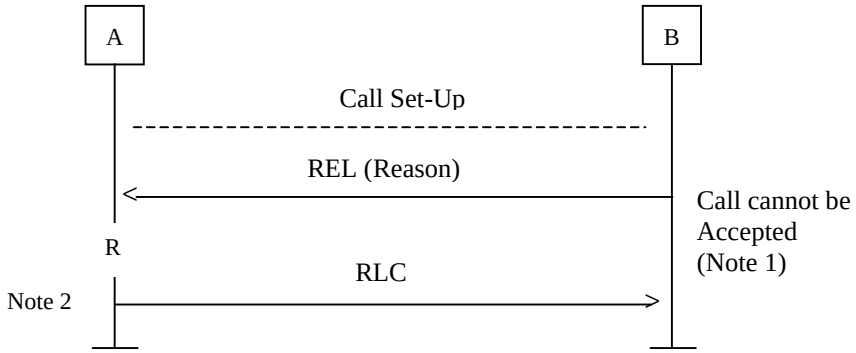
Note 4 : (a) An ACM shall be sent as soon as the destination exchange has determined that the complete Called Party number has been received, and the destination exchange established that the subscriber is free.

Indicators in the ACM will be set to indicate:-
- Called line status = 'SUBSCRIBER FREE'
- ISDN access indicator = 'NON ISDN'

(b) In the case of a PBX, an ACM shall be sent as soon as it has determined that the complete Called Party number has been received.

Indicators in ACM will be set to indicate:-
- Called line status = 'NO INDICATION'
- ISDN access indicator = 'NON ISDN'

No.	1.3/1	Unsuccessful Call Setup



Note 1: If at any time in the Call setup the connection cannot be completed a release message which contains the reason is returned. The initiating exchange shall release the switched path (if established).

Note 2: On receipt of REL, the originating exchange releases the Switch path. In addition, it (if applicable)

(a) returns an indication (in-band or out-band) to the Calling Party;

(b) attempts to re-route the Call setup.

When the A-exchange is ready for circuit re-selection, a RLC is sent to the succeeding exchange.

Note 3: Some Unsuccessful Calls are listed below:-

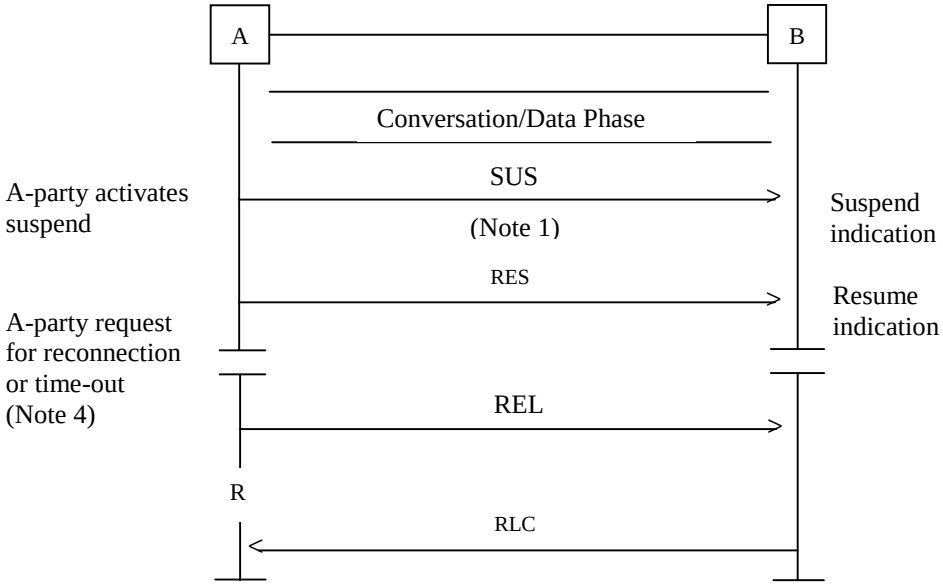
	Call Conditions	Cause Value
(a)	Unallotted (unassigned); Temporary disconnect; Terminating denied	1
(b)	Subscriber line busy, no Call waiting service or engaged in CW service	17
(c)	Ringing no reply	19
(d)	Calls rejected or not accepted due to SCR, SCA or incoming access bar service	21
(e)	Address incomplete	28

No.	1.3/2	Unsuccessful Call Setup

	Call Conditions	Cause Value
(f)	Subscriber line blocked for maintenance and operational reasons	27
(g)	All outgoing routes/trunks busy (at tandem exchange or PBX-DID routes)	34
(h)	Switching equipment congestion	42

Note 4 : The exchange also recognizes all the cause values based on ITU-T Rec. Q.763. For other Unsuccessful Calls like double or multiple Call diversion not allowed, time-out which lead to Call failure, etc., it is possible to define other 'cause values' under national standard (coding standard of cause indicators = '10').

