Asterisk Project Security Advisory - AST-2007-014

Product	Asterisk	
Summary	Stack buffer overflow in IAX2 channel driver	
Nature of Advisory	Exploitable Stack Buffer Overflow	
Susceptibility	Remote Unauthenticated Sessions	
Severity	Critical	
Exploits Known	No	
Reported On	July 12, 2007	
Reported By	Russell Bryant, Digium, Inc.	
Posted On	July 17, 2007	
Last Updated On	August 21, 2007	
Advisory Contact	Russell Bryant <russell@digium.com></russell@digium.com>	
CVE Name	CVE-2007-3762	

Description

The Asterisk IAX2 channel driver, chan_iax2, has a remotely exploitable stack buffer overflow vulnerability. It occurs when chan_iax2 is passed a voice or video frame with a data payload larger than 4 kB. This is exploitable by sending a very large RTP frame to an active RTP port number used by Asterisk when the other end of the call is an IAX2 channel. Exploiting this issue can cause a crash or allow arbitrary code execution on a remote machine.

The specific conditions that trigger the vulnerability are the following:

- iax2 write() is called with a frame with the following properties
 - o a voice or video frame
 - Its 4-byte timestamp has the same high 2 bytes as the previous frame that was sent
 - Its format is the one currently expected
 - Its data payload is larger than 4 kB

iax2_write() calls iax2_send() to send the frame. Inside of iax2_send(), there is a conditional check to determine whether the frame should be sent immediately (the now variable) or queued for transmission later.

If the frame is going to be transmitted later, an iax_frame struct is dynamically allocated with a data buffer that has the exact buffer size needed to accommodate for the provided ast_frame data. However, if the frame is being sent immediately, it uses a stack allocated iax frame, with a data buffer size of 4096 bytes.

Later, the iax_frame_wrap() function is used to copy the data from the ast_frame struct into the iax_frame struct. This function assumes the iax_frame data buffer has enough space for all of the data in the ast_frame.

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Resolution

This issue is only exploitable when the system is configured in such a way that calls between channels that use RTP and IAX2 channels are possible. Also, some additional protection against arbitrary code execution is provided if the call involves transcoding between audio formats as this will change the contents of the frame payload.

All users that have systems that connect calls between channels that use RTP and IAX2 channels should immediately update to versions listed in the corrected in section of this advisory.

Affected Versions				
Product	Release Series			
Asterisk Open Source	1.0.x	All versions		
Asterisk Open Source	1.2.x	All versions prior to 1.2.22		
Asterisk Open Source	1.4.x	All versions prior to 1.4.8		
Asterisk Business Edition	A.x.x	All versions		
Asterisk Business Edition	B.x.x	All versions prior to B.2.2.1		
AsteriskNOW	pre- release	All versions prior to beta7		
Asterisk Appliance Developer Kit	0.x.x	All versions prior to 0.5.0		
s800i (Asterisk Appliance)	1.0.x	All versions prior to 1.0.2		

Corrected In				
Product	Release			
Asterisk Open Source	1.2.22 and 1.4.8, available from http://downloads.digium.com/pub/telephony/asterisk			
Asterisk Business Edition	B.2.2.1, available from the Asterisk Business Edition user portal on http://www.digium.com or via Digium Technical Support			
AsteriskNOW	Beta7, available from http://www.asterisknow.org/ . Beta5 and Beta6 users can update using the system update feature in the appliance control panel.			
Asterisk Appliance Developer Kit	0.5.0, available from http://downloads.digium.com/pub/telephony/aadk/			
s800i (Asterisk Appliance)	1.0.2			

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Revision History				
Date	Editor	Revisions Made		
July 17, 2007	russell@digium.com	Initial Release		
July 17, 2007	russell@digium.com	Minor Spelling Fix		
August 21, 2007	russell@digium.com	Changed name prefix from ASA to AST, changed ftp.digium.com to downloads.digium.com		