

Known Issues

Please contact Samsung Technical Support with any questions or comments concerning this list of Known Issues. 1(800) 737 - 7008 or bcs.support@samsung.com

May 2011

| Ref # Issue Type Issue Name | Issue Description | Workaround Description |
|---|--|--|
| p110202001 - 1322 | Test across entire 7000 series platform using latest software v4.46d & v4.51 | No work around |
| MMC: 121 Doesnt function when using French/Canada on TDM phones | 1)From a TDM go to MMC 121 2)Change language to French/Canada 3)Leave yourself a quick memo 4)Line 1 still has "1 new message" in English but should be in French/Canada 5)Line2 on Display reads French/Canada 6)Change language to Spanish and both lines are correct. | |
| p110215002 - 1324 | All 7000 series system using SIP trunking | Contact Tech Support |
| SIP trunking- Possibility of system being open to call fraud | We have found that there is a possible way to make calls thru the phone system without the call coming from the SIP provider. If there is a digit that terminates to the (LCR) digit; 9 commonly calls can be made. This is the same thing as using the tandem trunking feature. | |
| p110303001 - 1325 | 7000 series systems running v4.53 use appropriate voice | No work around at this time. |
| Call coverage and VMAME feature dont function proper together | mail for each system. In MMC 701 under useable features enable the VMAME feature. Ext 202 will provide call coverage for ext 201. Assign a CC201 key to ext 202 Set up the VMAME feature on ext 201. Assign VMAME key to ext 201. Set the forward no answer feature on ext 201 to forward to voicemail. In MMC 502 lower the CC RNG DLY timer to 1 second so CC ext 202 will ring immediately. Ext 201 gets a outside call and after the ring no answer timer expires the call goes to voicemail. He can listen to the call, send the caller to voicemail or answer the call. The CC ext 202 is ringing too. The problem is the ext never stops ringing. The CC key is flashing the entire time. If there is a DS key on the phone it will flash too. | |
| p110310001 - 1326 | OS7200 with IPUMS setup and work ing | Remove prompt from MMC 748 and save. Add prompt back to MMC 748 and save. |
| IPUMS-Music on hold stops playing | Convert music file to 16 bit 8kHZ WAV file. Use voice studio to upload and name file to proper prompt number. I used 5012. Since this is IPUMS and the port are not physical you have to use IT Tool MMC 748 to assign the port with the prompt 5012. This will remove the port from the voice mail group. Assign voice mail port to MMC 308 back ground music, MMC 309 for station music on hold and MMC 714 for DID music on hold. Make test call internally or externally and put call on hold the music plays. Come in the next morning and attempt to listen to back ground music; nothing plays. | |