No.	1.4	Suspend and Resume Procedure



Note 1 : The suspend message indicates a temporary cessation of communication without releasing the Call.

Note 2 : The above procedure applies to suspend initiated by the Called Party, except that the functions at the originating and destination exchanges are transposed.

Note 3 : A suspend message can also be generated by the Network in response to clear-back from a interworking node or an on-hook condition from a telephone Called Party. On the other hand, RES may be initiated by Network in response to a reanswer signal from an interworking point or an off-hook condition from an analogue Called Party.

Note 4 : If a request for reconnection or a resume message is not received within timer (T2) or timer (T6), then the controlling exchange will initiate the release procedure.

ANNEX A

SECTION 1A

Interconnect Testing

1. TESTING PRINCIPLES

1.1 The purpose of the Interconnect Testing is to provide reassurance that AAT's Network can inter-work correctly with the SingTel Network and that the Interconnection will not adversely affect the existing services provided by SingTel to SingTel Customers.

1.2 Interconnection to SingTel's Network shall be carried out and provision of services under this Agreement provided only after the satisfactory completion of the Interconnect Testing under this Annex and after SingTel is satisfied with the Interconnect Testing results in accordance with this Schedule.

2. PRE-REQUISITES FOR INTERCONNECT TESTING

2.1 Prior to the conduct of Interconnect Testing, AAT shall fully test its Network to ensure that it conforms to the Interface Specification as specified in Section 1 of Annex A. Any defects in hardware or software of AAT's Network so discovered must be corrected before the commencement of Interconnect Testing.

3. TESTING ITEMS

3.1 Interconnect Testing shall be carried out in accordance with SingTel's testing manuals. AAT shall perform Interconnect Testing in accordance with this Annex or as otherwise agreed by the SingTel, where:

(i) initial Interconnection, whether Physical Interconnection or Virtual Interconnection, is to occur;

(ii) a new POI is to be established; or

(iii) the Parties have agreed to implement a Network Change.

4. TIMELINE FOR TESTING

4.1 AAT shall book the required test date and the testing duration at least one (1) month prior to the requested testing date. AAT shall submit the application form as contained in the Attachment to SingTel to request for Interconnect Testing. The request shall contain the necessary details for the testing setup as well as the proposed test schedule.

4.2 SingTel shall respond in writing within ten (10) Business Days upon receipt of the written request on whether SingTel is able to accommodate the testing on the proposed test dates. Should SingTel not be able to perform the testing on the requested test dates, SingTel shall counter-propose an alternative test schedule with the response.

4.3 The Parties shall act in good faith and make reasonable endeavours to complete all test items within the estimated testing period.

4.4 The requested testing duration is subject to mutual agreement by the Parties.

4.5 Any request for extension to the testing duration beyond the agreed time frame by AAT is subject to mutual agreement by both Parties. AAT shall make its request for extension at least two (2) Business Days prior to the end of the testing duration.

4.6 SingTel shall not be liable to AAT for any delay in completing all the test items.

5. DAILY TIME TABLE FOR INTERCONNECT TESTING

5.1 All interconnect testing shall be carried out during Business Days between 0900 hours and 1700 hours, with one (1) hour lunch break in between.

6. TESTING RESULTS

6.1 Connection of AAT's Network to SingTel's designated IGS/SGS shall be carried out only upon satisfactory completion of the Interconnect Testing in accordance with SingTel's interconnect manuals and after SingTel is satisfied with the Interconnect Testing results.

6.2 In the event that SingTel identifies a critical problem(s), AAT shall ensure that such problems be resolved within the testing period. Otherwise, AAT shall make booking for a new testing date to verify these critical problem(s) when solutions are available.

7. CHARGES FOR INTERCONNECT TESTING

7.1 In the provision of SingTel's facilities and engineering support for the Interconnect Testing and the eventual connection to SingTel Network, AAT shall pay SingTel, the Charges based on the testing duration inclusive of any such extension period, as set out in Schedule 3.

7.2 All Calls made during the Interconnect Testing shall be chargeable to AAT.

8. CANCELLATION AND DELAY IN TESTING

8.1 AAT shall adhere to the testing date and testing duration as approved by SingTel.

8.2 Any request for cancellation of Interconnect Testing shall be made in writing to SingTel and AAT shall pay SingTel the cancellation charges in accordance with Schedule 3.

8.3 In the event that Interconnect Testing is completed or is terminated by AAT before the last day of the testing duration, AAT shall pay SingTel the Charges for the unused duration from the next Business Day after the completion or termination of Interconnect Testing to the last day of the testing duration. In which event testing shall terminate on the day on which testing was completed or terminated and SingTel shall not be obliged and shall not be required to effect any further testing for the remaining testing duration.

8.4 SingTel may unilaterally declare delay or postpone the testing date or duration due to its service exigency. SingTel shall allocate a corresponding extension of the testing period for the number of days so delayed or allocate a new testing date for Interconnect Testing. AAT shall not be liable to pay additional Charges for such extension period granted.

APPLICATION FORM FOR INTERCONNECT TESTING

OPERATOR	
Name of Operator	License Type/Class
Business Address Postal Code	
I wish to apply for SS7 Interworking Test For the period from _____ to _____ . (___ Days) **Testing Fees – S$1,200 per day**	
In support of my application, I provide the following Technical Information for the Setting up of Interconnect Testing **For SS7 Testing** 4 Digit Access Code: Signalling Point Code: Signalling Mode: Signalling Timeslot: Signalling Link Code: Circuit Direction: Circuit Selection Order:	

I confirm that we have a valid License from the Authority to operate telecommunication services. I agree that approval of this application is subject to SingTel's discretion and that SingTel reserves the right to decline the application or to make variation to the requested testing period without giving any reason.

I understand and agree that I shall execute the Agreement prior to the conduct of the Interconnect Testing. I am liable for all charges that may arise from any delay or cancellation of Interconnect Testing should the Agreement not be executed prior to the test.

I understand and agree that in additional to the charges for Interconnect Testing, all Calls made during the Interconnect Testing shall be chargeable to me.

I understand and agree that any request for cancellation of Interconnect Testing shall be made in writing to SingTel and I shall pay SingTel the cancellation charges as follows:

No. of calendar days (from the receipt of cancellation notice to the date of commencement of testing):

<7	100%
7-13	80%
14-20	35%
>21	20%

I acknowledge that the interconnect testing may only be carried out subject to the Terms and Conditions of the Agreement, and the Terms and Conditions of this Application. I agree to be bound by the said terms and conditions and in consideration of my application being approved. Upon approval, I agree to pay the charges as required.

I confirm that all the information given in making this application is true, correct and complete.

Signature Designation

Name Date

FOR SINGTEL USE

The application is Approved
Rejected
Reason for rejection: _____

Agreed Schedule for Interconnect Testing

from _____ to _____ . (___ Days)

SingTel SS7 Signalling information

Signalling Point Code

Signalling Mode

Signalling Timeslot

Signalling Link Code

Circuit Direction

Circuit Selection Order

Signature Designation

Name Date

ANNEX A

SECTION 2

ANNEX A - INTERFACE SPECIFICATION

SECTION 2 - SS7 INTERWORKING TESTING MANUAL

1 INTRODUCTION

1.1 GENERAL

1.1.1 This manual describes the test items for the SS7 testing, the testing principles and the criteria for successful testing.

1.1.2 The ITU-T Rec Q.78X referred to in this manual are the 1992 version.

2 TESTING ACTIVITIES

2.1 SS7 COMPATIBILITY TEST

2.1.1 The ITU-T Rec. Q.781 to Rec. Q.785 shall be used as guidelines for the following SS7 compatibility tests:-

(a) MTP Level 2;

(b) MTP Level 3;

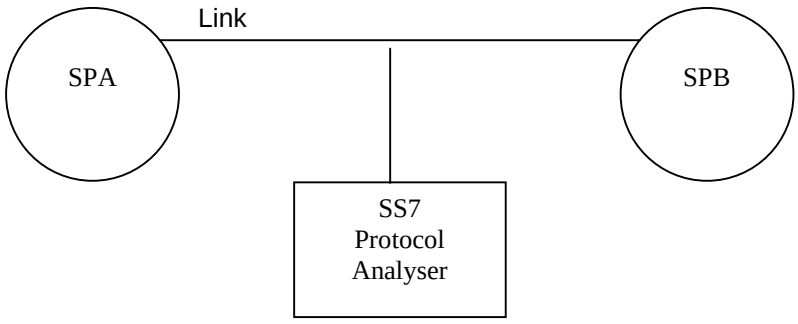
(c) ISUP Basic Call Control; and

(d) ISUP Supplementary Services.

2.1.2 The above SS7 compatibility tests are described below and shall be performed in a test environment. The details can be found in Sections 2A to 2D.

2.2 TEST CONFIGURATION

Signalling links are required to be connected as shown below. An SS7 Protocol Analyser shall be terminated across the two links to monitor and trap the SS7 messages exchanged between the two signalling points.



2.3 SS7 MTP LEVEL 2

2.3.1 The MTP level 2 test items are selected from the ITU-T Rec. Q.781 to confirm the protocol specified in Rec. Q.703.

2.3.2 The test items to verify normal/emergency alignment and activation/deactivation procedures are identified. Details of the tests can be found in Section 2A.

2.4 SS7 MTP LEVEL 3

2.4.1 The MTP level 3 test items are selected from the ITU-T Rec. Q.782 to confirm the protocol specified in Rec. Q.704 and Rec. Q.707.