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| Issue Name | | | |
| P110312001 - 1327 | | MP10a and 7200S | Change MMC 835 option DUAL FLT EC to DISABLE |
| Network mailbox do properly | o not work | Program network mailbox on each system. Program a 500/7200/7400 using a 16E or 20E. Assign network mailboxes to call between the two type of systems. Program analog trunks on each system. Leave message in a mailbox to initiate the AMIS feature. The systems goes off hook and dial the opposite system without a problem. The system answers and you can see them communicating thru port activity. The two systems start the process but fails. Calls can originate from the 7100/7200 to the500/7200/7400 or vice versa and the results are the same. If you do the same thing from 16E to 20E the calls connect and complete the message transfer every time. | DIOADLE |
| p110322001 - 1331 Voice announce-tra | nsferred call | 3/28/11-Robert- opened qnet ticket of P110322001. Create a SPNET network using (2) 7100 systems running v4.53. The trunks coming in on each system are analog. | I found there is a way to avoid the speaker phone activating. After the call is released and the phone starts to ring at the station, press the speaker button and then the call |
| to be heard over the and handset. | e speaker | The call flow is as follows: Calls come into node (A) they are passed thru to the node (B) using (MMC 406). The call is answered by a station. The call is then transferred back across the network from node (B) to (A). The phone in node (A) being transferred to is in voice announce mode. The party receiving the call picks up his handset to say "send the call". The party transferring the call press the release button to release the call. The phone begins to ring, when it is answered the caller will be heard over the speaker phone and the handset. | button and the problem doesn't happen. The only other work around is to not use the voice announce mode |
| p110413001 - 1334 | | 7200 MP20 v4.51 or 4.53b 7200 cabinet | Use v4.30 software. |
| E&M T1 doesnt sho | _ | MP20 v4.51 or 4.53b TEPRIa v4.28 set as T1 | |
| number on a incom | ing idle call | System started out on v4.30 and was upgraded to v4.51 and the ANI/DNIS digits no longer show up on an incoming call to an idle station. System has pri card in proper slot and it set as a T1 E&M wink. Set up MMC 420 to proper ANI/DNIS construction based on Telco instructions. A call comes into the system via DID in MMC 714. Digits ex: 4321 point to ext 201. There is no name on the T1 but caller id (ani/dnis) is set to be delivered. The caller id doesn't appear with MMC 119 'ring line 1" &" ring line 2" both set to number first. If the person at ext 201 is on a call the caller id shows. Another strange thing is even though there is no caller id when idle if you review the call logs the number is there. Note: v4.30 doesnt exhibit the same issue. | |
| TS4520111 - 1335 | | 7000 series systems running v4.53b (MP10a, MP20, MP20S & MP40) | Use IT Tool to input either field. Press the * button twice. |
| MMC 837 Cannot in | - | Use a defaulted system with a DLI card Plug any digital keyset | |
| address "REGIST A "OUT PROXY" usin | | Log into KMMC; Transfer 800, password 4321, enable tech programming, press speaker button, input 837. Scroll to ISP1, right soft key and volume up to "REGIST ADDR" or "OUT PROXY". You will notice that the normal "periods" that separate each octet is missing. Use the keypad to enter the ip address for both field and the number run together. Even if you try and use the "*" to separate it doesn't work. | |

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| 41520111 - 1336 7200-S v4.51-Cannot upg downgrade PRI card | grade or | 7200-S v4.51 Configure a system with a MP20-S processor, DLi card and PRI installed in proper slot. Use MMC 818 to upgrade or downgrade a PRI card to 4.XX software. The phone that this is executed from will constantly show "media busy". The PRI will appear to be upgrading but it isnt. | Upgrade to V4.53x software and the upgrade or downgrade will work. |
| p110429001 - 1341 iDCS 28 button phone automatically execute fe on button 12 | eature | 7100 MP10a, 7200-S, 7200 and 7400 8 DLi or 16 DLi card From an iDCS 28 button perform a function that causes the system to give an error tone. While getting the error tone press the speaker button. The phone will automatically execute whatever feature is assigned to button 12. An example: Assign button 12 as DS201. Dial a invalid number and while the error tone is sounding press the speaker button. The phone will automatically call DS201 which is assigned to button 12. | Allow error tone to finish or press the ANS/RLS button |
| P110505001 - 1342 SMT5210&3105 Headset lose audio and touch touch the heard over speakerphone | nes are | When you are in Headset mode and have the volume set at least one level and then make a call or check VM If you then adjust the VOL and then hit ANY DIGIT, you loose audio and the tones start coming out of the speaker . | No work around |
| 50520112 - 1343 DS-5021 performs redial pressing end button | when | 7100 MP10a, 7200-S, 7030 8 DLI or 16 DLI card 8 TRK, 16 TRK ot TEPRi From a DS-5021 perform a function that causes the system to give an error tone. While getting the error tone press the end button. The phone will automatically redial the last number. An example:From the DS5021 phone make a call out of the phone system.9-18007377008 allow the call to setup and then hang up. Dial a invalid number and while the error tone is sounding press the end button. The phone will automatically redial the last number 9-18007377008. | Wait for error tone to stop. |
| p110527002 - 1345 "Find me" feature doesn or works very inconsiste | | T030, 7100 7200-S v4.53c Login into web management or use device manager for voicemail programming. Select "open block" Select "extension block" Click on the desired extension: Ex 201 Click on the "authorization" tab Authorize the feature by setting the "Find me allowed" to YES To enable it set "Enabled" option to YES. In order to add the numbers select the "additional information" tab. Add up to 6 numbers to be called. In order to test call into the station DID, forward etc. EXT 201 forward all to voicemail. When the call is forwarded to voicemail it will initiate the feature. When using cell phone numbers the system is attempting to call out but the cell never rings or gets 1 ring. The system is already seeing a loop open and it starts to dial the next cell number and it keeps repeating the same thing. We have also seen where it rings and once the cell call is answered it never cuts through so you can talk. Using a "land line" discovered that it will work as designed but it can be intermittent. | |