2.4.2 The test items to verify Signalling Link management, signalling message handling, changeover, changeback and Signalling Link test are identified. Details of the tests can be found in Section 2B.

2.5 SS7 ISUP BASIC CALL CONTROL

2.5.1 The ISUP Basic Call Control Test are selected from ITU-T Rec. Q.784 to confirm the protocol specified in the ITU-T Rec. Q.761 to Rec. Q.764.

2.5.2 The test items to verify circuit supervision, normal Call setup, normal release, unsuccessful Call setup, and abnormal situations during a Call, special Call setup and various bearer services are identified. Details of the tests can be found in Section 2C.

2.6 SS7 ISUP SUPPLEMENTARY SERVICES

2.6.1 The ISUP protocol test for supplementary services are selected from the ITU-T Rec. Q.785 to confirm the implementation of supplementary services in the ITU-T Rec. Q.730 and Rec. Q.767.

2.6.2 The test items to verify CLI supplementary services are identified. Details of the tests can be found in Section 2D.

2.7 JUNCTION OR CIRCUIT TEST

2.7.1 The junction or circuit test shall confirm the speech connection between the IGS/SGS and AAT's Network through actual Call through test.

ANNEX A

SECTION 2A

ANNEX A - INTERFACE SPECIFICATION

SECTION 2A - SS7 MTP LEVEL 2 TEST SPECIFICATION - Q.781 RECOMMENDATION

1 Link State Control - expected signal units/orders

1.2 Timer T2

To check "Not Aligned" Timer T2. Timer T2 shall be in the range of 5 secs to 150 secs.

1.5 Normal alignment - correct procedure (FISU)

To check normal alignment procedure. Confirm that the link aligns and enters “In-service” state. Confirm that the timer T4 normal proving period is in the range of 7.5 secs to 9.5 secs (nominally at 8.2 secs).

1.21 Both ends set emergency

To check the emergency alignment procedure and timer T4 (Pe). Confirm that correct emergency alignment procedure is performed. T4 (Pe) shall be between 400 to 600 ms.

1.29 Deactivation during link in service

To check the deactivation of a Signalling Link from the "In Service" state. Confirm that an “In service” link can be put to “Out of Service” state by command.

3 Transmission failure

3.5 Link in service (Break Tx path)

To test the response to a transmission failure when link is "In service". Confirm that SIOS is returned by A when the Tx link fails.

SECTION 2A : SS7 MTP LEVEL 2 TEST SPECIFICATION – Q.781 RECOMMENDATION

TEST ITEM	Q.781 TEST ITEM	TITLE	RESULT	DATE	TESTED BY	REMARK
1	1.2	Timer T2				
2	1.5	Normal alignment - correct procedure (FISU)				
3	1.21	Both ends set emergency				
4	1.29	Deactivation during link in service				
5	3.5	Link in service (Break Tx path)				

ANNEX A

SECTION 2B

ANNEX A - INTERFACE SPECIFICATION

SECTION 2B - SS7 MTP LEVEL 3 TEST SPECIFICATION - Q.782 RECOMMENDATION

1 Signalling Link Management

1.1 First Signalling Link activation

To put into service a Signalling Linkset with 1 Signalling Link. Confirm that the Signalling Link becomes available after alignment.

1.2 Signalling linkset deactivation (where applicable)

To remove from service a Signalling Linkset with 2 Signalling Link. Confirm that the Signalling Linkset becomes unavailable.

2 Signalling message handling

2.4 Load sharing within a linkset (where applicable)

2.4.1 All links available

To check the load sharing within a linkset with all the links available. Confirm that Calls with different values of SLS are shared among the various Signalling Links.

2.4.2 With one link unavailable

To check the load sharing within a linkset when one link is unavailable. Confirm that Calls with different values of SLS are shared among the remaining available Signalling Links.

3 Changeover

3.20 Changeover as a compatibility test (where applicable)

To check the changeover procedure as compatibility test. Confirm that changeover procedure is performed over the other available Signalling Link. The procedure may be activated from one side or from both sides depending on the switch.

4 Changeback

4.1 Changeback within a linkset (where applicable)

To check that the changeback procedure is correctly performed on restoration of a link in a linkset. Confirm that the changeback procedure is correctly performed and that the link can carry traffic.

12 Signalling link test

12.1 After activation of a link

To check the Signalling Link test procedure after activation of a Signalling Link. Confirm that SLTM and SLTA messages are exchanged between the two signalling points. Confirm that the link becomes available and can carry traffic.