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| | vork | 7030, 7100 7200-S v4.53c Login into web management or use device manager for voicemail programming. Select "open block" Select "mailbox block" Click on the desired mailbox: MBX 201 Click on the "alerts" tab Enable the feature by setting the "Message alert is currently on" to YES In the "Alert phone number" input the phone number to alert. Ex 9727922000 Next enable the auto forward feature by clicking the "MWI & Auto Forward" tab. Enable the "Enable auto forward" to YES and the "Delete after forwarding" to YES. Set the "Auto forward delay" ex: 0:3 which would be 3 minutes. Select the "call director" and find and select the mailbox the message is to forward too. Ex: MBX 202. You will then follow the same procedure to set up MBX 202 and any subsequent mailboxes. In order to test call into the station DID, forward etc. EXT 201 forward all to voicemail. When the call is forwarded to voicemail, the caller leaves a message; the system it will initiate the feature. The voicemail calls out and there is no answer. The system waits the 3 minutes before auto forwarding and deleting the message. The message. The message | Workaround Description |
| | | number stamp is different from the original message but, the system doesn't even attempt call out. | |
| TS02520101 - 1311 | | System Type: IPUMS System Version: 1.3.5.3-20091110 | Please contact Tech Support for file to load to IPUMS |
| "Query" block not allow recordings | ving | Problem: When creating a Query Block. Take input from "VOICE" is not capturing or allowing the recording of comments. The system play query prompt and automatically hits the next pointer. This issue has only been found with "VOICE", validates DTMF receives if input is placed as "Entry" | |
| 112220103 - 1317 Getting "File Not Found problem when trying to a VM MSG in Outlook. | | File Not Found - File Not Found when playing back messages from previous day. At the time of the question the users that reported the issue were receiving the error when they would open up a message that was left two days prior. We found that they could close the error and click on the file attachment to play the message back. | Contact Tech Support for so0ftware. |
| 112220104 - 1318 | | Error does not occur from Midnight to 8 am, From 8 to 830 am will have 1 or 2 errors, from 830 to noon, steady | Contact Tech Support for beta software |
| Getting a "MailFetch" s error | server | continual increase in error (No specific number was given) Errors Peek at Noon, Error reduces from Noon to 4 PM | |
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| p110421001 - 1337 SW 4.53 or newer softwa improper End key featur 5121 or DS5121 | | On SW 4.53 or newer, if you create an error tone by dialing a phone that is plug out or an invalid feature number and then press the END key it will do the LNR feature instead of ending the call. This was not happening prior to 4.53. | There is no workaround at this time. |
| P100916001 - 1293 Dialtone is heard throug computer speakers. | CTI h | OS Communicator Desk phone mode 7200S with v4.46Dd software DS 5021 or iDCS 18 Install OS Link3 Install OS Communicator Activate Link Set OS Communicator to use desk phone mode thru configuration option Activate Communicator in desk phone mode (Ext 201) Make test call to phone to confirm OS Communicator is working. (Ext 202 calls 201) You can see screen pop on computer. Ext 201 goes off hook and the dial tone can be heard through the computer's speakers. Customer logs into "keyset user options" MMCs dialtone is heard through the speakers. Ex: Ext 201 press "transfer" button to log in (dial tone is heard) it stays there until user logs out of programming. | No work around |