SECTION 2B : SS7 MTP LEVEL 3 TEST SPECIFICATION – Q.782 RECOMMENDATION

TEST ITEM	Q.782 TEST ITEM	TITLE	RESULT	DATE	TESTED BY	REMARK
1	1.1	First Signalling Link activation				
2	1.2	Signalling linkset deactivation				
3	2.4.1	Load sharing within linkset - All links available				
4	2.4.2	Load sharing within linkset - With one link unavailable				
5	3.20	Changeover as a compatibility test				
6	4.1	Changeback within a linkset				
7	12.1	Signalling link test after activation of a link				

ANNEX A

SECTION 2C

ANNEX A - INTERFACE SPECIFICATION

SECTION 2C - SS7 ISUP BASIC CALL CONTROL TEST SPECIFICATION - Q.784 RECOMMENDATION

1. Circuit supervision

1.2 Reset of circuits

1.2.1 RSC received on an idle circuit

To verify that on receipt of a reset circuit message SP A will respond by sending a release complete message.

1.2.2 RSC sent on an idle circuit

To verify that SP A is able to generate reset-circuit message.

1.2.5 Circuit group reset received

To verify that on receipt of one circuit group reset message SP A will respond by sending a circuit group reset acknowledge message.

1.2.6 Circuit group reset sent

To verify that SP A is able to generate a circuit group reset message.

1.3 Blocking of circuits

1.3.1 Circuit group blocking/unblocking

1.3.1.1 CGB and CGU received

To verify that the circuit group blocking feature can be correctly initiated.

1.3.1.2 CGB and CGU sent

To verify that SP A is able to generate one circuit group blocking message and one circuit group unblocking message.

1.3.2 Circuit blocking/unblocking

1.3.2.1 BLO received

To verify that the blocking/unblocking procedure can be correctly initiated.

1.3.2.2 BLO sent

To verify that SP A is able to generate blocking messages.

1.3.2.3 Blocking from both ends; removal of blocking from one end

To verify that the blocking/unblocking procedure can be correctly initiated.

1.3.2.4 IAM received on a remotely blocked circuit

To verify that an IAM will unblock a remotely blocked circuit.

2. Normal Call Setup

2.2 Called address sending

2.2.1 “en bloc” operation

To verify that a Call can be successfully established (all digits included in the IAM).

2.2.2 Overlap operation (with SAM)

To verify that signalling point A can initiate a Call using an IAM followed by a SAM.

2.3 Successful Call setup

2.3.1 Ordinary Call (with various indications in ACM)

To verify that a Call can be successfully completed using various indications in address complete messages.

2.3.2 Ordinary Call (with ACM, CPG and ANM)

To verify that a Call can be successfully completed using address complete message, Call progress message and answer message.

2.3.3 Ordinary Call (with various indications in CON)

To verify that a Call can be successfully completed using various indications in the connect message.

2.3.6 Blocking and unblocking during a Call (initiated)

To verify that the circuit blocking and unblocking procedure can be correctly initiated during a Call.

2.3.7 Blocking and unblocking during a Call (received)

To verify that the circuit blocking and unblocking procedure can be correctly received during a Call.

3. Normal Call release

3.1 Calling party clears before address complete

To verify that the Calling party can successfully release a Call prior to receipt of any backward message.

3.2 Calling party clears before answer

To verify that the Calling Party can successfully release a Call prior to receipt of answer.

3.3 Calling Party clears after answer

To verify that the Calling Party can successfully release a Call after answer.

3.4 Called Party clears after answer

To verify that a Call can be successfully released in the backward direction.

3.5 Suspend initiated by the Network

To verify that the Called subscriber can successfully clear and reanswer a Call.

3.6 Suspend and resume initiated by a Calling Party

To verify that the Calling subscriber can successfully suspend and resume a Call.

3.7 Suspend and resume initiated by a Called Party

To verify that the Called subscriber can successfully suspend and resume a Call.

4. Unsuccessful Call setup

4.1 Validate a set of known causes for release

To verify that the Call will be immediately released by the outgoing signalling point if a release message with a given cause is received and the correct indication is given to the Calling Party.

4.1.1 Called subscriber busy : # 17 user busy

4.1.2 Destination circuits are busy : # 42 switching equipment congestion

4.1.3 Call rejected or not accepted due to SCR,SCA or IAB service : #21 Call rejected

4.1.4 Calling to an unallocated number : #1 unallocated number

4.1.5 All outgoing routes/trunk busy : #34

4.1.6 Q.118 timer; no answer from Called Party : # 19 no answer from user

4.1.7 Address incomplete : #28

5. Abnormal situation during a Call

5.2 Timers

5.2.2 T9 : waiting for an answer message

To verify that if an answer message is not received within T9 after receiving an address complete message the connection is released by the outgoing signalling point.

5.2.4 T6 : waiting for RES (Network) message

To verify that the Call is released at the expiration of timer T6.

5.3 Reset of circuits during a Call

5.3.1 Of an outgoing circuit

To verify that on receipt of a reset message the Call is immediately released - outgoing Call.

5.3.2 Of an incoming circuit

To verify that on receipt of a reset message, a Call is immediately released - incoming Call.

7. Bearer services

7.1 64 kb/s unrestricted

7.1.1 Successful Call setup

To verify that a 64 kb/s Call can be successfully completed using appropriate transmission medium requirement and user service information parameters.

7.1.2 Unsuccessful Call setup

To verify that the Call will be immediately released by the outgoing signalling point if a release message with a given cause is received and, for circuits equipped with echo control, the echo control device is enabled.

7.2 3.1 kHz audio

7.2.1 Successful Call setup

To verify that a 3.1 kHz audio Call can be successfully completed using appropriate transmission medium requirement and information parameters.

7.3 Speech

7.3.1 Successful Call setup

To verify that a speech Call can be successfully completed using appropriate transmission medium requirement and information parameters.

SECTION 2C : SS7 ISUP BASIC CALL CONTROL TEST SPECIFICATION – Q.784 RECOMMENDATION

TEST ITEM	Q.784 TEST ITEM	TITLE	RESULT	DATE	TESTED BY	REMARK
1	1.2.1	RSC received on an idle circuit				
2	1.2.2	RSC sent on an idle circuit				
3	1.2.5	Circuit group reset received				
4	1.2.6	Circuit group reset sent				
5	1.3.1.1	CGB and CGU received				
6	1.3.1.2	CGB and CGU sent				
7	1.3.2.1	BLO received				
8	1.3.2.2	BLO sent				
9	1.3.2.3	Blocking from both ends; removal of blocking from one end				
10	1.3.2.4	IAM received on a remotely blocked circuit				
11	2.2.1	"EN BLOC" operation				
12	2.2.2	Overlap sending (with SAM)				
13	2.3.1	Ordinary Call (with various indications in ACM)				

TEST ITEM	Q.784 TEST ITEM	TITLE	RESULT	DATE	TESTED BY	REMARK
14	2.3.2	Ordinary Call (with ACM, CPG and ANM)				
15	2.3.3	Ordinary Call (with various indications in CON)				
16	2.3.6	Blocking and unblocking during a Call (initiated)				
17	2.3.7	Blocking and unblocking during a Call (received)				
18	3.1	Calling Party clears before address complete				
19	3.2	Calling Party clear before answer				
20	3.3	Calling Party clear after answer				
21	3.4	Called Party clear after answer				
22	3.5	Suspend initiated by the Network				
23	3.6	Suspend and resume initiated by a Calling Party				
24	3.7	Suspend and resume initiated by a Called Party				
25	4.1.1	Called subscriber busy: #17 user busy				
26	4.1.2	Destination circuits are busy: #42 switching equipment congestion				
27	4.1.3	Call rejected or not accepted due to SCR,SCA or IAB service : #21 Call rejected				
28	4.1.4	Calling to an unallocated number: #1 unallocated number				

TEST ITEM	Q.784 TEST ITEM	TITLE	RESULT	DATE	TESTED BY	REMARK
29	4.1.5	All outgoing routes/trunks busy : #34				
30	4.1.6	Address incomplete : #28				
31	4.1.7	Q.118 timer, no answer from Called Party: #19 no answer from user				
32	5.2.2	T9: waiting for an answer message				
33	5.2.4	T6: waiting for RES (Network) message				
34	5.3.1	Of an outgoing circuit				
35	5.3.2	Of an incoming circuit				
36	7.1.1	Successful Call setup (64kbit/s unrestricted)				
37	7.1.2	Unsuccessful Call setup (64kbit/s unrestricted)				
38	7.2.1	Successful Call setup (3.1kHz audio)				
39	7.3.1	Successful Call setup (Speech)				

ANNEX A

SECTION 2D

ANNEX A - INTERFACE SPECIFICATION

SECTION 2D - SS7 ISUP PROTOCOL TEST SPECIFICATION FOR SUPPLEMENTARY

SERVICES - Q.785 RECOMMENDATION

3 CLI

3.1.1 CLIP - Network provided : sent

To verify that CLIP (Network provided) can be correctly sent in the Calling Party number parameter.

3.1.2 CLIP - Network provided : received

To verify that CLIP (Network provided) can be correctly received in the Calling Party number parameter.

3.2.1 CLIP - user provided : sent

To verify that CLIP (user provided) can be correctly sent in the Calling Party number parameter.

3.2.2 CLIP - user provided : received

To verify that CLIP (user provided) can be correctly received in the Calling Party number parameter.

3.3.1 CLIR - Network provided : sent

To verify that CLIR (Network provided) can be correctly sent in the Calling Party number parameter.

3.3.2 CLIR - Network provided : received

To verify that CLIR (Network provided) can be correctly received in the Calling Party number parameter.

3.4.1 CLIR - user provided : sent

To verify that CLIR (user provided) can be correctly sent in the Calling Party number parameter.

3.4.2 CLIR - user provided : received

To verify that CLIR (user provided) can be correctly received in the Calling Party number parameter.

SECTION 2D : SS7 ISUP PROTOCOL TEST SPECIFICATION FOR SUPPLEMENTARY SERVICES - Q.785 RECOMMENDATION

TEST ITEM	Q.785 TEST ITEM	TITLE	RESULT	DATE	TESTED BY	REMARK
1	3.1.1	CLIP - Network provided: sent				
2	3.1.2	CLIP - Network provided: received				
3	3.2.1	CLIP - user provided: sent				
4	3.2.2	CLIP - user provided: received				
5	3.3.1	CLIR - Network provided: sent				
6	3.3.2	CLIR - Network provided: received				
7	3.4.1	CLIR - user provided: sent				
8	3.4.2	CLIR - user provided: received				

ANNEX A

SECTION 2E

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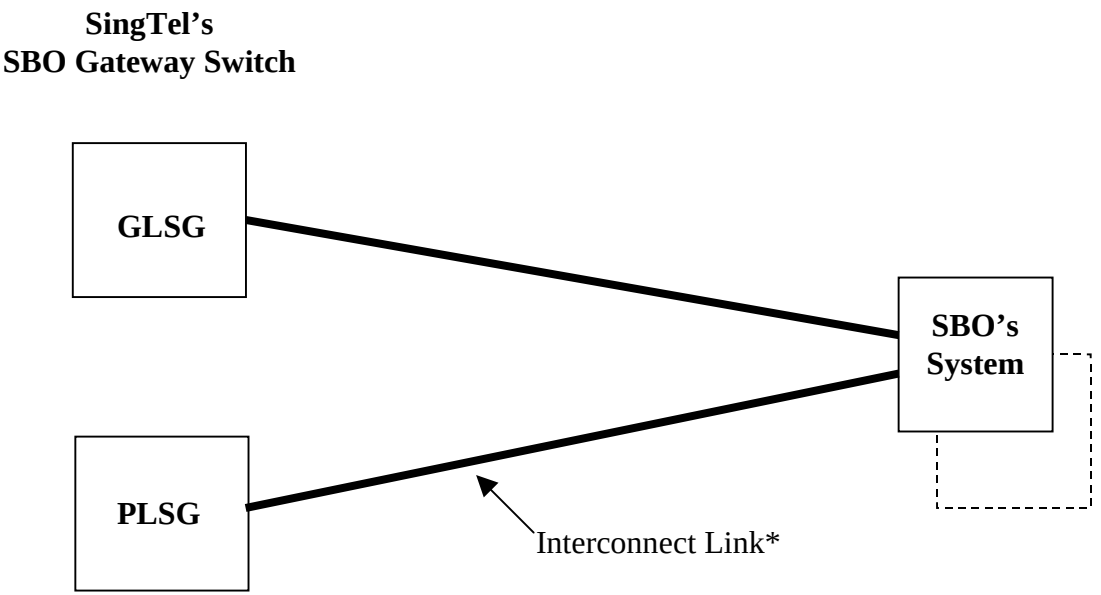
ANNEX A

SECTION 2F

Interconnect Links and Interconnect Configuration between SingTel and SBO



Legend :

GLSG – Geylang SGS

PLSG – Paya Lebar SGS

* Minimum Interconnection Capacity of 2 E1s per SGS

ANNEX B

OPERATIONAL PROCEDURES

ANNEX B – OPERATIONAL PROCEDURES

1. INTRODUCTION

1.1 This Annex provides the operations and maintenance procedures to be carried out by AAT to maintain satisfactory connection to the SingTel's Network. It generally provides the fault handling procedures related to the Network. Procedures for carrying out planned engineering works, testing and monitoring are also described in this Section.

2. FAULT HANDLING PROCEDURES

2.1 General

2.1.1 Prior to activating the fault handling procedures, the Party reporting the fault (**reporting Party**) must reasonably establish that a genuine fault exists and also that every effort has been made to prove that the fault is not within the reporting Party's side of the POI.

2.1.2 Faults related to the Local Leased Circuits from SingTel is not covered in this Agreement and its Annexes. AAT is responsible for the Local Leased Circuit. It is AAT's responsibility to ensure that the fault does not lie within its side of the POI, including fault relating to the Interconnection Link before reporting the fault to SingTel under this Agreement.

2.1.3 Each Party shall maintain its own fault reporting centre which shall be responsible for handling the faults between Networks, coordinating the fault clearance (including escalations) within its own Network and subsequently reporting the clearance of the fault to the other Party. Appendices 1 and 2 contain details of both Parties fault reporting centres.

2.1.4 Both Parties shall co-operate in any investigation and follow up actions and keep each other informed on the status of the progress of the fault clearance in a timely manner.

2.1.5 Each Party shall establish 24 hour contact points for fault reporting at its nominated fault reporting centre. Appendices 1 & 2 contain information on the contact points of the Parties for such purpose.

2.2 Type Of Faults

2.2.1 Faults reported may be classified as follows:

(i) Signalling Link faults; and

(ii) Gateway Switch Network faults.

2.2.2 Signalling Link Faults

2.2.2.1 All Signalling Links provided by SingTel shall be supervised closely by AAT and report any fault to the reporting centre of SingTel as soon as possible.

2.2.3 Gateway Switch Network Faults

2.2.3.1 Faults related to the IGS/SGS or Licensee's system shall be referred to the related IGS/SGS Switch during office hours, or NMC during after office hours.

3. PLANNED ENGINEERING WORKS

3.1 For any planned engineering works within AAT's Network, which will result in momentarily outage of service of the local leased circuit, SS7 Signalling Links, or Gateway Exchange, AAT shall inform SingTel through the contact points, by fax, given in Appendices 1 & 2.

3.2 The details of the works to be carried out shall be recorded on an "Advice of Planned Engineering Work" form, (**Advice form**). The Advice form as provided in Appendix 3 shall state the date, time and duration of such works, the impact to the conveyance of Calls between the Parties' Network, any Network management procedures required, and contingency measures to be taken by either Party or both Parties. The schedule and duration of the planned work proposed by AAT shall be agreed by SingTel before the commencement of such works.

3.3 AAT, prior to performing the planned engineering works, shall give advance notice of at least five (5) Business Days to the other Party.

3.4 The preferred times and duration allowed for carrying out various planned engineering works shall be between 0100 through 0500 hrs, applicable on everyday, including public holiday.

3.5 AAT shall notify SingTel that the works have been completed by completing and fax to SingTel, the last section of the Advice form.

4. TESTING AND MONITORING

4.3 AAT shall be responsible to test and monitor the performance of its own Network. Testing of the Interconnection Link and Signalling Links shall be kept to a minimum and shall avoid the busy hour periods. No testing shall be carried out before SingTel agreed to the conduct such tests, including any routine tests.

4.4 For handling problems which can only be localised through a series of test Calls (eg difficulty in reaching certain number groups), both Parties shall agree upon the details of the testing required. Test numbers and contact points shall be exchanged to facilitate the testing.

APPENDIX 1

1 Notification Points for Contact in SingTel:-

(a) SingTel Network Management Centre (NMC) (after office hours)

Location :

Telephone :

Facsimile :

Supervisor :

(b) SingTel IGS/SGS

Location :

Telephone :

Facsimile :

Supervisor :

APPENDIX 2

1 Notification Contact Points in Requesting Licensee

(a) Requesting Licensee's Network Management Centre (NMC) - 24 hours

Location :

Telephone :

Facsimile :

Supervisor :

(b) Name of Requesting Licensee's Network Location:

Location :

Telephone :

Facsimile :

Supervisor :

APPENDIX 3

Advice of Planned Engineering Works

Subject:	*Title of the planned works*
Switch/ Location:	*Indicate the Switch or location of the planned work*
Type of planned works:	*Signalling Link /Interconnection Link/Exchange*
Outage Date:	*Indicate the date of the planned work*
Outage Time:	*Indicate the start time of the planned work.*
Service Interruption Duration:	*Provide an estimated duration on the service interruption*
Number of local leased circuit/ Signalling Links affected:	*Indicate the number and system ID of the local leased circuit or Signalling Links affected by the planned work*
Effect of planned work:	*Describe the effect of the planned works on Calls and in which direction*
Reason of planned work:	*Describe the reason for the planned works eg due to routine/urgent maintenance or software upgrade etc*
Remarks:	*To include additional comments or remarks eg Preparation work will commence at around "time" on "date"*
Issuing Officer:	*Indicate the name and designation of the officer issuing the advice of planned work.*

ANNEX C – FORECASTING OF INTERCONNECT CAPACITY

Interconnect Capacity Forecasts shall be in accordance with clause 6 of Schedule 1 and include the maximum Interconnect Capacity required at each POI.

Interconnect Capacity Forecast for the Interconnect Link between _____ and _____ .				
Forecasting Period: From _____ to _____				
	Year 1		Year 2	Year 3
	First 6 months	Subsequent 6 months		
No. of outgoing E1s:				
No. of incoming E1s:				
No. of both-way E1s:				
		Outgoing Circuits	Incoming Circuits	Both-way Circuits
No. of E1s to be turned on in 1st month of Forecasting period:				
No. of E1s to be turned on in 2nd month of Forecasting period:				
No. of E1s to be turned on in 3rd month of Forecasting period:				
No. of E1s to be turned on in 4th month of Forecasting period:				
No. of E1s to be turned on in 5th month of Forecasting period:				